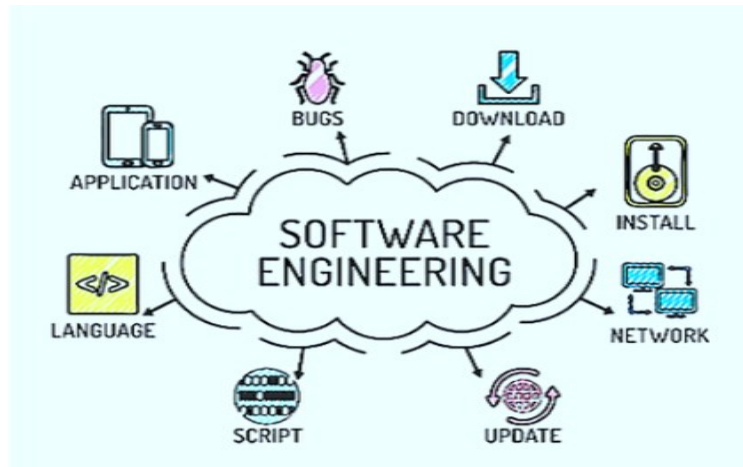


SOFTWARE ENGINEERING

UNIT-I

INTRODUCTION

Computers were very slow in the initial years and lacked sophistication. If similar improvements could have occurred to aircrafts, now personal mini-airplanes should have become available, costing as much as a bicycle, and flying at over 1000 times the speed of the supersonic jets. The more powerful a computer is, the more sophisticated programs can it run. Therefore, with every increase in the raw computing capabilities of computers, software engineers have been called upon to solve increasingly larger and complex problems, and that too in cost-effective and efficient ways. Software engineers have coped up with this challenge by innovating and building upon their past programming experiences.



Software engineering is a systematic and cost-effective technique for software development. These techniques help develop software using an engineering approach. The innovations and past experiences towards writing good quality programs cost- effectively, have contributed to the emergence of the software engineering discipline.

For example: If someone wants to travel from Punjab to Delhi. There are two approaches one can follow to achieve the same result:

1. The normal approach is just to go out and catch the bus/train that is available.
2. A systematic approach is constructed as Firstly check on Google Maps about distance and after analyzing the timing of trains and buses online and after that, match user's preference suppose user have some work till 4:00 PM and trains slot are: 1:00 PM, 6:00 PM then the user will choose 6:00 PM time slot and reach Delhi.

From the above situation, one can easily analyze that Creating a systematic Approach is more optimal and time and cost-effective as compared to a normal approach. This will same occur while designing Software. So in Software Engineering working on an Engineering or a Systematic approach is more beneficial.

What is software engineering?

A popular definition of software engineering is: "A systematic collection of good program development practices and techniques". Good program development techniques have resulted from research innovations as well as from the lessons learnt by programmers through years of programming experiences. An alternative definition of software engineering is : "An engineering approach to develop software".

Suppose you have asked a petty contractor to build a small house for you. Petty contractors are not really experts in house building. They normally carry out minor repair works and at most undertake very small building works such as the construction of boundary walls. For example, he might not know the optimal proportion in which cement and sand should be mixed to realize sufficient strength for supporting the roof. In such situations, he would have to fallback upon his intuitions. Of course, the house constructed by him may not look as good as one constructed by a professional, may lack proper planning, and display several defects and imperfections. It may even cost more and take longer to build. The failure might come in several forms—the building might collapse during the construction stage itself due to his ignorance of the basic theories concerning the strengths of materials; the construction might get unduly delayed, and quantities of raw materials required, the times at which these are required, etc. In short, to be successful in constructing a building of large magnitude, one needs a good understanding of various civil and architectural engineering techniques such as analysis, estimation, prototyping, planning, designing, and testing.

For sufficiently small-sized problems, one might proceed according to one's intuition and succeed; though the solution may have several imperfections, cost more, take longer to complete, etc.

As is usual in all engineering disciplines, in software engineering several conflicting goals are encountered while solving a problem. In such situations, several alternate solutions are first proposed. An appropriate solution is chosen out of the candidate solutions based on various trade-offs that need to be made on account of issues of cost, maintainability, and usability. Therefore, while arriving at the final solution, several iterations are possible.

Abstraction Versus Decomposition

Two important principles that are deployed by software engineering to overcome the problems arising due to human cognitive limitations are—abstraction and decomposition

Abstraction:

In software engineering and computer science, abstraction is a technique for arranging complexity of computer systems. Abstraction means displaying only essential information and hiding the details. It works by establishing a level of simplicity on which a person interacts with the system, suppressing the more complex details below the current level. Data abstraction refers to providing only essential information about the data to the outside world, hiding the background details or implementation. Consider a real life example of a man driving a car.

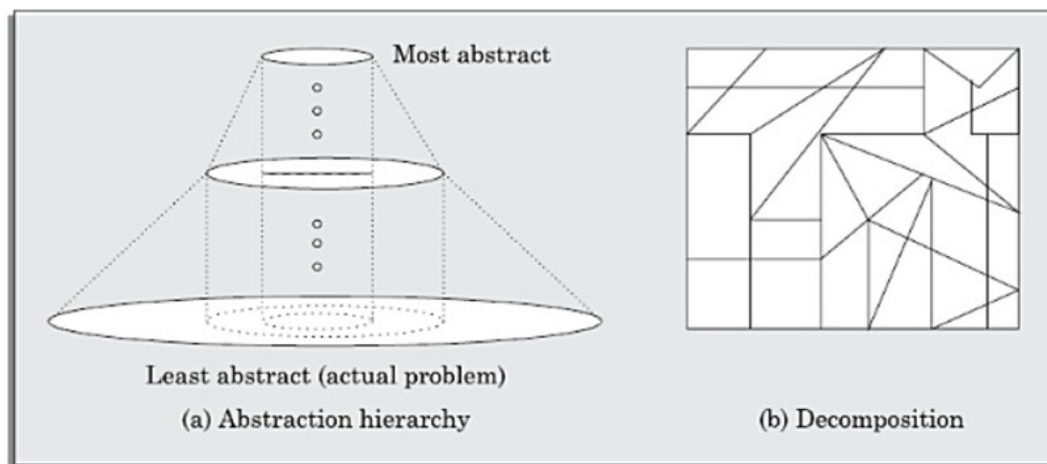


Fig: Abstraction and Decomposition

Decomposition:

It is another important principle that is available in the repertoire of a software engineer to handle problem complexity. This principle is profusely made use by several software engineering techniques to contain the exponential growth of the perceived problem complexity. The decomposition principle is popularly known as the divide and conquer principle. The decomposition principle advocates decomposing the problem into many small independent parts. The small parts are then taken up one by one and solved separately. The idea is that each small part would be easy to grasp and understand and can be easily solved. The full problem is solved when all the parts are solved.

Evolution of software design techniques

1. Early Computer Programming: Early commercial computers were very slow and too elementary as compared to today's standards. Even simple processing tasks took considerable computation time on those computers. No wonder that programs at that time were very small in size and lacked sophistication. Those programs were usually written in assembly languages.

Program lengths were typically limited to about a few hundreds of lines of monolithic assembly code. Every programmer developed his own individualistic style of writing programs according to his intuition and used this style ad hoc while writing different programs.

2. High-level Language Programming: Computers became faster with the introduction of semiconductor technology in the early 1960s. Faster semiconductor transistors replaced the prevalent vacuum tube-based circuits in a computer. With the availability of more powerful computers, it became possible to solve larger and more complex problems.

At this time, high-level languages such as FORTRAN, ALGOL, and COBOL were introduced. This considerably reduced the effort required to develop software and helped programmers to write larger programs. Writing each high-level programming construct in effect enables the programmer to write several machine instructions. However, the programmers were still using the exploratory style of software development.

3. Control Flow-based Design: A program's control flow structure indicates the sequence in which the program's instructions are executed. In order to help develop programs having good control flow structures, the flowcharting technique was developed. Even today, the flowcharting technique is being used to represent and design algorithms.

4. Data Structure-oriented Design: Computers became even more powerful with the advent of Integrated Circuits (ICs) in the early 1970s. These could now be used to solve more complex problems. Software developers were tasked to develop larger and more complicated software, which often required writing in excess of several tens of thousands of lines of source code.

It is much more important to pay attention to the design of the important data structures of the program than to the design of its control structure. Design techniques based on this principle are called Data Structure-oriented Design.

Example: Jackson's Structured Programming (JSP) technique developed by Michael Jackson (1975). In JSP methodology, a program's data structure is first designed using the notations for sequence, selection, and

iteration. The JSP methodology provides an interesting technique to derive the program structure from its data structure representation.

5. Data Flow-oriented Design: As computers became still faster and more powerful with the introduction of very large scale integrated (VLSI) Circuits and some new architectural concepts, more complex and sophisticated software were needed to solve further challenging problems. Therefore, software developers looked out for more effective techniques for designing software and Data Flow-Oriented Design techniques were proposed.

The functions are also called as processes and the data items that are exchanged between the different functions are represented in a diagram is known as a Data Flow Diagram (DFD).

6. Object-oriented Design: Object-oriented design technique is an intuitively appealing approach, where the natural objects (such as employees, etc.) relevant to a problem are first identified and then the relationships among the objects such as composition, reference, and inheritance are determined. Each object essentially acts as a data hiding is also known as data abstraction.

Object-oriented techniques have gained wide spread acceptance because of their simplicity, the scope for code and design reuse, promise of lower development time, lower development cost, more robust code, and easier maintenance.

let us first examine what are the current challenges in designing software. First, program sizes are further increasing as compared to what was being developed a decade back. Second, many of the present day software are required to work in a client-server environment through a web browser-based access (called web-based software).

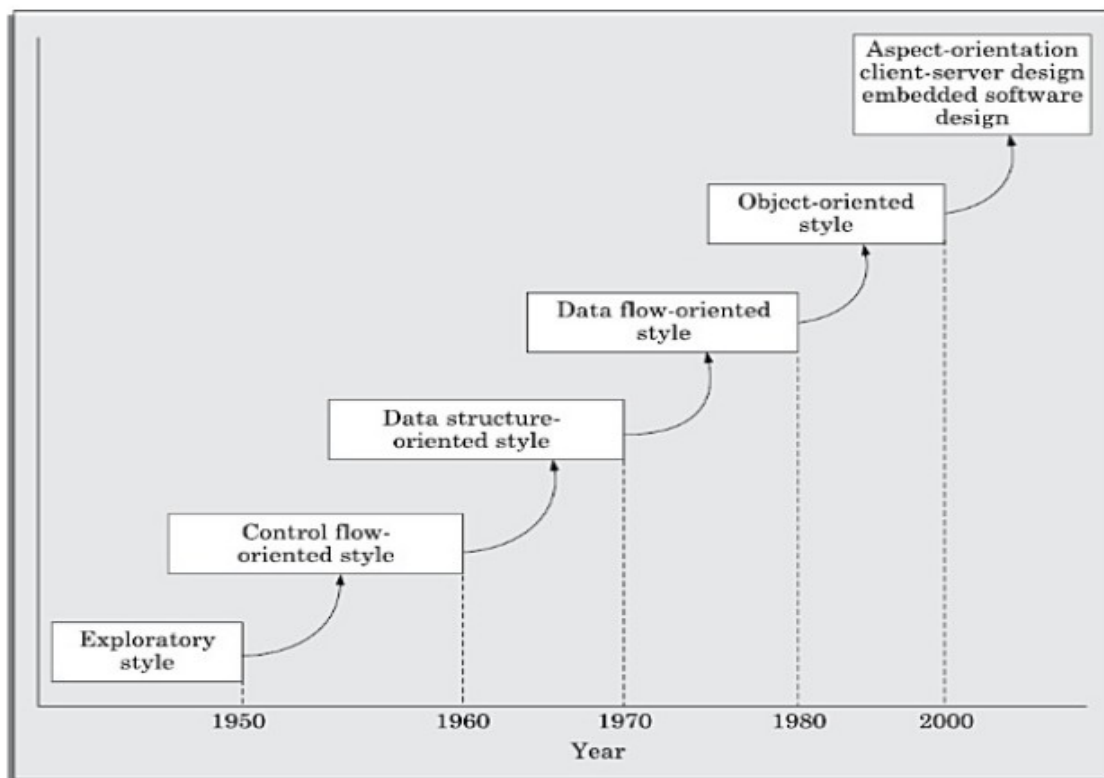


Figure 1.12: Evolution of software design techniques.

The improvements to the software design methodologies over the last five decades have indeed been remarkable. These new techniques include life cycle models, specification techniques, project management techniques, testing techniques, debugging techniques, quality assurance techniques, software measurement techniques, computer aided software engineering (CASE) tools, etc

Software life cycle: It is well known that all living organisms undergo a life cycle. For example when a seed is planted, it germinates, grows into a full tree, and finally dies. Based on this concept of a biological life cycle, the term software life cycle has been defined to imply the different stages (or phases) over which a software evolves from an initial customer request for it, to a fully developed software, and finally to a stage where it is no longer useful to any user, and then it is discarded.

Software Development life cycle

software life cycle model (also termed process model) is a pictorial and diagrammatic representation of the software life cycle. A life cycle model represents all the methods required to make a software product transit through its life cycle stages. SDLC is a systematic process for building software that ensures the quality and correctness of the software built. SDLC process aims to produce high-quality software that meets customer expectations. The system development should be complete in the pre-defined time frame and cost. SDLC consists of a detailed plan which explains how to plan, build, and maintain specific software. Every phase of the SDLC life cycle has its own process and deliverables that feed into the next phase. SDLC stands for Software Development Life Cycle and is also referred to as the Application Development life-cycle.

- It offers a basis for project planning, scheduling, and estimating
- Provides a framework for a standard set of activities and deliverables
- It is a mechanism for project tracking and control
- Increases visibility of project planning to all involved stakeholders of the development process
- Increased and enhance development speed
- Improved client relations
- Helps you to decrease project risk and project management plan overhead.

Requirement analysis--→ Feasibility study--→ Design--→ Coding--→ Testing--→ Installation/Deployment--→ Maintenance

- Phase 1: Requirement collection and analysis
- Phase 2: Feasibility study
- Phase 3: Design
- Phase 4: Coding
- Phase 5: Testing
- Phase 6: Installation/Deployment
- Phase 7: Maintenance

Phase 1: Requirement collection and analysis:

The requirement is the first stage in the SDLC process. It is conducted by the senior team members with inputs from all the stakeholders and domain experts in the industry. Planning for the quality assurance requirements and recognition of the risks involved is also done at this stage.

This stage gives a clearer picture of the scope of the entire project and the anticipated issues, opportunities, and directives which triggered the project.

Requirements Gathering stage need teams to get detailed and precise requirements. This helps companies to finalize the necessary timeline to finish the work of that system.

Phase 2: Feasibility study:

Once the requirement analysis phase is completed the next sdhc step is to define and document software needs. This process conducted with the help of 'Software Requirement Specification' document also known as 'SRS' document. It includes everything which should be designed and developed during the project life cycle.

There are mainly five types of feasibilities checks:

- **Economic:** Can we complete the project within the budget or not?
- **Legal:** Can we handle this project as cyber law and other regulatory framework/compliances.
- **Operation feasibility:** Can we create operations which is expected by the client?
- **Technical:** Need to check whether the current computer system can support the software
- **Schedule:** Decide that the project can be completed within the given schedule or not.

Phase 3: Design:

In this third phase, the system and software design documents are prepared as per the requirement specification document. This helps define overall system architecture. This design phase serves as input for the next phase of the model. There are two kinds of design documents developed in this phase:

High-Level Design (HLD)

- Brief description and name of each module
- An outline about the functionality of every module
- Interface relationship and dependencies between modules
- Database tables identified along with their key elements
- Complete architecture diagrams along with technology details

Low-Level Design (LLD)

- Functional logic of the modules

- Database tables, which include type and size
- Complete detail of the interface
- Addresses all types of dependency issues
- Listing of error messages
- Complete input and outputs for every module

Phase 4: Coding:

Once the system design phase is over, the next phase is coding. In this phase, developers start build the entire system by writing code using the chosen programming language. In the coding phase, tasks are divided into units or modules and assigned to the various developers. It is the longest phase of the Software Development Life Cycle process.

In this phase, Developer needs to follow certain predefined coding guidelines. They also need to use programming tools like compiler, interpreters, debugger to generate and implement the code.

Phase 5: Testing:

Once the software is complete, and it is deployed in the testing environment. The testing team starts testing the functionality of the entire system. This is done to verify that the entire application works according to the customer requirement.

During this phase, QA and testing team may find some bugs/defects which they communicate to developers. The development team fixes the bug and send back to QA for a re-test. This process continues until the software is bug-free, stable, and working according to the business needs of that system.

Phase 6: Installation/Deployment:

Once the software testing phase is over and no bugs or errors left in the system then the final deployment process starts. Based on the feedback given by the project manager, the final software is released and checked for deployment issues if any.

Phase 7: Maintenance:

Once the system is deployed, and customers start using the developed system, following 3 activities occur

- Bug fixing - bugs are reported because of some scenarios which are not tested at all
- Upgrade - Upgrading the application to the newer versions of the Software
- Enhancement - Adding some new features into the existing software

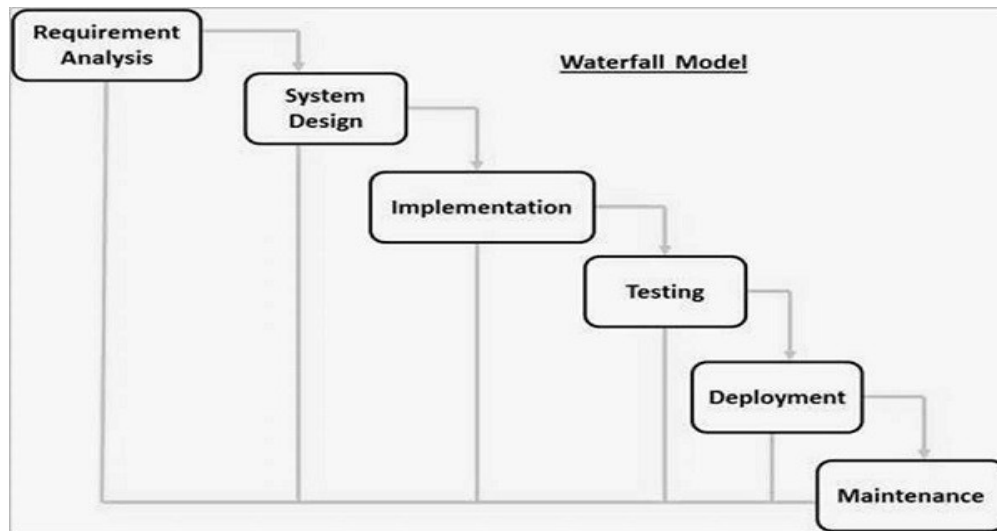
Iterative waterfall model:

The waterfall is a widely accepted SDLC model. In this approach, the whole process of the software development is divided into various phases of SDLC. In this SDLC model, the outcome of one phase acts as the input for the next phase.

This SDLC model is documentation-intensive, with earlier phases documenting what need be performed in the subsequent phases.

Waterfall approach was first SDLC Model to be used widely in Software Engineering to ensure success of the project. In "The Waterfall" approach, the whole process of software development is divided into separate phases. In this Waterfall model, typically, the outcome of one phase acts as the input for the next phase sequentially.

The following illustration is a representation of the different phases of the Waterfall Model.



The sequential phases in Waterfall model are –

- **Requirement Gathering and analysis** – All possible requirements of the system to be developed are captured in this phase and documented in a requirement specification document.
- **System Design** – The requirement specifications from first phase are studied in this phase and the system design is prepared. This system design helps in specifying hardware and system requirements and helps in defining the overall system architecture.
- **Implementation** – With inputs from the system design, the system is first developed in small programs called units, which are integrated in the next phase. Each unit is developed and tested for its functionality, which is referred to as Unit Testing.
- **Integration and Testing** – All the units developed in the implementation phase are integrated into a system after testing of each unit. Post integration the entire system is tested for any faults and failures.
- **Deployment of system** – Once the functional and non-functional testing is done; the product is deployed in the customer environment or released into the market.
- **Maintenance** – There are some issues which come up in the client environment. To fix those issues, patches are released. Also to enhance the product some better versions are released. Maintenance is done to deliver these changes in the customer environment.

Advantages and Disadvantages of Waterfall-Model

Advantages	Dis-Advantages
<ul style="list-style-type: none">• Before the next phase of development, each phase must be completed	<ul style="list-style-type: none">• Error can be fixed only during the phase
<ul style="list-style-type: none">• Suited for smaller projects where requirements are well defined	<ul style="list-style-type: none">• It is not desirable for complex project where requirement changes frequently
<ul style="list-style-type: none">• They should perform quality assurance test (Verification and Validation) before completing each stage.	<ul style="list-style-type: none">• Testing period comes quite late in the developmental process
<ul style="list-style-type: none">• Elaborate documentation is done at every phase of the software's development cycle	<ul style="list-style-type: none">• Documentation occupies a lot of time of developers and testers
<ul style="list-style-type: none">• Project is completely dependent on project team with minimum client intervention	<ul style="list-style-type: none">• Clients valuable feedback cannot be included with ongoing development phase
<ul style="list-style-type: none">• Any changes in software is made during the process of the development	<ul style="list-style-type: none">• Small changes or errors that arise in the completed software may cause a lot of problems

Prototyping Model:

Prototyping Model is a software development model in which prototype is built, tested, and reworked until an acceptable prototype is achieved. It also creates base to produce the final system or software. It works best in scenarios where the project's requirements are not known in detail. It is an iterative, trial and error method which takes place between developer and client.

Requirements→Quick Design→Build prototype→User Evaluation→Refining prototype→Implement and prototype

Step 1: Requirements gathering and analysis

A prototyping model starts with requirement analysis. In this phase, the requirements of the system are defined in detail. During the process, the users of the system are interviewed to know what is their expectation from the system.

Step 2: Quick design

The second phase is a preliminary design or a quick design. In this stage, a simple design of the system is created. However, it is not a complete design. It gives a brief idea of the system to the user. The quick design helps in developing the prototype.

Step 3: Build a Prototype

In this phase, an actual prototype is designed based on the information gathered from quick design. It is a small working model of the required system.

Step 4: Initial user evaluation

In this stage, the proposed system is presented to the client for an initial evaluation. It helps to find out the strength and weakness of the working model. Comment and suggestion are collected from the customer and provided to the developer.

Step 5: Refining prototype

If the user is not happy with the current prototype, you need to refine the prototype according to the user's feedback and suggestions.

This phase will not over until all the requirements specified by the user are met. Once the user is satisfied with the developed prototype, a final system is developed based on the approved final prototype.

Step 6: Implement Product and Maintain

Once the final system is developed based on the final prototype, it is thoroughly tested and deployed to production. The system undergoes routine maintenance for minimizing downtime and prevent large-scale failures.

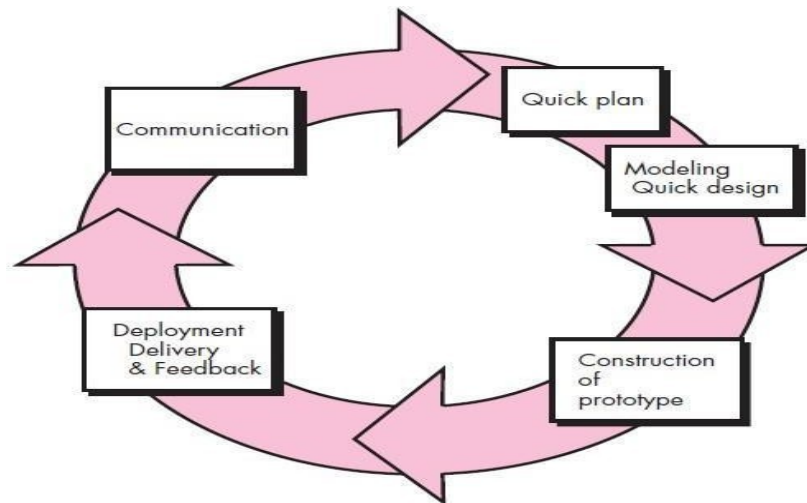


Fig : prototyping paradigm

Phases of prototyping Model

1. **Requirement Identification:** Here identification of product requirements is cleared in details. It is done through interview some product's future users and other members of the departments.
2. **Design Stage:** A first-round design is created in this stage for the new system.
3. **Build the Initial Prototype:** An initial prototype the target software is built from the original design. Working off all the product components may not be perfect or accurate. The first sample model is tailored as per the comments were given by the users and based on that the second prototype is built.
4. **Review of the Prototype:** After the product completes all the iterations of the update, it is presented to the customer or other stakeholders of the project. The response is accumulated in an organized way so that they can be used for further system enhancements.

5. **Iteration and Enhancement of Prototype:** Once the review of the product is done, it is set for further enhancement based on factors like - time, workforce as well as budget. Also, the technical feasibility of actual implementation is checked.

The final system is thoroughly evaluated and tested. Periodic maintenance is conceded on an ongoing basis for preventing large-scale breakdown as well as to minimize downtime.

There are four types of model available:

A) Rapid Throwaway Prototyping –

This technique offers a useful method of exploring ideas and getting customer feedback for each of them. In this method, a developed prototype need not necessarily be a part of the ultimately accepted prototype. Customer feedback helps in preventing unnecessary design faults and hence, the final prototype developed is of better quality.

B) Evolutionary Prototyping –

In this method, the prototype developed initially is incrementally refined on the basis of customer feedback till it finally gets accepted. In comparison to Rapid Throwaway Prototyping, it offers a better approach which saves time as well as effort. This is because developing a prototype from scratch for every iteration of the process can sometimes be very frustrating for the developers.

C) Incremental Prototyping – In this type of incremental Prototyping, the final expected product is broken into different small pieces of prototypes and being developed individually. In the end, when all individual pieces are properly developed, then the different prototypes are collectively merged into a single final product in their predefined order. It's a very efficient approach which reduces the complexity of the development process, where the goal is divided into sub-parts and each sub-part is developed individually. The time interval between the project begin and final delivery is substantially reduced because all parts of the system are prototyped and tested simultaneously.

D) Extreme Prototyping – This method is mainly used for web development. It consists of three sequential independent phases:

D.1) In this phase a basic prototype with all the existing static pages are presented in the HTML format.

D.2) In the 2nd phase, Functional screens are made with a simulate data process using a prototype services layer.

D.3) This is the final step where all the services are implemented and associated with the final prototype.

Advantages –

- The customers get to see the partial product early in the life cycle. This ensures a greater level of customer satisfaction and comfort.
- New requirements can be easily accommodated as there is scope for refinement.
- Missing functionalities can be easily figured out.
- Errors can be detected much earlier thereby saving a lot of effort and cost, besides enhancing the quality of the software.
- The developed prototype can be reused by the developer for more complicated projects in the future.
- Flexibility in design.

Disadvantages –

- Costly w.r.t time as well as money.
- There may be too much variation in requirements each time the prototype is evaluated by the customer.
- Poor Documentation due to continuously changing customer requirements.
- It is very difficult for developers to accommodate all the changes demanded by the customer.
- There is uncertainty in determining the number of iterations that would be required before the prototype is finally accepted by the customer.
- After seeing an early prototype, the customers sometimes demand the actual product to be delivered soon.
- Developers in a hurry to build prototypes may end up with sub-optimal solutions.
- The customer might lose interest in the product if he/she is not satisfied with the initial prototype.

Use –

The Prototyping Model should be used when the requirements of the product are not clearly understood or are unstable. It can also be used if requirements are changing quickly. This model can be successfully used for developing user interfaces, high technology software-intensive systems, and systems with complex algorithms and interfaces. It is also a very good choice to demonstrate the technical feasibility of the product.

EVOLUTIONARY MODEL:

The evolutionary approach is suitable for large problems which can be decomposed into a set of modules for incremental development and delivery. This model is also widely used for object-oriented development projects. Of course, this model can only be used if the incremental delivery of the system is acceptable to the customer.

An evolutionary approach lets the customer experiment with a working product much earlier than the monolithic approaches. Another important advantage of the incremental model is that it reduces the customer's trauma of getting used to an entirely new system. The gradual introduction of the product via incremental phases provides time to the customer to adjust to the new product. Also, from the customer's financial viewpoint, incremental development does not require a large upfront capital outlay. The customer can order the incremental versions as and when he can afford them.

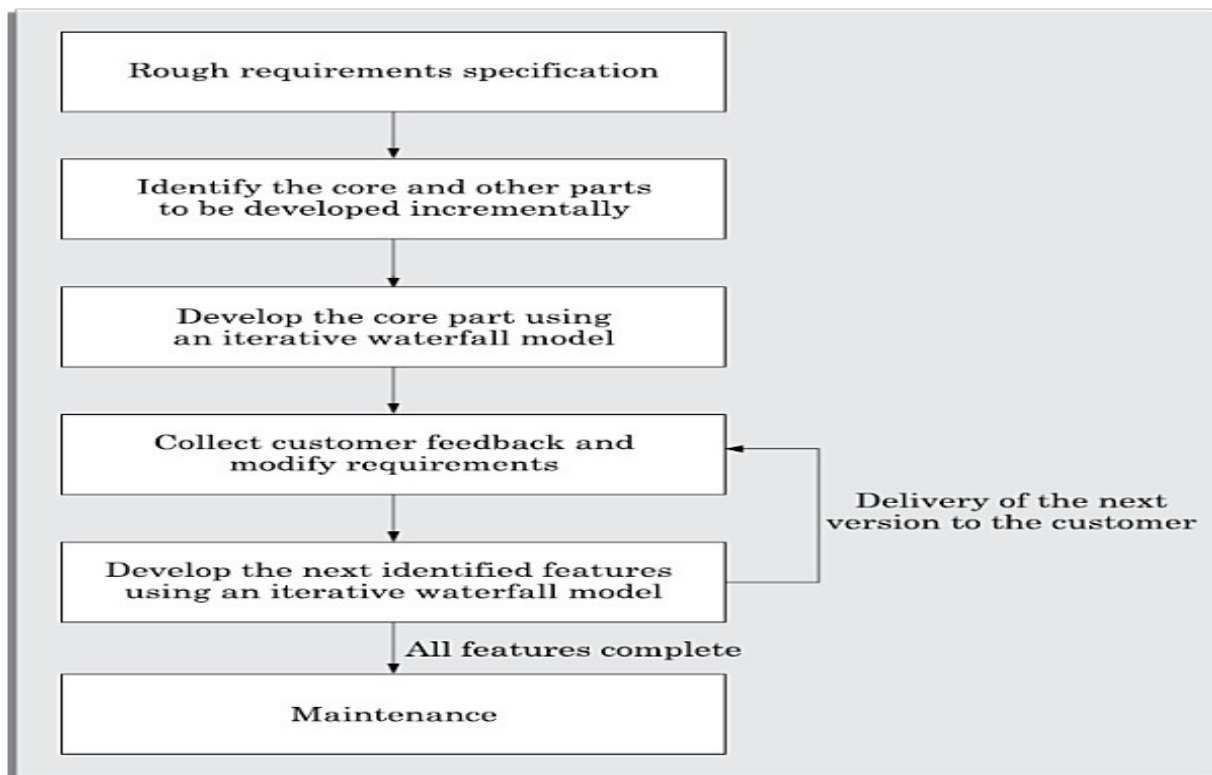
It is also called successive versions model or incremental model. At first, a simple working model is built. Subsequently it undergoes functional improvements & we keep on adding new functions till the desired system is built. Applications:

- ⌚ Large projects where you can easily find modules for incremental implementation. Often used when the customer wants to start using the core features rather than waiting for the full software.
- ⌚ Also used in object oriented software development because the system can be easily portioned into units in terms of objects.

Advantages: 1. User gets a chance to experiment partially developed system

2. Reduce the error because the core modules get tested thoroughly.

Disadvantages: It is difficult to divide the problem into several versions that would be acceptable to the customer which can be incrementally implemented & delivered.



This model has many of the features of the incremental model. As in case of the incremental model, the software is developed over a number of increments. At each increment, a concept (feature) is implemented and is deployed at the client site. The software successively refined and feature-enriched until the full software is realised. The principal idea behind the evolutionary life cycle model is conveyed by its name. In the incremental development model, complete requirements are first developed and the SRS document prepared. In contrast, in the evolutionary model, the requirements, plan, estimates, and solution evolve over the iterations, rather than fully defined and frozen in a major up-front specification effort before the development iterations begin. Such evolution is consistent with the pattern of unpredictable feature discovery and feature changes that take place in new product development.

Though the evolutionary model can also be viewed as an extension of the waterfall model, but it incorporates a major paradigm shift that has been widely adopted in many recent life cycle models. Due to obvious reasons, the evolutionary software development process is sometimes referred to as *design a little, build a little, test a little, deploy a little* model. This means that after the requirements have been specified, the design, build, test, and deployment activities are iterated.

Advantages

The evolutionary model of development has several advantages. Two important advantages of using this model are the following:

- **Effective elicitation of actual customer requirements:** In this model, the user gets a chance to experiment with a partially developed software much before the complete requirements are developed. Therefore, the evolutionary model helps to accurately elicit user requirements with the help of feedback obtained on the delivery of different versions of the software. As a result, the change requests after delivery of the complete software gets substantially reduced.

- **Easy handling change requests:** In this model, handling change requests is easier as no long term plans are made. Consequently, reworks required due to change requests are normally much smaller compared to the sequential models.

Disadvantages: The main disadvantages of the successive versions model are as follows

- **Feature division into incremental parts can be non-trivial:** For many development projects, especially for small-sized projects, it is difficult to divide the required features into several parts that can be incrementally implemented and delivered. Further, even for larger problems, often the features are so intertwined and dependent on each other that even an expert would need considerable effort to plan the incremental deliveries.
- **Ad hoc design:** Since at a time design for only the current increment is done, the design can become ad hoc without specific attention being paid to maintainability and optimality. Obviously, for moderate sized problems and for those for which the customer requirements are clear, the iterative waterfall model can yield a better solution.

The evolutionary model is well-suited to use in object-oriented software development projects.

Evolutionary model is appropriate for object-oriented development project, since it is easy to partition the software into stand alone units in terms of the classes. Also, classes are more or less self contained units that can be developed independently.

The Spiral Model : Originally proposed by **Barry Boehm**, the spiral model is an evolutionary software process model that couples the iterative nature of prototyping with the controlled and systematic aspects of the waterfall model. It provides the potential for rapid development of increasingly more complete versions of the software. Boehm describes the model in the following manner.

The spiral development model is a **risk-driven process model** generator that is used to **guide multi-stakeholder concurrent engineering** of software intensive systems. It has **two** main distinguishing features. One is a **cyclic approach** for incrementally growing a system's degree of definition and implementation while decreasing its degree of risk. The other is a set of **anchor point milestones** for ensuring stakeholder commitment to feasible and mutually satisfactory system solutions.

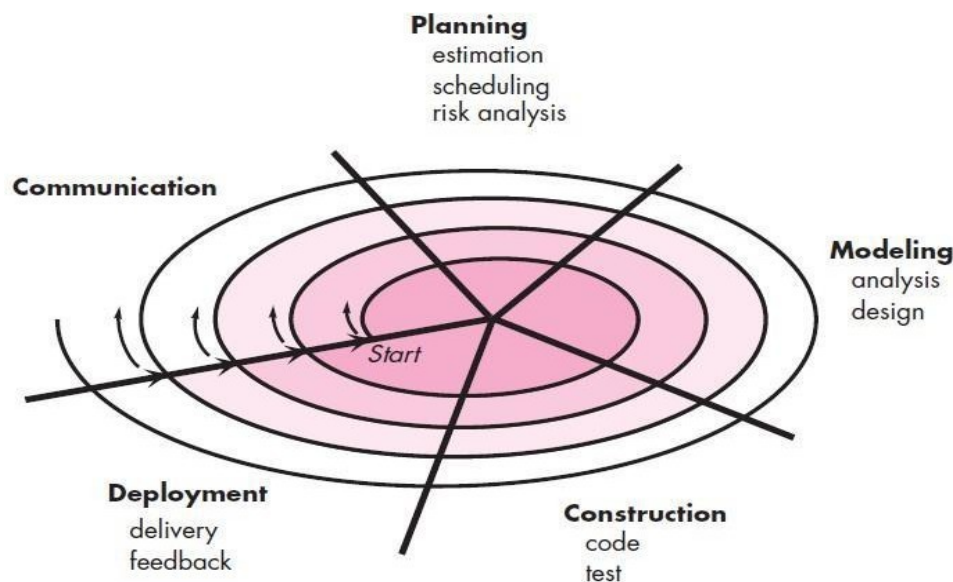


Fig : The Spiral Model

First quadrant (Objective Setting)

- During the first quadrant, it is needed to identify the objectives of the phase.
- Examine the risks associated with these objectives.

Second Quadrant (Risk Assessment and Reduction)

- A detailed analysis is carried out for each identified project risk.
- Steps are taken to reduce the risks. For example, if there is a risk that the requirements are inappropriate, a prototype system may be developed.

Third Quadrant (Development and Validation)

- Develop and validate the next level of the product after resolving the identified risks.

Fourth Quadrant (Review and Planning)

- Review the results achieved so far with the customer and plan the next iteration around the spiral. Progressively more complete version of the software gets built with each iteration around the spiral.

Using the spiral model, software is developed in a series of evolutionary releases. During early iterations, the release might be a model or prototype. During later iterations, increasingly more complete versions of the engineered system are produced.

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A spiral model is divided into a set of **framework activities** defined by the software engineering team. As this evolutionary process begins, the software team performs activities that are implied by a circuit around the spiral in a **clockwise** direction, beginning at the **center**. Risk is considered as each revolution is made. **Anchor point milestones** are a combination of work products and conditions that are attained along the path of the spiral are noted for each evolutionary pass.

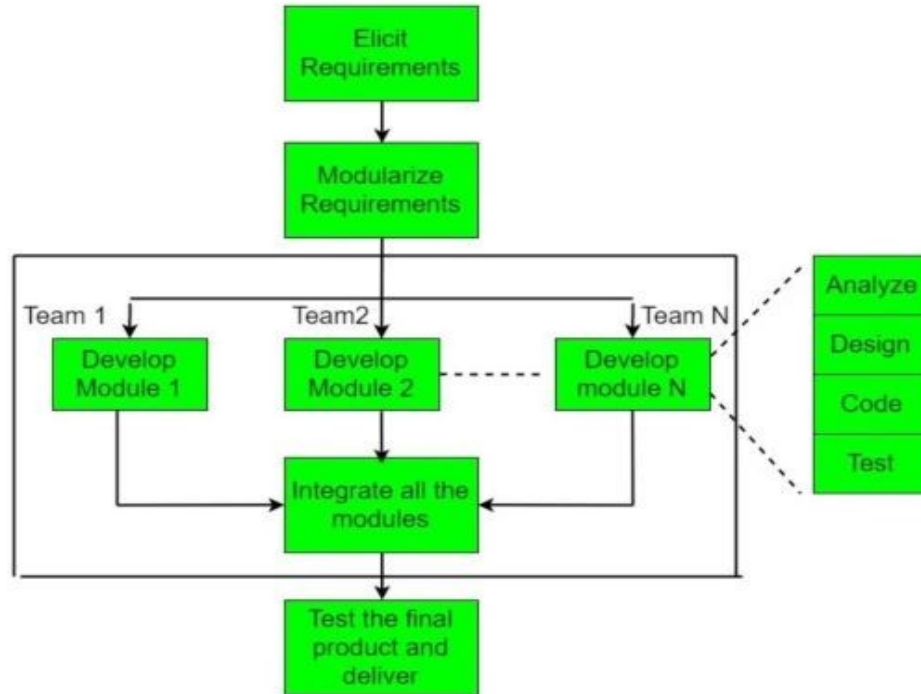
The first circuit around the spiral might result in the development of a **product** specification; subsequent passes around the spiral might be used to develop a **prototype** and then progressively more sophisticated versions of the software. Each pass through the planning region results in adjustments to the project plan. The spiral model can be adapted to apply throughout the life of the computer software. Therefore, the first circuit around the spiral might represent a “**concept development project**” that starts at the core of the spiral and continues for multiple iterations until concept development is complete. The new product will evolve through a number of iterations around the spiral. Later, a circuit around the spiral might be used to represent a “**product enhancement project**.”

The spiral model is a **realistic approach** to the development of **large-scale systems** and software. Because software evolves as the process progresses, the developer and customer better understand and react to risks at each evolutionary level. It maintains the systematic stepwise approach suggested by the classic life cycle but incorporates it into an iterative framework that more realistically reflects the real world.

RAD (Rapid Application Development) Model:

The Rapid Application Development Model was first proposed by IBM in 1980's. The critical feature of this model is the use of powerful development tools and techniques. A software project can be implemented using this model if the project can be broken down into small modules wherein each module can be assigned independently to separate teams. These modules can finally be combined to form the final product.

Development of each module involves the various basic steps as in waterfall model i.e analyzing, designing, coding and then testing, etc. as shown in the figure. Another striking feature of this model is a short time span i.e the time frame for delivery (time-box) is generally 60-90 days.



The use of powerful developer tools such as JAVA, C++, Visual BASIC, XML, etc. is also an integral part of the projects.

This model consists of 4 basic phases:

1. Requirements Planning –It involves the use of various techniques used in requirements elicitation like brainstorming, task analysis, form analysis, user scenarios, FAST (Facilitated Application Development Technique), etc. It also consists of the entire structured plan describing the critical data, methods to obtain it and then processing it to form final refined model.
2. User Description –This phase consists of taking user feedback and building the prototype using developer tools. In other words, it includes re-examination and validation of the data collected in the first phase. The dataset attributes are also identified and elucidated in this phase.
3. Construction –In this phase, refinement of the prototype and delivery takes place. It includes the actual use of powerful automated tools to transform process and data models into the final working product. All the required modifications and enhancements are too done in this phase.
4. Cutover –All the interfaces between the independent modules developed by separate teams have to be tested properly. The use of powerfully automated tools and subparts makes testing easier. This is followed by acceptance testing by the user.

The process involves building a rapid prototype, delivering it to the customer and the taking feedback. After validation by the customer, SRS document is developed and the design is finalised.

Advantages –

- Use of reusable components helps to reduce the cycle time of the project.
- Feedback from the customer is available at initial stages.
- Reduced costs as fewer developers are required.
- Use of powerful development tools results in better quality products in comparatively shorter time spans.
- The progress and development of the project can be measured through the various stages.
- It is easier to accommodate changing requirements due to the short iteration time spans.

Disadvantages –

- The use of powerful and efficient tools requires highly skilled professionals.
- The absence of reusable components can lead to failure of the project.
- The team leader must work closely with the developers and customers to close the project in time.
- The systems which cannot be modularized suitably cannot use this model.
- Customer involvement is required throughout the life cycle.
- It is not meant for small scale projects as for such cases, the cost of using automated tools and techniques may exceed the entire budget of the project.

Applications –

- 1.This model should be used for a system with known requirements and requiring short development time.
- 2.It is also suitable for projects where requirements can be modularized and reusable components are also available for development.
- 3.The model can also be used when already existing system components can be used in developing a new system with minimum changes.
- 4.This model can only be used if the teams consist of domain experts. This is because relevant knowledge and ability to use powerful techniques is a necessity.
- 5.The model should be chosen when the budget permits the use of automated tools and techniques required.

Agile Model

In earlier days Iterative Waterfall model was very popular to complete a project. But nowadays developers face various problems while using it to develop a software. The main difficulties included handling change requests from customers during project development and the high cost and time required to incorporate these changes. To overcome these drawbacks of Waterfall model, in the mid-1990s the Agile Software Development model was proposed. "**Agile process model**" refers to a software development approach based on iterative development. Agile methods break tasks into smaller iterations, or parts do not directly involve long term planning. The project scope and requirements are laid down at the beginning of the development process. Plans regarding the number of iterations, the duration and the scope of each iteration are clearly defined in advance.

Each iteration is considered as a short time "frame" in the Agile process model, which typically lasts from one to four weeks. The division of the entire project into smaller parts helps to minimize the project risk and to reduce the overall project delivery time requirements. Each iteration involves a team working through a full software development life cycle including planning, requirements analysis, design, coding, and testing before a working product is demonstrated to the client.

Phases of Agile Model: Following are the phases in the Agile model are as follows:

- 1.Requirements gathering
- 2.Design the requirements
- 3.Construction/ iteration
- 4.Testing/ Quality assurance
- 5.Deployment
- 6.Feedback

1. Requirements gathering: In this phase, you must define the requirements. You should explain business opportunities and plan the time and effort needed to build the project. Based on this information, you can evaluate technical and economic feasibility.

2. Design the requirements: When you have identified the project, work with stakeholders to define requirements. You can use the user flow diagram or the high-level UML diagram to show the work of new features and show how it will apply to your existing system.

3. Construction/ iteration: When the team defines the requirements, the work begins. Designers and developers start working on their project, which aims to deploy a working product. The product will undergo various stages of improvement, so it includes simple, minimal functionality.

4. Testing: In this phase, the Quality Assurance team examines the product's performance and looks for the bug.

5. Deployment: In this phase, the team issues a product for the user's work environment.

6. Feedback: After releasing the product, the last step is feedback. In this, the team receives feedback about the product and works through the feedback.

Agility principles for those who want to achieve agility:

1. Our highest priority is to satisfy the customer through early and continuous delivery of valuable software.
2. Welcome changing requirements, even late in development. Agile processes harness change for the customer's competitive advantage.
3. Deliver working software frequently, from a couple of weeks to a couple of months, with a preference to the shorter timescale.
4. Business people and developers must work together daily throughout the project.
5. Build projects around motivated individuals. Give them the environment and support they need, and trust them to get the job done.
6. The most efficient and effective method of conveying information to and within a development team is face-to-face conversation.
7. Working software is the primary measure of progress.
8. Agile processes promote sustainable development. The sponsors, developers, and users should be able to maintain a constant pace indefinitely.
9. Continuous attention to technical excellence and good design enhances agility.
10. Simplicity—the art of maximizing the amount of work not done—is essential.
11. The best architectures, requirements, and designs emerge from self-organizing teams.

12. At regular intervals, the team reflects on how to become more effective, then tunes and adjusts its behavior accordingly.

Human Factors: Agile development focuses on the talents and skills of individuals, molding the process to specific people and teams.” The key point in this statement is that *the process molds to the needs of the people and team*

1.Competence. In an agile development context, “competence” encompasses innate talent, specific software-related skills, and overall knowledge of the process that the team has chosen to apply. Skill and knowledge of process can and should be taught to all people who serve as agile team members.

2.Common focus. Although members of the agile team may perform different tasks and bring different skills to the project, all should be focused on one goal—to deliver a working software increment to the customer within the time promised. To achieve this goal, the team will also focus on continual adaptations (small and large) that will make the process fit the needs of the team.

3.Collaboration. Software engineering (regardless of process) is about assessing, analyzing, and using information that is communicated to the software team; creating information that will help all stakeholders understand the work of the team; and building information (computer software and relevant databases) that provides business value for the customer. To accomplish these tasks, team members must collaborate—with one another and all other stakeholders.

4.Decision-making ability. Any good software team (including agile teams) must be allowed the freedom to control its own destiny. This implies that the team is given autonomy—decision-making authority for both technical and project issues.

5.Fuzzy problem-solving ability. Software managers must recognize that the agile team will continually have to deal with ambiguity and will continually be buffeted by change.

6.Mutual trust and respect. The agile team must become what DeMarco and Lister call a “jelled” team. A jelled team exhibits the trust and respect that are necessary to make them “so strongly knit that the whole is greater than the sum of the parts.”

7.Self-organization. In the context of agile development, self-organization implies **three** things:

1. The agile team organizes itself for the work to be done,
2. The team organizes the process to best accommodate its local environment
- 3.The team organizes the work schedule to best achieve delivery of the software increment. Self-organization has a number of technical benefits, but more importantly, it serves to improve collaboration and boost team morale.

1.EXTREME PROGRAMMING (XP):

Extreme Programming (XP), the most widely used approach to agile software development, emphasizes business results first and takes an incremental, get-something-started approach to building the product, using continual testing and revision.

1.1 XP Values

Beck defines a set of **five values** that establish a foundation for all work performed as part of XP—**communication, simplicity, feedback, courage, and respect**. Each of these values is used as a driver for specific XP **activities, actions, and tasks**.

Communication :In order to achieve effective **communication** between software engineers and other stakeholders, XP emphasizes close, yet informal collaboration between customers and developers, the establishment of effective metaphors³ for communicating important concepts, continuous feedback, and the avoidance of voluminous documentation as a communication medium.

Simplicity :To achieve **simplicity**, XP restricts developers to design only for immediate needs, rather than consider future needs. The intent is to create a simple design that can be easily implemented in code). If the design must be improved, it can be *refactored* at a later time.

Feedback: is derived from three sources: the implemented software itself, the customer, and other software team members. By designing and implementing an effective testing strategy the software provides the agile team with

feedback. XP makes use of the **unit test** as its primary testing tactic. As each class is developed, the team develops a unit test to exercise each operation according to its specified functionality.

Courage : Beck argues that strict adherence to certain XP practices demands **courage**. A better word might be **discipline**. An agile XP team must have the discipline (courage) to design for today, recognizing that future requirements may change dramatically, thereby demanding substantial rework of the design and implemented code.

Respect:By following each of these values, the agile team inculcates **respect** among its members, between other stakeholders and team members, and indirectly, for the software itself. As they achieve successful delivery of software increments, the team develops growing respect for the XP process.

1.2 The XP Process

Extreme Programming uses an object-oriented approach as its preferred development paradigm and encompasses a set of rules and practices that occur within the context of four framework activities: planning, design, coding, and testing. Following figure illustrates the XP process and notes some of the key ideas and tasks that are associated with each framework activity.

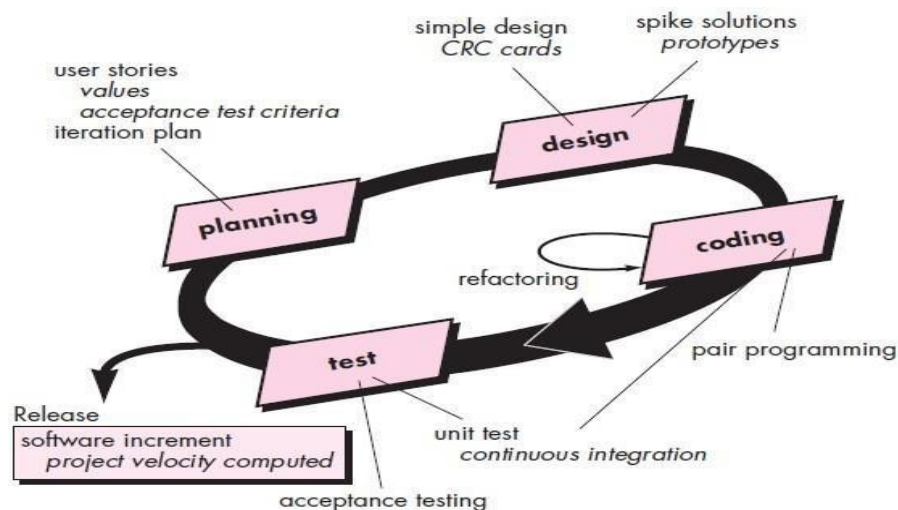


Fig : The Extreme Programming process

- **Planning**:The planning activity (also called *the planning game*) begins with *listening*—a requirements gathering activity that enables the technical members of the XP team to understand the business context for the software and to get a broad feel for required output and major features and functionality.
- **Design**:XP design rigorously follows the KIS (keep it simple) principle. A simple design is always preferred over a more complex representation. In addition, the design provides implementation guidance for a story as it is written—nothing less, nothing more. The design of extra functionality If a difficult design problem is encountered as part of the design of a story, XP recommends the immediate creation of an operational prototype of that portion of the design. Called a **spike solution**, the design prototype is implemented and evaluated. XP encourages **refactoring**—a construction technique that is also a method for design optimization.
- **refactoring** in the following manner: Refactoring is the process of changing a software system in such a way that it does not alter the external behavior of the code yet improves the internal structure. It is a disciplined way to clean up code [that minimizes the chances of introducing bugs].
- **Coding**. After stories are developed and preliminary design work is done, the team does *not* move to code, but rather develops a series of unit tests that will exercise each of the stories that is to be included in the current release Once the code is complete, it can be unit-tested immediately, thereby providing instantaneous feedback to the developers.

- A key concept during the coding activity is *pair programming*. XP recommends that two people work together at one computer workstation to create code for a story. This provides a mechanism for real time problem solving (two heads are often better than one) and real-time quality assurance.
- **Testing.** The creation of unit tests before coding commences is a key element of the XP approach. The unit tests that are created should be implemented using a framework that enables them to be automated. This encourages a regression testing strategy whenever code is modified. As the individual unit tests are organized into a “universal testing suite” integration and validation testing of the system can occur on a daily basis. This provides the XP team with a continual indication of progress and also can raise warning flags early if things go awry. Wells states: “Fixing small problems every few hours takes less time than fixing huge problems just before the deadline.”
- **XP acceptance tests**, also called **customer tests**, are specified by the customer and focus on overall system features and functionality that are visible and reviewable by the customer. Acceptance tests are derived from user stories that have been implemented as part of a software release.

1.3 Industrial XP:

Industrial Extreme Programming (IXP) in the following manner: “IXP is an organic evolution of XP. It is imbued with XP’s minimalist, customer-centric, test- driven spirit. IXP differs most from the original XP in its greater inclusion of management, its expanded role for customers, and its upgraded technical practices.” IXP incorporates six new practices that are designed to help ensure that an XP project works successfully for significant projects within a large organization.

- **Readiness assessment.** Prior to the initiation of an IXP project, the organization should conduct a *readiness assessment*. The assessment ascertains whether (1) an appropriate development environment exists to support IXP, (2) the team will be populated by the proper set of stakeholders, (3) the organization has a distinct quality program and supports continuous improvement, (4) the organizational culture will support the new values of an agile team, and (5) the broader project community will be populated appropriately.
- **Project community.** Classic XP suggests that the right people be used to populate the agile team to ensure success. The implication is that people on the team must be well- trained, adaptable and skilled, and have the proper temperament to contribute to a self- organizing team. When XP is to be applied for a significant project in a large organization, the concept of the “team” should morph into that of a *community*. In IXP, the community members and their roles should be explicitly defined and mechanisms for communication and coordination between community members should be established.
- **Project chartering.** The IXP team assesses the project itself to determine whether an appropriate business justification for the project exists and whether the project will further the overall goals and objectives of the organization. Chartering also examines the context of the project to determine how it complements, extends, or replaces existing systems or processes.
- **Test-driven management.** An IXP project requires measurable criteria for assessing the state of the project and the progress that has been made to date. Test-driven management establishes a series of measurable “destinations” and then defines mechanisms for determining whether or not these destinations have been reached.
- **Retrospectives.** An IXP team conducts a specialized technical review after a software increment is delivered. Called a *retrospective*, the review examines “issues, events, and lessons-learned” across a software increment and/or the entire software release. The intent is to improve the IXP process.
- **Continuous learning.** Because learning is a vital part of continuous process improvement, members of the XP team are encouraged (and possibly, incited) to learn new methods and techniques that can lead to a higher quality product.

1.4 THE XP DEBATE:

All new process models and methods spur worthwhile discussion and in some instances heated debate. Extreme Programming has done both. In an interesting book that examines the efficacy of XP Stephens and Rosenberg argue that many XP practices are worthwhile, but others have been overhyped, and a few are problematic.

- **Requirements volatility.** Because the customer is an active member of the XP team, changes to requirements are requested informally. As a consequence, the scope of the project can change and earlier work may have to be modified

to accommodate current needs. Proponents argue that this happens regardless of the process that is applied and that XP provides mechanisms for controlling scope creep.

- Conflicting customer needs.** Many projects have multiple customers, each with his own set of needs. In XP, the team itself is tasked with assimilating the needs of different customers, a job that may be beyond their scope of authority.

- Requirements are expressed informally.** User stories and acceptance tests are the only explicit manifestation of requirements in XP. Critics argue that a more formal model or specification is often needed to ensure that omissions, inconsistencies, and errors are uncovered before the system is built.

- Lack of formal design.** XP deemphasizes the need for architectural design and in many instances, suggests that design of all kinds should be relatively informal. Critics argue that when complex systems are built, design must be emphasized to ensure that the overall structure of the software will exhibit quality and maintainability. XP proponents suggest that the incremental nature of the XP process limits complexity (simplicity is a core value) and therefore reduces the need for extensive design.

OTHER AGILE PROCESS MODELS

Other agile process models have been proposed and are in use across the industry.

- Adaptive Software Development (ASD)
- Scrum
- Dynamic Systems Development Method (DSDM)
- Crystal
- Feature Drive Development (FDD)
- Lean Software Development (LSD)
- Agile Modeling (AM)
- Agile Unified Process (AUP)

1.Adaptive Software Development (ASD)

Adaptive Software Development (ASD) has been proposed by Jim Highsmith as a technique for building complex software and systems. The philosophical underpinnings of ASD focus on human collaboration and team self-organization.

- High smith argues that an agile, adaptive development approach based on collaboration is “as much a source of *order* in our complex interactions as discipline and engineering.” He defines an ASD “life cycle” that incorporates three phases, speculation, collaboration, and learning.

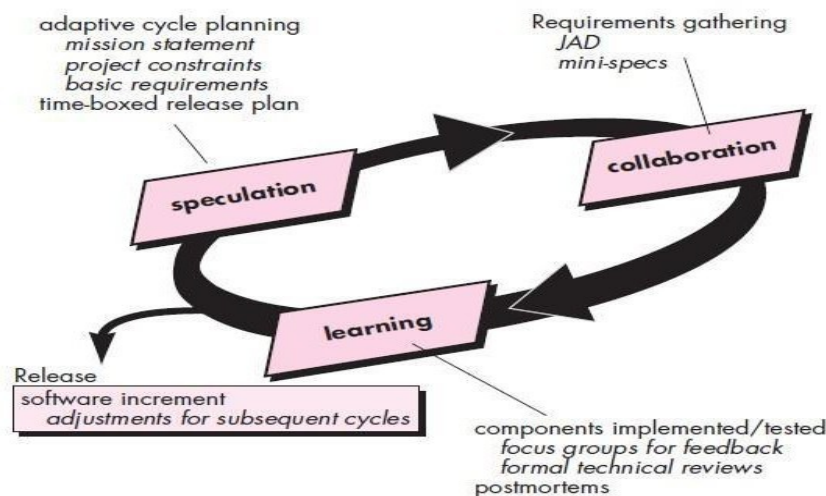


Fig : Adaptive software development

During **speculation**, the project is initiated and **adaptive cycle planning** is conducted. Adaptive cycle planning uses project initiation information—the customer’s mission statement, project constraints

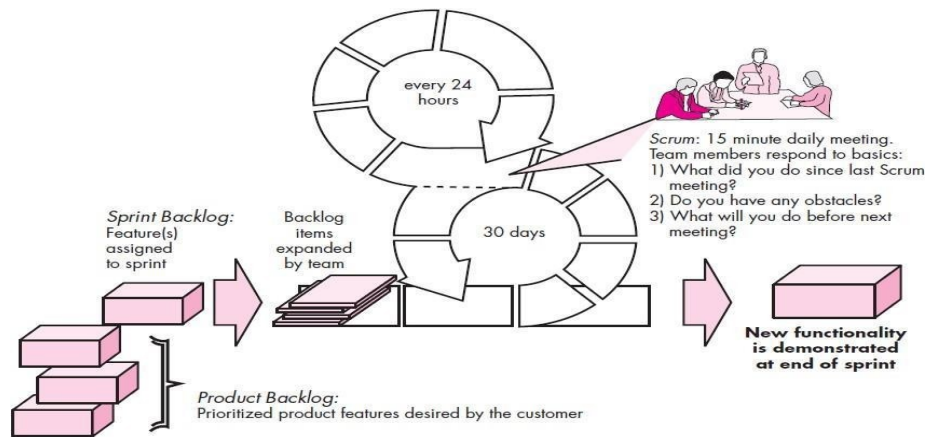
- motivated people use **collaboration** in a way that multiplies their talent and creative output beyond their absolute numbers. This approach is a recurring theme in all agile methods. But collaboration is not easy. It encompasses communication and teamwork, but it also emphasizes individualism, because individual creativity plays an important role in collaborative thinking.

- It is, above all, a matter of trust. People working together must trust one another to (1) criticize without animosity, (2) assist without resentment, (3) work as hard as or harder than they do, (4) have the skill set to contribute to the work at hand, and (5) communicate problems or concerns in a way that leads to effective action. As members of an ASD team begin to develop the components that are part of an adaptive cycle, the emphasis is on **“learning”** as much as it is on progress toward a completed cycle.

- ASD teams learn in **three** ways: **focus groups, technical reviews , and project postmortems**. ASD’s overall emphasis on the dynamics of self-organizing teams, interpersonal collaboration, and individual and team learning yield software project teams that have a much higher likelihood of success.

2.SCRUM

Scrum is an agile software development method that was conceived by Jeff Sutherland and his development team in the early 1990s. Scrum principles are consistent with the agile manifesto and are used to guide development activities within a process that incorporates the following framework activities: requirements, analysis, design, evolution, and delivery. Within each framework activity, work tasks occur within a process pattern called a **sprint**. The work conducted within a sprint is adapted to the problem at hand and is defined and often modified in real time by the Scrum team. The overall flow of the Scrum process is illustrated in following figure



- Scrum** emphasizes the use of a set of software process patterns that have proven effective for projects with tight timelines, changing requirements, and business criticality. Each of these process patterns defines a set of development actions:

- Backlog**—a prioritized list of project requirements or features that provide business value for the customer. Items can be added to the backlog at any time. The product manager assesses the backlog and updates priorities as required.

- Sprints**—consist of work units that are required to achieve a requirement defined in the backlog that must be fit into a predefined time-box (typically 30 days). Changes (e.g., backlog work items) are not introduced during the sprint. Hence, the sprint allows team members to work in a short-term, but stable environment.

- Scrum meetings**—are short (typically 15 minutes) meetings held daily by the Scrum team. Three key questions are asked and answered by all team members

- What did you do since the last team meeting?

- What obstacles are you encountering?

- What do you plan to accomplish by the next team meeting?

A team leader, called a **Scrum master**, leads the meeting and assesses the responses from each person. The Scrum meeting helps the team to uncover potential problems as early as possible. Also, these daily meetings lead to “**knowledge socialization**”

Demos—deliver the software increment to the customer so that functionality that has been implemented can be demonstrated and evaluated by the customer. It is important to note that the demo may not contain all planned functionality, but rather those functions that can be delivered within the time-box that was established.

3. Dynamic Systems Development Method (DSDM)

The *Dynamic Systems Development Method* (DSDM) is an agile software development approach that “provides a framework for building and maintaining systems which meet tight time constraints through the use of incremental prototyping in a controlled project environment” The DSDM philosophy is borrowed from a modified version of the **Pareto principle—80 percent of an application can be delivered in 20 percent of the time**. It would take to deliver the complete (100 percent) application. DSDM is an iterative software process in which each iteration follows the 80 percent rule. That is, only enough work is required for each increment to facilitate movement to the next increment. The remaining detail can be completed later when more business requirements are known or changes have been requested and accommodated.

The *DSDM life cycle* that defines **three** different iterative cycles, preceded by **two** additional life cycle activities:

- **Feasibility study**—establishes the basic business requirements and constraints associated with the application to be built and then assesses whether the application is a viable candidate for the DSDM process
- **Business study**—establishes the functional and information requirements that will allow the application to provide business value; also, defines the basic application architecture and identifies the maintainability requirements for the application.
- **Functional model iteration**—produces a set of incremental prototypes that demonstrate functionality for the customer.
- **Design and build iteration**—revisits prototypes built during **functional model iteration** to ensure that each has been engineered in a manner that will enable it to provide operational business value for end users. In some cases, functional model iteration and design and build iteration occur concurrently.
- **Implementation**—places the latest software increment into the operational environment. It should be noted that (1) the increment may not be 100 percent complete or (2) changes may be requested as the increment is put into place. In either case, DSDM development work continues by returning to the functional model iteration activity.

4 .CRYSTAL

Alistair Cockburn and Jim Highsmith created the *Crystal family of agile methods* in order to achieve a software development approach that puts a premium on “maneuverability” during what Cockburn characterizes as “a resource limited, cooperative game of invention and communication, with a primary goal of delivering useful, working software and a secondary goal of setting up for the next game”

- The Crystal family is actually a set of example agile processes that have been proven effective for different types of projects. The intent is to allow agile teams to select the member of the crystal family that is most appropriate for their project and environment.

5. FEATURE DRIVEN DEVELOPMENT (FDD)

Feature Driven Development (FDD) was originally conceived by Peter Coad and his colleagues as a practical process model for object-oriented software engineering. Stephen Palmer and John Felsing have extended and improved Coad’s work, describing an adaptive, agile process that can be applied to moderately sized and larger software projects.

Like other agile approaches, FDD adopts a philosophy that (1) emphasizes collaboration among people on an FDD team; (2) manages problem and project complexity using feature- based decomposition followed by the integration of software increments, and (3) communication of technical detail using verbal, graphical, and text-based means.

Agile Modeling suggests a wide array of “core” and “supplementary” modeling principles, those that make AM unique are :

- Model with a purpose.** A developer who uses AM should have a specific goal in mind before creating the model. Once the goal for the model is identified, the type of notation to be used and level of detail required will be more obvious.
- Use multiple models.** There are many different models and notations that can be used to describe software. Only a small subset is essential for most projects. AM suggests that to provide needed insight, each model should present a different aspect of the system and only those models that provide value to their intended audience should be used.
- Travel light.** As software engineering work proceeds, keep only those models that will provide long-term value and jettison the rest. Every work product that is kept must be maintained as changes occur. This represents work that slows the team down. Ambler notes that “Every time you decide to keep a model you trade-off agility for the convenience of having that information available to your team in an abstract manner
- Content is more important than representation.** Modeling should impart information to its intended audience. A syntactically perfect model that imparts little useful content is not as valuable as a model with flawed notation that nevertheless provides valuable content for its audience.
- Know the models and the tools you use to create them.** Understand the strengths and weaknesses of each model and the tools that are used to create it.
- Adapt locally.** The modeling approach should be adapted to the needs of the agile team.

8.AGILE UNIFIED PROCESS (AUP)

The Agile Unified Process (AUP) adopts a “serial in the large” and “iterative in the small” philosophy for building computer-based systems. By adopting the classic UP phased activities— inception, elaboration, construction, and transition—AUP provides a serial overlay that enables a team to visualize the overall process flow for a software project. However, within each of the activities, the team iterates to achieve agility and to deliver meaningful software increments to end users as rapidly as possible. **Each AUP iteration addresses the following activities.**

- Modeling.** UML representations of the business and problem domains are created.
 - Implementation.** Models are translated into source code.
 - Testing.** Like XP, the team designs and executes a series of tests to uncover errors and ensure that the source code meets its requirements.
 - Deployment.** Like the generic process activity deployment in this context focuses on the delivery of a software increment and the acquisition of feedback from end users.
 - Configuration and project management.** In the context of AUP, configuration management addresses change management, risk management, and the control of any persistent work products that are produced by the team. Project management tracks and controls the progress of the team and coordinates team activities.
- Environment management.** Environment management coordinates a process infrastructure that includes standards, tools, and other support technology available to the team.

Software project management :project planning

A theoretical knowledge of different project management techniques is certainly necessary to become a successful project manager. However, effective software project management frequently calls for good qualitative judgment and decision taking capabilities. In addition to having a good grasp of the latest software project management techniques such as cost estimation, risk management, configuration management, project managers need good communication skills and the ability get work done. However, some skills such as tracking and controlling the progress of the project, customer interaction, managerial presentations, and team building are largely acquired through experience. None the less, the importance of sound knowledge of the prevalent project management techniques cannot be overemphasized

Project planning:

Once a project is found to be feasible, software project managers undertake project planning. Project planning is undertaken and completed even before any development activity starts. Project planning consists of the following essential activities:

Project size: What will be problem complexity in terms of the effort and time required to develop the product?

Cost: How much is it going to cost to develop the project?

Duration: How long is it going to take to complete development?

Effort: How much effort would be required?

The effectiveness of the subsequent planning activities is based on the accuracy of these estimations.

- Scheduling manpower and other resources
- Staff organization and staffing plans
- Risk identification, analysis, and abatement planning
- Miscellaneous plans such as quality assurance plan, configuration management plan, etc.

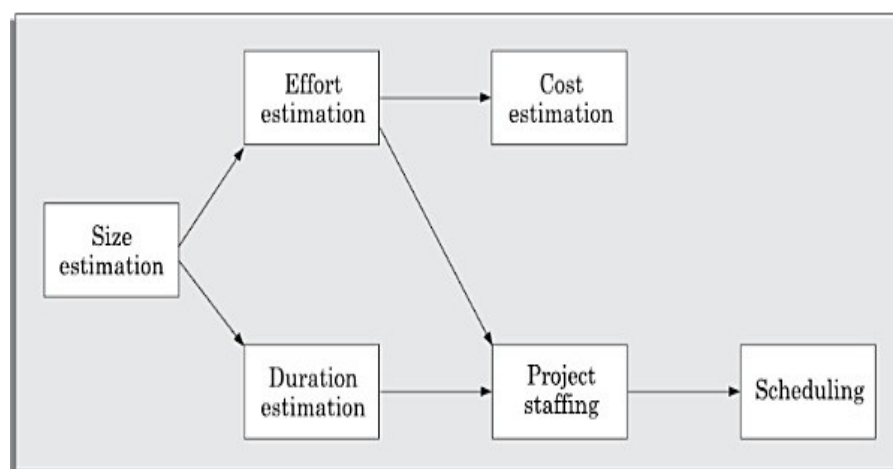


Fig:Precedence ordering among planning activities

Metrics for software project size estimation: Accurate estimation of the problem size is fundamental to satisfactory estimation of effort, time duration and cost of a software project. In order to be able to accurately estimate the project size, some important metrics should be defined in terms of which the project size can be expressed. The size of a problem is obviously not the number of bytes that the source code occupies. It is neither the byte size of the executable code. The project size is a measure of the problem complexity in terms of the effort and time required to develop the product. Currently two metrics are popularly being used widely to estimate size: lines of code (LOC) and function point (FP). The usage of each of these metrics in project size estimation has its own advantages and disadvantages.

1.Lines of Code (LOC): LOC is the simplest among all metrics available to estimate project size. This metric is very popular because it is the simplest to use. Using this metric, the project size is estimated by counting the number of source instructions in the developed program. Obviously, while counting the number of source instructions, lines used for commenting the code and the header lines should be ignored.

Determining the LOC count at the end of a project is a very simple job. However, accurate estimation of the LOC count at the beginning of a project is very difficult. In order to estimate the LOC count at the beginning of a project, project managers usually divide the problem into modules, and each module into submodules and so on, until the sizes of the different leaf-level modules can be approximately predicted. To be able to do this, past experience in developing similar products is helpful. By using the estimation of the lowest level modules, project managers arrive at the total size estimation.

2.Function point (FP): Function point metric was proposed by Albrecht [1983]. This metric overcomes many of the shortcomings of the LOC metric. Since its inception in late 1970s, function point metric has been slowly gaining popularity. One of the important advantages of using the function point metric is that it can be used to easily estimate the size of a software product directly from the problem specification. This is in contrast to the LOC metric, where the size can be accurately determined only after the product has fully been developed. The conceptual idea behind the function point metric is that the size of a software product is directly dependent on the number of different functions or features it supports. A software product supporting many features would certainly be of larger size than a product with less number of features. Each function when invoked reads some input data and transforms it to the corresponding output data.

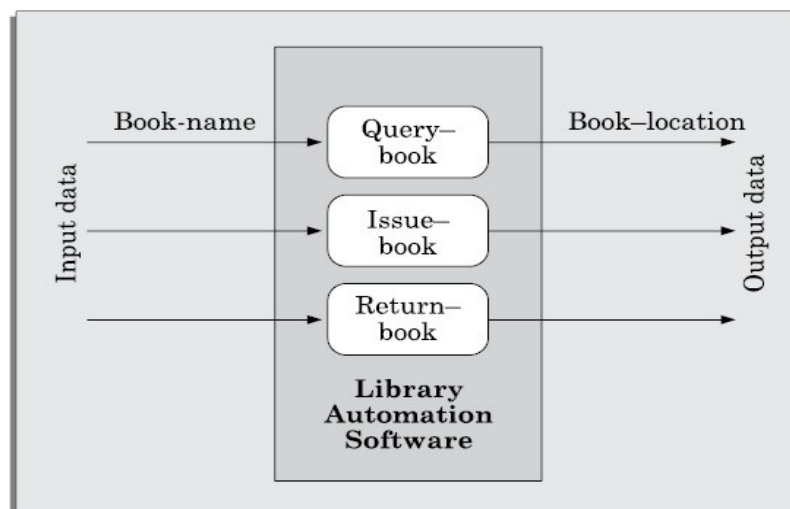


Fig: System function as a map of input data to output data

It is computed using the following three steps:

- **Step 1:** Compute the unadjusted function point (UFP) using a heuristic expression.
- **Step 2:** Refine UFP to reflect the actual complexities of the different parameters used in UFP computation.
- **Step 3:** Compute FP by further refining UFP to account for the specific characteristics of the project that can influence the entire development effort.

Step 1: UFP computation

Function point is computed in two steps. The first step is to compute the unadjusted function point (UFP).

$$\text{UFP} = (\text{Number of inputs}) * 4 + (\text{Number of outputs}) * 5 + (\text{Number of inquiries}) * 4 + (\text{Number of files}) * 10 + (\text{Number of interfaces}) * 10$$

Number of inputs: Each data item input by the user is counted. Data inputs should be distinguished from user inquiries. Inquiries are user commands such as print-account-balance. Inquiries are counted separately. It must be noted that individual data items input by the user are not considered in the calculation of the number of inputs, but a group of related inputs are considered as a single input.

For example, while entering the data concerning an employee to an employee pay roll software; the data items name, age, sex, address, phone number, etc. are together considered as a single input. All these data items can be considered to be related, since they pertain to a single employee.

Number of outputs: The outputs considered refer to reports printed, screen outputs, error messages produced, etc. While outputting the number of outputs the individual data items within a report are not considered, but a set of related data items is counted as one input.

Number of inquiries: Number of inquiries is the number of distinct interactive queries which can be made by the users. These inquiries are the user commands which require specific action by the system.

Number of files: Each logical file is counted. A logical file means groups of logically related data. Thus, logical files can be data structures or physical files.

Number of interfaces: Here the interfaces considered are the interfaces used to exchange information with other systems. Examples of such interfaces are data files on tapes, disks, communication links with other systems etc.

Step 2: Refine parameters

UFP computed at the end of step 1 is a gross indicator of the problem size. This UFP needs to be refined. This is possible, since each parameter (input, output, etc.) has been implicitly assumed to be of average complexity. However, this is rarely true. For example, some input values may be extremely complex, some very simple, etc. In order to take this issue into account, UFP is refined by taking into account the complexities of the parameters of UFP computation. The complexity of each parameter is graded into three broad categories—simple, average, or complex. The weights for the different parameters are determined based on the numerical values shown in Table 3.1. Based on these weights of the parameters, the parameter values in the UFP are refined. For example, rather than each input being computed as four FPs, very simple inputs are computed as three FPs and very complex inputs as six FPs

Table : Refinement of Function Point Entities

Type	Simple	Average	Complex
Input(I)	3	4	6
Output (O)	4	5	7
Inquiry (E)	3	4	6
Number of files (F)	7	10	15
Number of interfaces	5	7	10

Step 3: Refine UFP based on complexity of the overall project

In the final step, several factors that can impact the overall project size are considered to refine the UFP computed in step 2. Examples of such project parameters that can influence the project sizes include high transaction rates, response time requirements, scope for reuse, etc. Albrecht identified 14 parameters that can influence the development effort. Each of these 14 parameters is assigned a value from 0 (not present or no influence) to 6 (strong influence). The resulting numbers are summed, yielding the total degree of influence (DI). A technical complexity factor (TCF) for the project is computed and the TCF is multiplied with UFP to yield FP. The TCF expresses the overall impact of the corresponding project parameters on the development effort. TCF is computed as $(0.65 + 0.01 \times DI)$. As DI can vary from 0 to 84, TCF can vary from 0.65 to 1.49. Finally, FP is given as the product of UFP and TCF. That is, $FP = UFP \times TCF$.

Example : Determine the function point measure of the size of the following supermarket software. A supermarket needs to develop the following software to encourage regular customers. For this, the customer needs to supply his/her residence address, telephone number, and the driving license number. Each customer who registers for this scheme is assigned a unique customer number (CN) by the computer. Based on the generated CN, a clerk manually prepares a customer identity card after getting the market manager's signature on it. A customer can present his customer identity card to the check out staff when he makes any purchase. In this case, the value of his purchase is credited against his CN. At the end of each year, the supermarket intends to award surprise gifts to 10 customers who make the highest total purchase over the year. Also, it intends to award a 22 carat gold coin to every customer whose purchase exceeded Rs. 10,000. The entries against the CN are reset on the last day of every year after the prize winners' lists are generated. Assume that various project characteristics determining the complexity of software development to be average.

Answer:

Step 1: From an examination of the problem description, we find that there are two inputs, three outputs, two files, and no interfaces. Two files would be required, one for storing the customer details and another for storing the daily purchase records. Now, using equation ,we get:

$$UFP = 2 \times 4 + 3 \times 5 + 1 \times 4 + 10 \times 2 + 0 \times 10 = 47$$

Step 2: All the parameters are of moderate complexity, except the output parameter of customer registration, in which the only output is the CN value. Consequently, the complexity of the output parameter of the customer registration function can be categorized as simple. By consulting Table 3.1, we find that the value for simple output is given to be 4. The UFP can be refined as follows:

$$UFP = 3 \times 4 + 2 \times 5 + 1 \times 4 + 10 \times 2 + 0 \times 10 = 46$$

Therefore, the UFP will be 46.

Step 3: Since the complexity adjustment factors have average values, therefore the total degrees of influence would be: $DI = 14 \times 4 = 56$

$$TCF = 0.65 + 0.01 + 56 = 1.21$$

Therefore, the adjusted $FP = 46 \times 1.21 = 55.66$

Project Estimation techniques

Estimation of various project parameters is a basic project planning activity. The important project parameters that are estimated include: project size, effort required to develop the software, project duration, and cost. These estimates not only help in quoting the project cost to the customer, but are also useful in resource planning and scheduling. There are three broad categories of estimation techniques:

- Empirical estimation techniques
- Heuristic techniques
- Analytical estimation techniques

Empirical Estimation Techniques

Empirical estimation techniques are based on making an educated guess of the project parameters. While using this technique, prior experience with development of similar products is helpful. Although empirical estimation techniques are based on common sense, different activities involved in estimation have been formalized over the years. Two popular empirical estimation techniques are: Expert judgment technique and Delphi cost estimation.

Expert Judgment Technique

Expert judgment is one of the most widely used estimation techniques. In this approach, an expert makes an educated guess of the problem size after analyzing the problem thoroughly. Usually, the expert estimates the cost of the different components (i.e. modules or subsystems) of the system and then combines them to arrive at the overall estimate. However, this technique is subject to human errors and individual bias. Also, it is possible that the expert may overlook some factors inadvertently. Further, an expert making an estimate may not have experience and knowledge of all aspects of a project. For example, he may be conversant with the database and user interface parts but may not be very knowledgeable about the computer communication part.

A more refined form of expert judgment is the estimation made by group of experts. Estimation by a group of experts minimizes factors such as individual oversight, lack of familiarity with a particular aspect of a project, personal bias, and the desire to win contract through overly optimistic estimates. However, the estimate made by a group of experts may still exhibit bias on issues where the entire group of experts may be biased due to reasons such as political considerations. Also, the decision made by the group may be dominated by overly assertive members.

Delphi cost estimation: Delphi cost estimation approach tries to overcome some of the shortcomings of the expert judgment approach. Delphi estimation is carried out by a team comprising of a group of experts and a coordinator. In this approach, the coordinator provides each estimator with a copy of the software requirements specification (SRS) document and a form for recording his cost estimate. Estimators complete their individual estimates anonymously and submit to the coordinator. In their estimates, the estimators mention any unusual characteristic of the product which has influenced his estimation. The coordinator prepares and distributes the summary of the responses of all the estimators, and includes any unusual rationale noted by any of the estimators. Based on this summary, the estimators re-estimate. This process is iterated for several rounds.

However, no discussion among the estimators is allowed during the entire estimation process. The idea behind this is that if any discussion is allowed among the estimators, then many estimators may easily get influenced by the rationale of an estimator who may be more experienced or senior. After the completion of several iterations of estimations, the coordinator takes the responsibility of compiling the results and preparing the final estimate.

Heuristic Techniques

Heuristic techniques assume that the relationships among the different project parameters can be modeled using suitable mathematical expressions. Once the basic (independent) parameters are known, the other (dependent) parameters can be easily determined by substituting the value of the basic parameters in the mathematical expression. Different heuristic estimation models can be divided into the following two classes: single variable model and the multi variable model.

Single variable estimation models provide a means to estimate the desired characteristics of a problem, using some previously estimated basic (independent) characteristic of the software product such as its size. A single variable estimation model takes the following form:

$$\text{Estimated Parameter} = c_1 * e^{d_1}$$

In the above expression, e is the characteristic of the software which has already been estimated (independent variable). *Estimated Parameter* is the dependent parameter to be estimated. The dependent parameter to be estimated could be effort, project duration, staff size, etc. c_1 and d_1 are constants. The values of the constants c_1 and d_1 are usually determined using data collected from past projects (historical data). The basic COCOMO model is an example of single variable cost estimation model.

A multivariable cost estimation model takes the following form:

$$\text{Estimated Resource} = c_1 * e_1^{d_1} + c_2 * e_2^{d_2} + \dots$$

Where e_1, e_2, \dots are the basic (independent) characteristics of the software already estimated, and $c_1, c_2, d_1, d_2, \dots$ are constants. Multivariable estimation models are expected to give more accurate estimates compared to the single variable models, since a project parameter is typically influenced by several independent parameters. The independent parameters influence the dependent parameter to different extents. This is modeled by the constants $c_1, c_2, d_1, d_2, \dots$. Values of these constants are usually determined from historical data. The intermediate COCOMO model can be considered to be an example of a multivariable estimation model.

Analytical Estimation Techniques

Analytical estimation techniques derive the required results starting with basic assumptions regarding the project. Thus, unlike empirical and heuristic techniques, analytical techniques do have scientific basis. Halstead's software science is an example of an analytical technique. Halstead's software science can be used to derive some interesting results starting with a few simple assumptions. Halstead's software science is especially useful for estimating software maintenance efforts. In fact, it outperforms both empirical and heuristic techniques when used for predicting software maintenance efforts.

Three basic classes of software development projects

Boehm postulated that any software development project can be classified into one of the following three categories based on the development complexity: organic, semidetached, and embedded. In order to classify a product into the identified categories, Boehm not only considered the characteristics of the product but also those of the development team and development environment. Roughly speaking, these three product classes correspond to application, utility and system programs, respectively. Normally, data processing programs are considered to be application programs. Compilers, linkers, etc., are utility programs. Operating systems and real-time system programs, etc. are system programs. System programs interact directly with the hardware and typically involve meeting timing constraints and concurrent processing.

Boehm's [1981] definition of organic, semidetached, and embedded systems are elaborated below.

Person-month (PM) is considered to be an appropriate unit for measuring effort, because developers are typically assigned to a project for a certain number of months.

Organic: A development project can be considered of organic type, if the project deals with developing a well understood application program, the size of the development team is reasonably small, and the team members are experienced in developing similar types of projects.

Semidetached: A development project can be considered of semidetached type, if the development consists of a mixture of experienced and inexperienced staff. Team members may have limited experience on related systems but may be unfamiliar with some aspects of the system being developed.

Embedded: A development project is considered to be of embedded type, if the software being developed is strongly coupled to complex hardware, or if the stringent regulations on the operational procedures exist.

COCOMO

COCOMO (Constructive Cost Estimation Model) was proposed by Boehm [1981]. According to Boehm, software cost estimation should be done through three stages: Basic COCOMO, Intermediate COCOMO, and Complete COCOMO.

Basic COCOMO Model

The basic COCOMO model gives an approximate estimate of the project parameters. The basic COCOMO estimation model is given by the following expressions:

$$\text{Effort} = a_1 \times (\text{KLOC})^{a_2} \text{ PM}$$

$$\text{Tdev} = b_1 \times (\text{Effort})^{b_2} \text{ months}$$

- KLOC is the estimated size of the software product expressed in Kilo Lines Of Code.
- a_1, a_2, b_1, b_2 are constants for each category of software product.
- Tdev is the estimated time to develop the software, expressed in months

- Effort is the total effort required to develop the software product, expressed in person- months (PMs).

Estimation of development effort: For the three classes of software products, the formulas for estimating the effort based on the code size are shown below:

Organic : Effort = $2.4(KLOC)^{1.05}$ PM

Semi-detached : Effort = $3.0(KLOC)^{1.12}$ PM

Embedded : Effort = $3.6(KLOC)^{1.20}$ PM

Estimation of development time: For the three classes of software products, the formulas for estimating the development time based on the effort are given below:

Organic : Tdev = $2.5(Effort)^{0.38}$ Months

Semi-detached : Tdev = $2.5(Effort)^{0.35}$ Months

Embedded : Tdev = $2.5(Effort)^{0.32}$ Months

Example: Assume that the size of an organic type software product has been estimated to be 32,000 lines of source code. Assume that the average salary of a software developer is Rs. 15,000 per month. Determine the effort required to develop the software product, the nominal development time, and the cost to develop the product.

From the basic COCOMO estimation formula for organic software:

Effort = $2.4 \times (32)^{1.05} = 91$ PM

Nominal development time = $2.5 \times (91)^{0.38} = 14$ months

Staff cost required to develop the product = $91 \times \text{Rs. } 15,000 = \text{Rs. } 1,465,000$

Some insight into the basic COCOMO model can be obtained by plotting the estimated characteristics for different software sizes. Thus, the effort required to develop a product increases very rapidly with project size.

Intermediate COCOMO model

The basic COCOMO model assumes that effort and development time are functions of the product size alone. However, a host of other project parameters besides the product size affect the effort required to develop the product as well as the development time. Therefore, in order to obtain an accurate estimation of the effort and project duration, the effect of all relevant parameters must be taken into account. The intermediate COCOMO model recognizes this fact and refines the initial estimate obtained using the basic COCOMO expressions by using a set of 15 cost drivers (multipliers) based on various attributes of software development. For example, if modern programming practices are used, the initial estimates are scaled downward by multiplication with a cost driver having a value less than 1. If there are stringent reliability requirements on the software product, this initial estimate is scaled upward. Boehm requires the project manager to rate these 15 different parameters for a particular project on a scale of one to three. Then, depending on these ratings, he suggests appropriate cost driver values which should be multiplied with the initial estimate obtained using the basic COCOMO. In general, the cost drivers can be classified as being attributes of the following items:

Product: The characteristics of the product that are considered include the inherent complexity of the product, reliability requirements of the product, etc.

Computer: Characteristics of the computer that are considered include the execution speed required, storage space required etc.

Personnel: The attributes of development personnel that are considered include the experience level of personnel, programming capability, analysis capability, etc.

Development Environment: Development environment attributes capture the development facilities available to the developers. An important parameter that is considered is the sophistication of the automation (CASE) tools used for software development.

Complete COCOMO model

A major shortcoming of both the basic and intermediate COCOMO models is that they consider a software product as a single homogeneous entity. However, most large systems are made up of several smaller sub-systems. These sub-systems may have widely different characteristics. For example, some sub-systems may be considered as organic type, some semidetached, and some embedded. Not only that the inherent development complexity of the subsystems may be different, but also for some subsystems the reliability requirements may be high, for some the development team might have no previous experience of similar development, and so on. The complete COCOMO model considers these differences in characteristics of the subsystems and estimates the effort and development time as the sum of the estimates for the individual subsystems. The cost of each subsystem is estimated separately. This approach reduces the margin of error in the final estimate.

The following development project can be considered as an example application of the complete COCOMO model. A distributed Management Information System (MIS) product for an organization having offices at several places across the country can have the following sub-components:

- Database part
- Graphical User Interface (GUI) part
- Communication part

Of these, the communication part can be considered as embedded software. The database part could be semi-detached software, and the GUI part organic software. The costs for these three components can be estimated separately, and summed up to give the overall cost of the system.

Halstead's Software Science – An Analytical Technique

Halstead's software science is an analytical technique to measure size, development effort, and development cost of software products. Halstead used a few primitive program parameters to develop the expressions for over all program length, potential minimum value, actual volume, effort, and development time.

For a given program, let:

- η_1 be the number of unique operators used in the program,
- η_2 be the number of unique operands used in the program,
- N_1 be the total number of operators used in the program,
- N_2 be the total number of operands used in the program.

Operators and Operands for the ANSI C language

The following is a suggested list of operators for the ANSI C language:

- ([. , -> * + - ~ ! ++ -- * / % + - << >> < > <= >= != == & ^ | && || = *= /= %= += -= <=> >=& ^= |= : ? { ;
- CASE DEFAULT IF ELSE SWITCH WHILE DO FOR GOTO CONTINUE BREAK RETURN and a function name in a function call

Operands are those variables and constants which are being used with operators in expressions. Note that variable names appearing in declarations are not considered as operands.

Example : The function name in a function definition is not counted as an operator.

```
int func ( int a, int b )
{
    . . .
}
```

For the above example code, the operators are: {}, () We do not consider func, a, and b as operands, since these are part of the function definition.

Length and Vocabulary

The length of a program as defined by Halstead, quantifies total usage of all operators and operands in the program. Thus, length $N = N_1 + N_2$. Halstead's definition of the length of the program as the total number of operators and operands roughly agrees with the intuitive notation of the program length as the total number of tokens used in the program.

The program vocabulary is the number of unique operators and operands used in the program. Thus, *program vocabulary* $\eta = \eta_1 + \eta_2$.

Program Volume

The length of a program (i.e. the total number of operators and operands used in the code) depends on the choice of the operators and operands used. In other words, for the same programming problem, the length would depend on the programming style. This type of dependency would produce different measures of length for essentially the same problem when different programming languages are used. Thus, while expressing program size, the programming language used must be taken into consideration:

$$V = N \log_2 \eta$$

Here the program volume V is the minimum number of bits needed to encode the program. In fact, to represent η different identifiers uniquely, at least $\log_2 \eta$ bits (where η is the program vocabulary) will be needed. In this scheme, $N \log_2 \eta$ bits will be needed to store a program of length N . Therefore, the volume V represents the size of the program by approximately compensating for the effect of the programming language used.

Potential Minimum Volume

The potential minimum volume V^* is defined as the volume of most succinct program in which a problem can be coded. The minimum volume is obtained when the program can be expressed using a single source code instruction., say a function call like `foo() ;`. In other words, the volume is bound from below due to the fact that a program would have at least two operators and no less than the requisite number of operands.

Thus, if an algorithm operates on input and output data d_1, d_2, \dots, d_n , the most succinct program would be $f(d_1, d_2, \dots, d_n)$; for which $\eta_1 = 2, \eta_2 = n$. Therefore, $V^* = (2 + \eta_2)\log_2(2 + \eta_2)$.

The program level L is given by $L = V^*/V$. The concept of program level L is introduced in an attempt to measure the level of abstraction provided by the programming language. Using this definition, languages can be ranked into levels that also appear intuitively correct. The above result implies that the higher the level of a language, the less effort it takes to develop a program using that language. This result agrees with the intuitive notion that it takes more effort to develop a program in assembly language than to develop a program in a high-level language to solve a problem.

Effort and Time

The effort required to develop a program can be obtained by dividing the program volume with the level of the programming language used to develop the code. Thus, effort $E = V/L$, where E is the number of mental discriminations required to implement the program and also the effort required to read and understand the program. Thus, the programming effort $E = V^2/V^*$ (since $L = V^*/V$) varies as the square of the volume. Experience shows that E is well correlated to the effort needed for maintenance of an existing program.

The programmer's time $T = E/S$, where S the speed of mental discriminations. The value of S has been empirically developed from psychological reasoning, and its recommended value for programming applications is 18.

Length Estimation

Even though the length of a program can be found by calculating the total number of operators and operands in a program, Halstead suggests a way to determine the length of a program using the number of unique operators and operands used in the program. Using this method, the program parameters such as length, volume, cost, effort, etc. can be determined even before the start of any programming activity. His method is summarized below.

Halstead assumed that it is quite unlikely that a program has several identical parts – in formal language terminology identical substrings – of length greater than η (η being the program vocabulary). In fact, once a piece of code occurs identically at several places, it is made into a procedure or a function. Thus, it can be assumed that any program of length N consists of N/η unique strings of length η . Now, it is standard combinatorial result that for any given alphabet of size K , there are exactly K^r different strings of length r .

$$N/\eta \leq \eta^\eta \quad \text{Or, } N \leq \eta^{\eta+1}$$

Since operators and operands usually alternate in a program, the upper bound can be further refined into $N \leq \eta \eta_1^{\eta_1} \eta_2^{\eta_2}$. Also, N must include not only the ordered set of n elements, but it should also include all possible subsets of that ordered sets, i.e. the power set of N strings (This particular reasoning of Halstead is not very convincing!!!).

$$\text{Therefore, } 2^N = \eta \eta_1^{\eta_1} \eta_2^{\eta_2}$$

Or, taking logarithm on both sides,

$$N = \log_2 \eta + \log_2 (\eta_1^{\eta_1} \eta_2^{\eta_2})$$

So we get,

$$N = \log_2(\eta_1^{n_1} \eta_2^{n_2})$$

(approximately, by ignoring $\log_2 \eta$)

$$N = \log_2 \eta_1^{n_1} + \log_2 \eta_2^{n_2}$$

$$= n_1 \log_2 \eta_1 + n_2 \log_2 \eta_2$$

Experimental evidence gathered from the analysis of larger number of programs suggests that the computed and actual lengths match very closely. However, the results may be inaccurate when small programs when considered individually.

In conclusion, Halstead's theory tries to provide a formal definition and quantification of such qualitative attributes as program complexity, ease of understanding, and the level of abstraction based on some low-level parameters such as the number of operands, and operators appearing in the program. Halstead's software science provides gross estimation of properties of a large collection of software, but extends to individual cases rather inaccurately.

Example: Let us consider the following C program:

```
main( )
{
    int a, b, c, avg;
    scanf("%d %d %d", &a, &b, &c);
    avg = (a+b+c)/3;
    printf("avg = %d", avg);
}
```

The unique operators are: main(), {}, int, scanf, &, ",", ";", "=", "+", "/", printf

The unique operands are: a, b, c, &a, &b, &c, a+b+c, avg, 3, "%d %d %d", "avg = %d"

Therefore, $\eta_1 = 12$, $\eta_2 = 11$

Estimated Length = $(12 * \log 12 + 11 * \log 11)$

$$= (12 * 3.58 + 11 * 3.45)$$

$$= (43 + 38) = 81$$

Volume = Length * $\log(23)$

$$= 81 * 4.52$$

$$= 366$$

Staffing level estimation

Once the effort required to develop a software has been determined, it is necessary to determine the staffing requirement for the project. Putnam first studied the problem of what should be a proper staffing pattern for software projects. He extended the work of Norden who had earlier investigated the staffing pattern of research and development (R&D) type of projects. In order to appreciate the staffing pattern of software projects, Norden's and Putnam's results must be understood.

Norden's Work

Norden studied the staffing patterns of several R & D projects. He found that the staffing pattern can be approximated by the Rayleigh distribution curve. Norden represented the Rayleigh curve by the following equation:

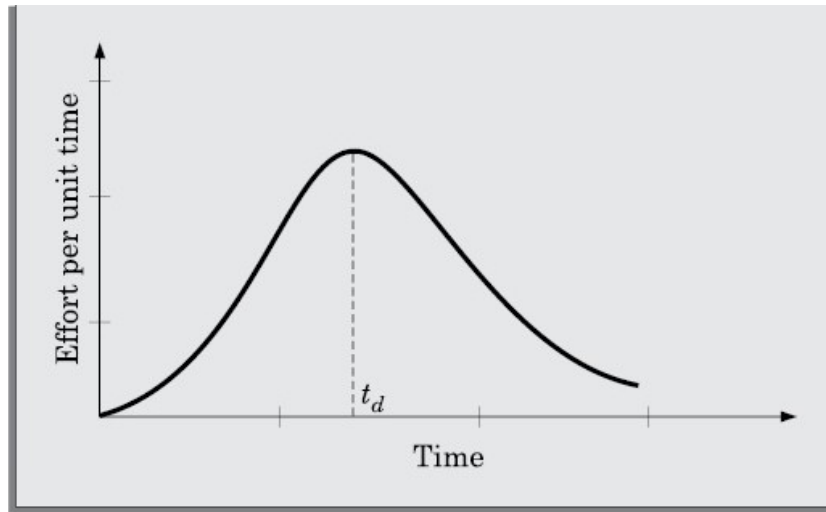


Figure: Rayleigh curve.

Norden represented the Rayleigh curve by the following equation:

$$E = \frac{K}{t_d^2} * t * e^{\frac{-t^2}{2t_d^2}}$$

where, E is the effort required at time t. E is an indication of the number of developers (or the staffing level) at any particular time during the duration of the project, K is the area under the curve, and t_d is the time at which the curve attains its maximum value. It must be remembered that the results of Norden are applicable to general R&D projects and were not meant to model the staffing pattern of software development projects.

Putnam's Work

Putnam studied the problem of staffing of software projects and found that the software development has characteristics very similar to other R & D projects studied by Norden and that the Rayleigh-Norden curve can be used to relate the number of delivered lines of code to the effort and the time required to develop the project. By analyzing a large number of army projects, Putnam derived the following expression:

$$L = C_k K^{1/3} t_d^{4/3}$$

The various terms of this expression are as follows:

- K is the total effort expended (in PM) in the product development and L is the product size in KLOC.
- t_d corresponds to the time of system and integration testing. Therefore, t_d can be approximately considered as the time required to develop the software.
- C_k is the state of technology constant and reflects constraints that impede the progress of the programmer. Typical values of $C_k = 2$ for poor development environment (no methodology, poor documentation, and review, etc.), $C_k = 8$ for good software development environment (software engineering principles are adhered to), $C_k = 11$ for an

excellent environment (in addition to following software engineering principles, automated tools and techniques are used). The exact value of C_k for a specific project can be computed from the historical data of the organization developing it.

Putnam suggested that optimal staff build-up on a project should follow the Rayleigh curve. Only a small number of engineers are needed at the beginning of a project to carry out planning and specification tasks. As the project progresses and more detailed work is required, the number of engineers reaches a peak. After implementation and unit testing, the number of project staff falls.

Effect of schedule change on cost

By analyzing a large number of army projects, Putnam derived the following expression:

$$K = \frac{L^3}{(C_k^3 t_d^4)}$$

or,

$$K = \frac{C}{t_d^4}$$

For the same product size, $C = \frac{L^3}{C_k^3}$ is a constant.

Or,

$$\frac{K_1}{K_2} = \frac{t_{d2}^4}{t_{d1}^4} \quad (3.3)$$

Where, K is the total effort expended (in PM) in the product development and L is the product size in KLOC, t_d corresponds to the time of system and integration testing and C_k is the state of technology constant and reflects constraints that impede the progress of the programmer.

Example : The nominal effort and duration of a project have been estimated to be 1000PM and 15 months. The project cost has been negotiated to be Rs. 200,000,000. The needs the product to be developed and delivered in 12 month time. What should be the new cost to be negotiated?

Answer: The project can be classified as a large project. Therefore, the new cost to be negotiated can be given by the Putnam's formula:

$$\text{new cost} = \text{Rs. } 200,000,000 \times \left(\frac{15}{12}\right)^4 = \text{Rs. } 488,281,250.$$

Project scheduling

Project-task scheduling is an important project planning activity. It involves deciding which tasks would be taken up when. In order to schedule the project activities, a software project manager needs to do the following:

- Identify all the tasks needed to complete the project.
- Break down large tasks into small activities.
- Determine the dependency among different activities.
- Establish the most likely estimates for the time durations necessary to complete the activities.
- Allocate resources to activities.
- Plan the starting and ending dates for various activities.

- Determine the critical path. A critical path is the chain of activities that determines the duration of the project.

The first step in scheduling a software project involves identifying all the tasks necessary to complete the project. A good knowledge of the intricacies of the project and the development process helps the managers to effectively identify the important tasks of the project. Next, the large tasks are broken down into a logical set of small activities which would be assigned to different engineers. The work breakdown structure formalism helps the manager to breakdown the tasks systematically.

After the project manager has broken down the tasks and created the work breakdown structure, he has to find the dependency among the activities. Dependency among the different activities determines the order in which the different activities would be carried out. If an activity A requires the results of another activity B, then activity A must be scheduled after activity B. In general, the task dependencies define a partial ordering among tasks, i.e. each tasks may precede a subset of other tasks, but some tasks might not have any precedence ordering defined between them (called concurrent task). The dependency among the activities are represented in the form of an activity network.

Once the activity network representation has been worked out, resources are allocated to each activity. Resource allocation is typically done using a Gantt chart. After resource allocation is done, a PERT chart representation is developed. The PERT chart representation is suitable for program monitoring and control. For task scheduling, the project manager needs to decompose the project tasks into a set of activities. The time frame when each activity is to be performed is to be determined. The end of each activity is called milestone. The project manager tracks the progress of a project by monitoring the timely completion of the milestones. If he observes that the milestones start getting delayed, then he has to carefully control the activities, so that the overall deadline can still be met.

Work breakdown structure

Work Breakdown Structure (WBS) is used to decompose a given task set recursively into small activities. WBS provides a notation for representing the major tasks need to be carried out in order to solve a problem. The root of the tree is labeled by the problem name. Each node of the tree is broken down into smaller activities that are made the children of the node. Each activity is recursively decomposed into smaller sub-activities until at the leaf level, the activities requires approximately two weeks to develop. It represents the WBS of an MIS (Management Information System) software.

While breaking down a task into smaller tasks, the manager has to make some hard decisions. If a task is broken down into large number of very small activities, these can be carried out independently. Thus, it becomes possible to develop the product faster (with the help of additional manpower). Therefore, to be able to complete a project in the least amount of time, the manager needs to break large tasks into smaller ones, expecting to find more parallelism. However, it is not useful to subdivide tasks into units which take less than a week or two to execute. Very fine subdivision means that a disproportionate amount of time must be spent on preparing and revising various charts.

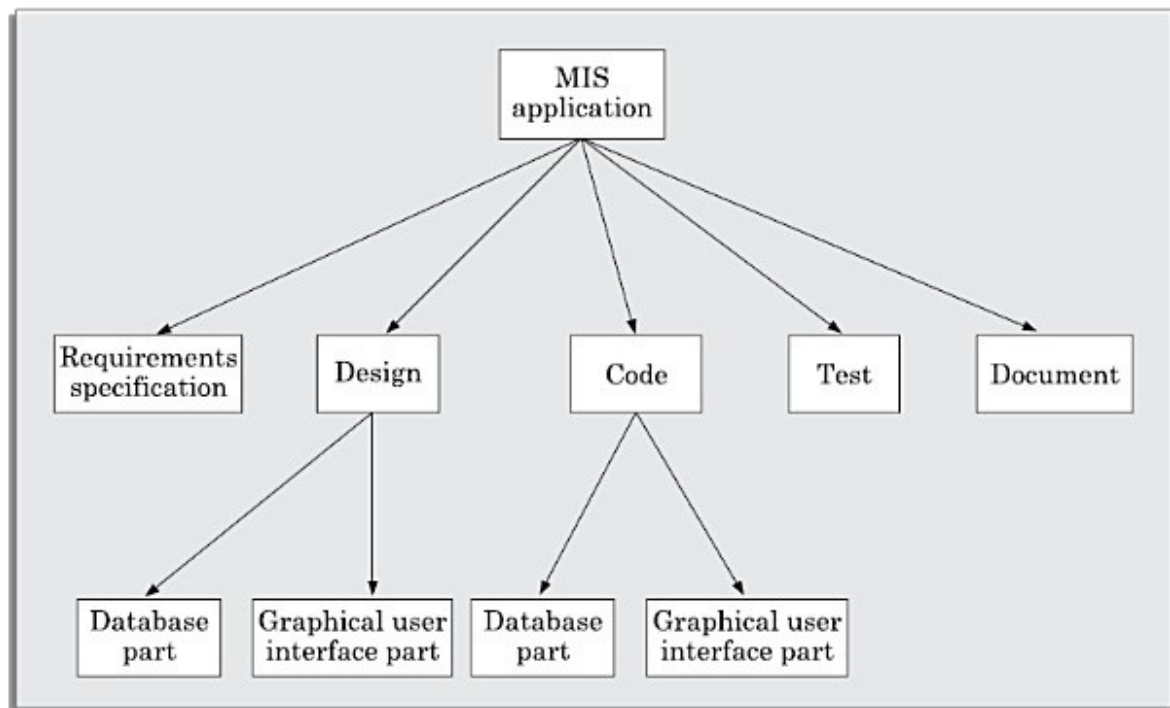


Figure: Work breakdown structure of an MIS problem.

How long to decompose?

The decomposition of the activities is carried out until any of the following is satisfied:

- A leaf-level subactivity (a task) requires approximately two weeks to develop.
- Hidden complexities are exposed, so that the job to be done is understood and can be assigned as a unit of work to one of the developers.
- Opportunities for reuse of existing software components is identified.

Activity Networks

An activity network shows the different activities making up a project, their estimated durations, and their interdependencies. Two equivalent representations for activity networks are possible and are in use:

Activity on Node (AoN): In this representation, each activity is represented by a rectangular (some use circular) node and the duration of the activity is shown alongside each task in the node. The inter-task dependencies are shown using directional edges.

Activity on Edge (AoE): In this representation tasks are associated with the edges. The edges are also annotated with the task duration. The nodes in the graph represent project milestones.

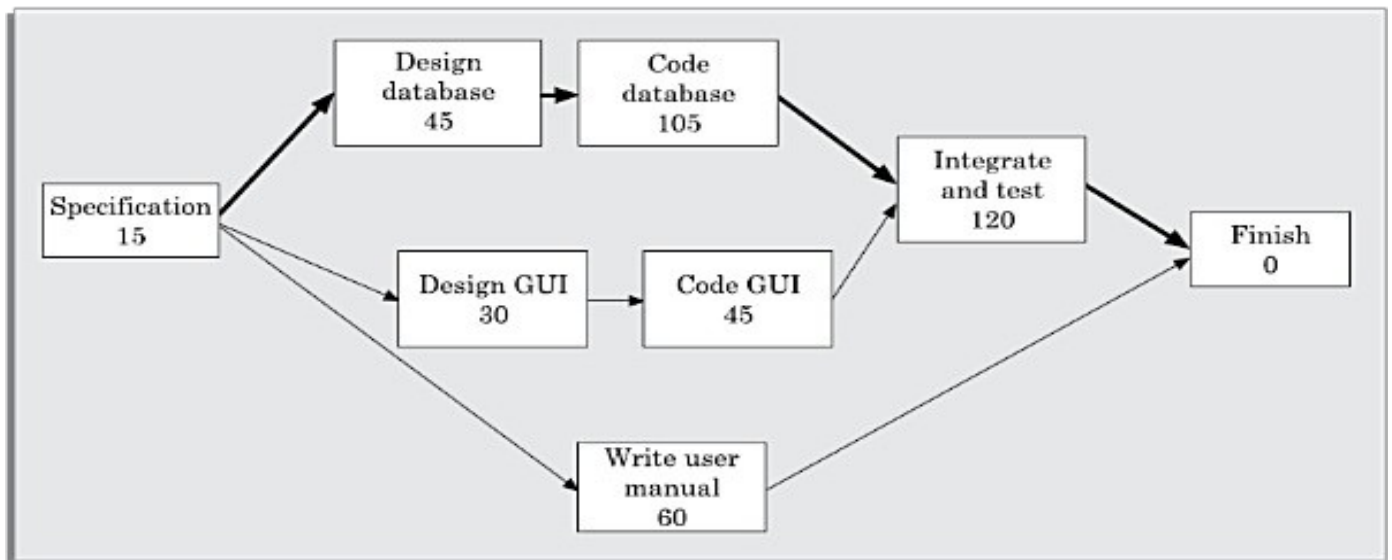


Figure: Activity network representation of the MIS problem.

Example: Determine the Activity network representation for the MIS development project of Example. Assume that the manager has determined the tasks to be represented from the work breakdown structure of Figure, and has determined the durations and dependencies for each task as shown in Table

Answer: The activity network representation has been shown in Figure

Table : Project Parameters Computed from Activity Network				
Task Number	Task	Duration	Dependent on	Tasks
T1	Specification	15	–	
T2	Design database	45	T 1	
T3	Design GUI	30	T 1	
T4	Code database	105	T 2	
T5	Code GUI part	45	T 3	
T6	Integrate and test	120	T 4 and T 5	
T7	Write user manual	60	T 1	

Critical Path Method (CPM)

CPM and PERT are operation research techniques that were developed in the late 1950s. Since then, they have remained extremely popular among project managers. Of late, these two techniques have got merged and many project management tools support them as CPM/PERT. However, we point out the fundamental differences between the two and discuss CPM in this subsection and PERT in the next subsection.

A path in the activity network graph is any set of consecutive nodes and edges in this graph from the starting node to the last node. A critical path consists of a set of dependent tasks that need to be performed in a sequence and which together take the longest time to complete.

CPM is an algorithmic approach to determine the critical paths and slack times for tasks not on the critical paths involves calculating the following quantities:

Minimum time (MT): It is the minimum time required to complete the project. It is computed by determining the maximum of all paths from start to finish.

Earliest start (ES): It is the time of a task is the maximum of all paths from the start to this task. The ES for a task is the ES of the previous task plus the duration of the preceding task.

Latest start time (LST): It is the difference between MT and the maximum of all paths from this task to the finish. The LST can be computed by subtracting the duration of the subsequent task from the LST of the subsequent task.

Earliest finish time (EF): The EF for a task is the sum of the earliest start time of the task and the duration of the task.

Latest finish (LF): LF indicates the latest time by which a task can finish without affecting the final completion time of the project. A task completing beyond its LF would cause project delay. LF of a task can be obtained by subtracting maximum of all paths from this task to finish from MT.

Slack time (ST): The slack time (or float time) is the total time that a task may be delayed before it will affect the end time of the project. The slack time indicates the "flexibility" in starting and completion of tasks. ST for a task is $LS - ES$ and can equivalently be written as $LF - EF$.

Example: Use the Activity network of Figure to determine the ES and EF for every task for the MIS problem of Example.

Answer: The activity network with computed ES and EF values has been shown in Figure

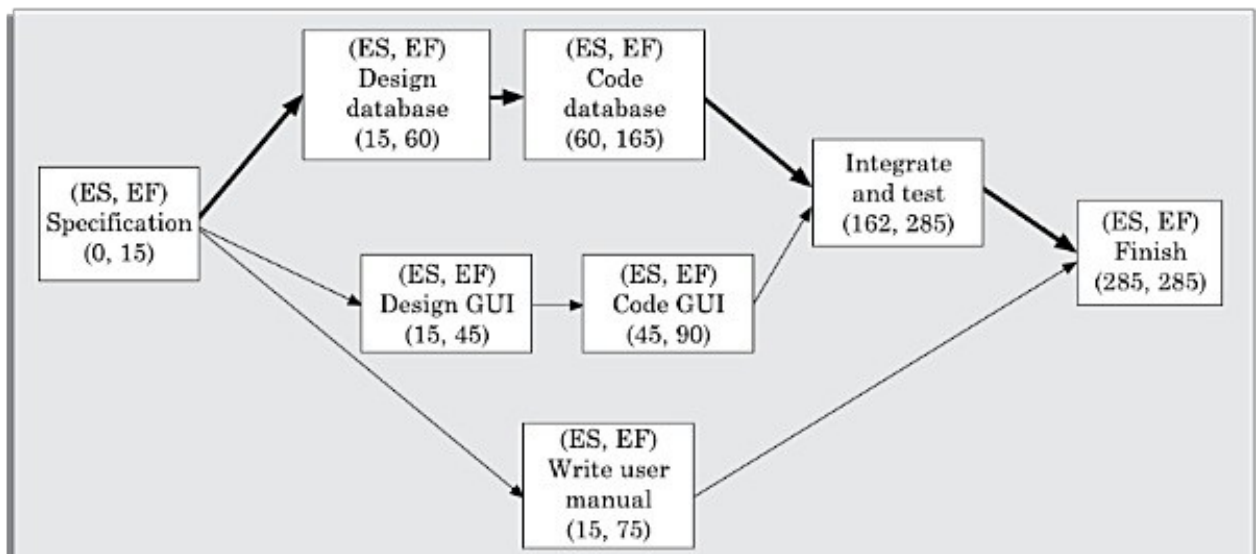


Figure 1 : AoN for MIS problem with (ES,EF).

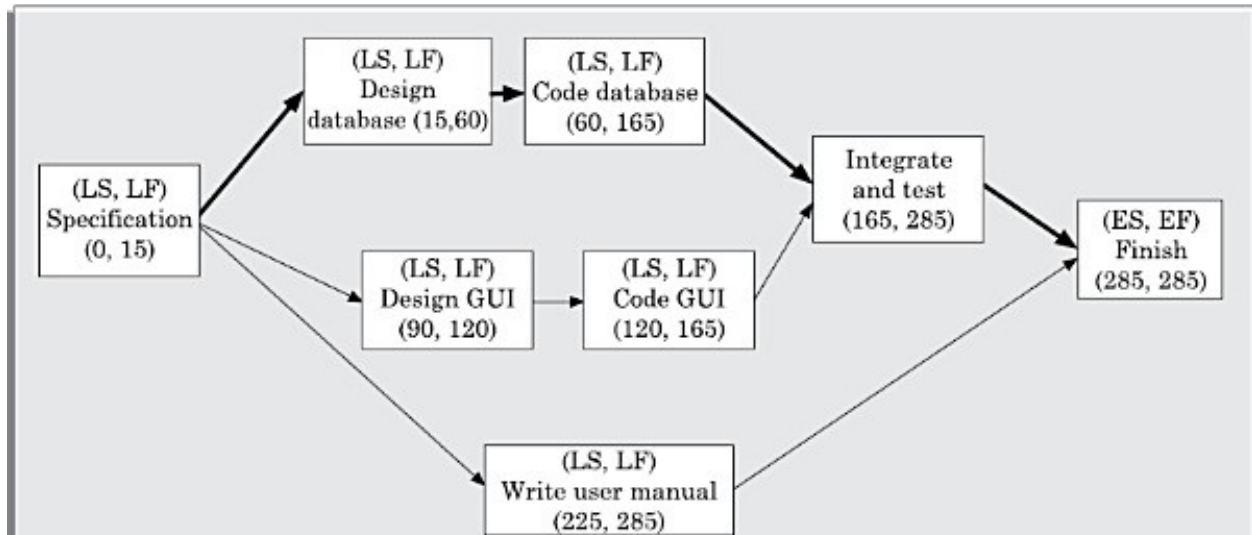


Figure 2: AoN of MIS problem with (LS,LF).

In Figure 1 and Figure 2, we show computation of (ES,EF) and (LS,LF) respectively. From this project parameters for different tasks for the MIS problem have been represented in Table 3.8.

Table : Project Parameters Computed From Activity Network

Task	ES	EF	LS	LF	ST
Specification	0	15	0	15	0
Design data base	15	60	15	60	0
Design GUI part	15	45	90	120	75
Code data base	60	165	60	165	0
Code GUI part	45	90	120	165	75
Integrate and test	165	285	165	285	0
Write user manual	15	75	225	285	210

The critical paths are all the paths whose duration equals MT. The critical path

PERT chart

PERT (Project Evaluation and Review Technique) charts consist of a network of boxes and arrows. The boxes represent activities and the arrows represent task dependencies. PERT chart represents the statistical variations in the project estimates assuming a normal distribution. Thus, in a PERT chart instead of making a single estimate for each task, pessimistic, likely, and optimistic estimates are made. The boxes of PERT charts are usually annotated with the pessimistic, likely, and optimistic estimates for every task. Since all possible completion times between the minimum and maximum duration for every task has to be considered, there are not one but many critical paths, depending on the permutations of the estimates for each task. This makes critical path analysis in PERT charts very complex. A critical path in a PERT chart is shown by using thicker arrows. The PERT chart representation of the MIS problem. PERT charts are a more sophisticated form of activity chart. In activity diagrams only the estimated task durations are represented. Since, the actual durations might vary from the estimated durations, the utility of the activity diagrams are limited.

Gantt chart representation of a project schedule is helpful in planning the utilization of resources, while PERT chart is useful for monitoring the timely progress of activities. Also, it is easier to identify parallel activities in a

project using a PERT chart. Project managers need to identify the parallel activities in a project for assignment to different engineers.

Each task is annotated with three estimates:

- **Optimistic (O):** The best possible case task completion time.
- **Most likely estimate (M):** Most likely task completion time.
- **Worst case (W):** The worst possible case task completion time.

The standard deviation for a task $ST = (P - O)/6$.

The mean estimated time is calculated as $ET = (O + 4M + W)/6$.

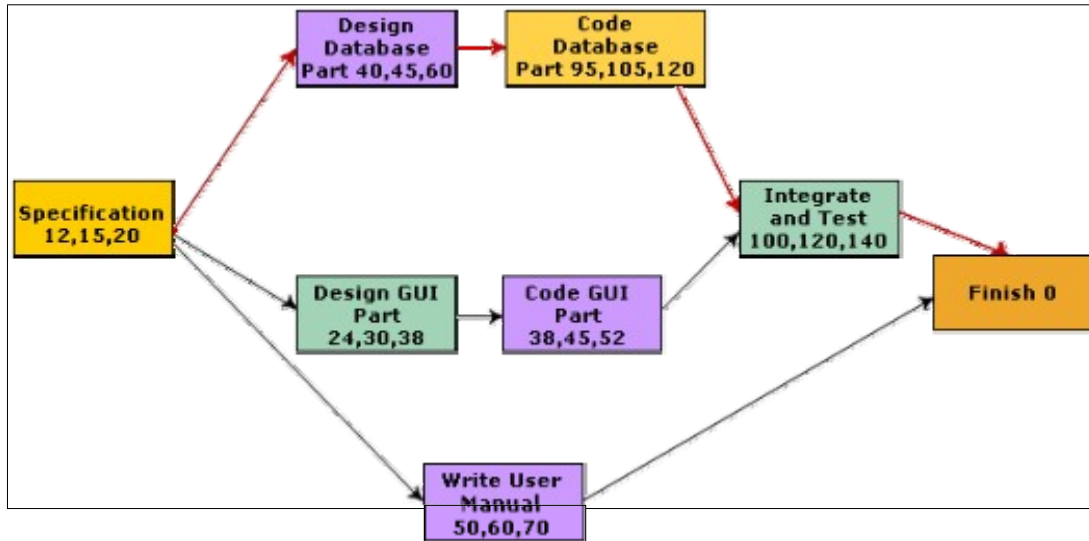


Fig: PERT chart representation of the MIS problem

Gantt Charts

Gantt chart has been named after its developer Henry Gantt. A Gantt chart is a form of bar chart. The vertical axis lists all the tasks to be performed. The bars are drawn along the y-axis, one for each task. Gantt charts used in software project management are actually an enhanced version of the standard Gantt charts. In the Gantt charts used for software project management, each bar consists of a unshaded part and a shaded part. The shaded part of the bar shows the length of time each task is estimated to take. The unshaded part shows the slack time or lax time. The lax time represents the leeway or flexibility available in meeting the latest time by which a task must be finished. Gantt charts are useful for resource planning (i.e. allocate resources to activities). The different types of resources that need to be allocated to activities include staff, hardware, and software.

Gantt chart representation of a project schedule is helpful in planning the utilisation of resources, while PERT chart is useful for monitoring the timely progress of activities. Also, it is easier to identify parallel activities in a project using a PERT chart. Project managers need to identify the parallel activities in a project for assignment to different developers.

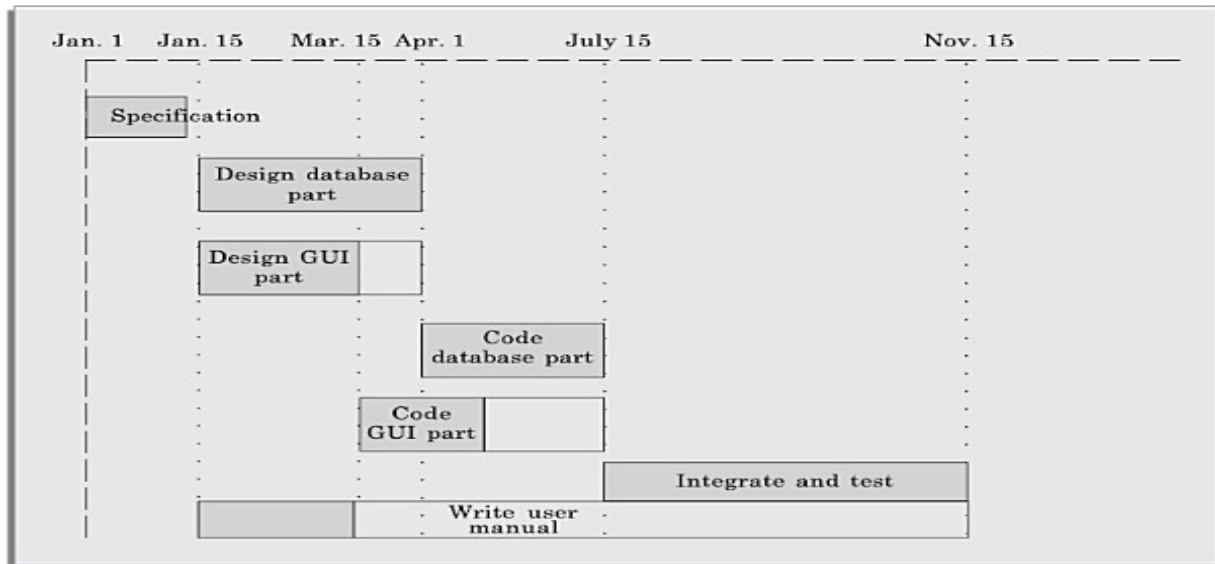


Figure 3.12: Gantt chart representation of the MIS problem.

A Gantt chart is a special type of bar chart where each bar represents an activity. The bars are drawn along a time line. The length of each bar is proportional to the duration of time planned for the corresponding activity.

ORGANISATION AND TEAM STRUCTURES

Usually every software development organisation handles several projects at any time. Software organisations assign different teams of developers to handle different software projects. With regard to staff organisation, there are two important issues—How is the organisation as a whole structured? And, how are the individual project teams structured? There are a few standard ways in which software organisations and teams can be structured.

Functional format

In the functional format, the development staff are divided based on the specific functional group to which they belong to. This format has schematically.

The different projects borrow developers from various functional groups for specific phases of the project and return them to the functional group upon the completion of the phase. As a result, different teams of programmers from different functional groups perform different phases of a project. For example, one team might do the requirements specification, another do the design, and so on. The partially completed product passes from one team to another as the product evolves. Therefore, the functional format requires considerable communication among the different teams and development of good quality documents because the work of one team must be clearly understood by the subsequent teams working on the project. The functional organisation therefore mandates good quality documentation to be produced after every activity.

Project format

In the project format, the development staff are divided based on the project for which they work. A set of developers is assigned to every project at the start of the project, and remain with the project till the completion of the project. Thus, the same team carries out all the life cycle activities. An advantage of the project format is that it provides job rotation. That is, every developer undertakes different life cycle activities in a project. However, it results in poor manpower utilisation, since the full project team is formed

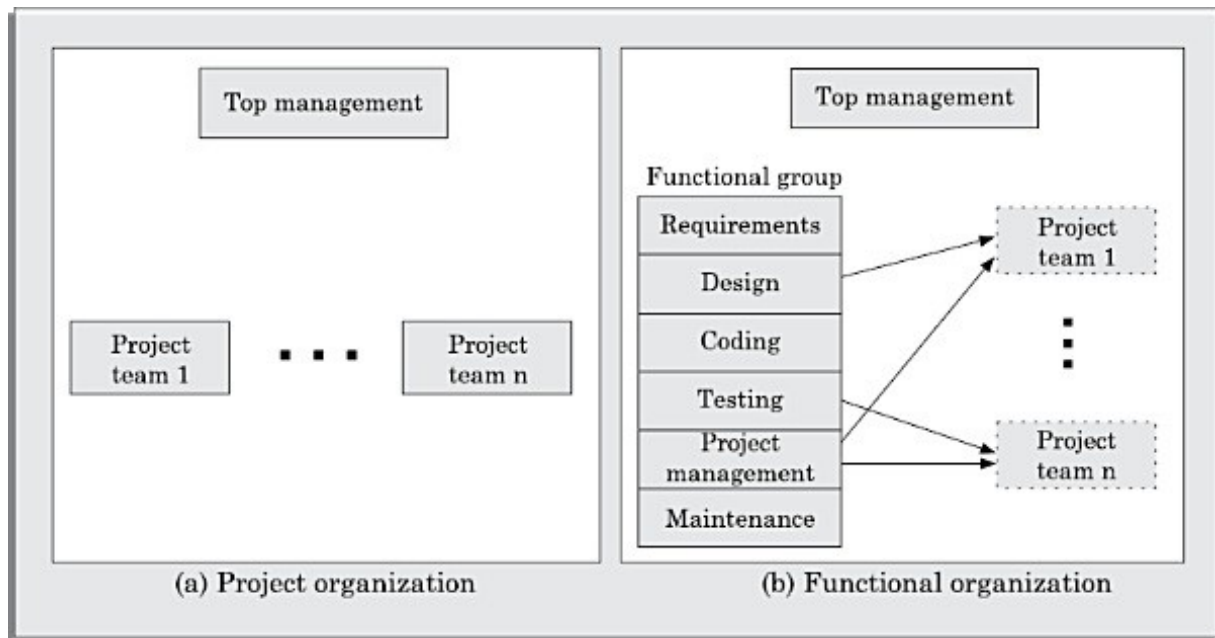


Figure: Schematic representation of the functional and project organisation.

Advantages of functional organization over project organization

Even though greater communication among the team members may appear as an avoidable overhead, the functional format has many advantages. The main advantages of a functional organization are:

- Ease of staffing
- Production of good quality documents
- Job specialization
- Efficient handling of the problems associated with manpower turnover.

The functional organization allows the engineers to become specialists in particular roles, e.g. requirements analysis, design, coding, testing, maintenance, etc. They perform these roles again and again for different projects and develop deep insights to their work. It also results in more attention being paid to proper documentation at the end of a phase because of the greater need for clear communication as between teams doing different phases. The functional organization also provides an efficient solution to the staffing problem. We have already seen that the staffing pattern should approximately follow the Rayleigh distribution for efficient utilization of the personnel by minimizing their wait times. The project staffing problem is eased significantly because personnel can be brought onto a project as needed, and returned to the functional group when they are no more needed. This possibly is the most important advantage of the functional organization. A project organization structure forces the manager to take in almost a constant number of engineers for the entire duration of his project. This results in engineers idling in the initial phase of the software development and are under tremendous pressure in the later phase of the development. A further advantage of the functional organization is that it is more effective in handling the problem of manpower turnover. This is because engineers can be brought in from the functional pool when needed. Also, this organization mandates production of good quality documents, so new engineers can quickly get used to the work already done.

Matrix format

A matrix organisation is intended to provide the advantages of both functional and project structures. In a matrix organisation, the pool of functional specialists are assigned to different projects as needed. Thus, the deployment of the different functional specialists in different projects can be represented

in a matrix observe that different members of a functional specialisation are assigned to different projects. Therefore in a matrix organisation, the project manager needs to share responsibilities for the project with a number of individual functional managers.

Functional group	Project			
	#1	#2	#3	
#1	2	0	3	Functional manager 1
#2	0	5	3	Functional manager 2
#3	0	4	2	Functional manager 3
#4	1	4	0	Functional manager 4
#5	0	4	6	Functional manager 5
	Project manager 1	Project manager 2	Project manager 3	

Figure: Matrix organisation.

Matrix organisations can be characterised as weak or strong, depending upon the relative authority of the functional managers and the project managers. In a strong functional matrix, the functional managers have authority to assign workers to projects and project managers have to accept the assigned personnel. In a weak matrix, the project manager controls the project budget, can reject workers from functional groups, or even decide to hire outside workers.

Two important problems that a matrix organisation often suffers from are:

- Conflict between functional manager and project managers over allocation of workers.
- Frequent shifting of workers in a firefighting mode as crises occur in different projects.

Team structures

Team structure addresses the issue of organization of the individual project teams. There are some possible ways in which the individual project teams can be organized. There are mainly three formal team structures: chief programmer, democratic, and the mixed team organizations although several other variations to these structures are possible. Problems of different complexities and sizes often require different team structures for chief solution.

Chief Programmer Team

In this team organization, a senior engineer provides the technical leadership and is designated as the chief programmer. The chief programmer partitions the task into small activities and assigns them to the team members. He also verifies and integrates the products developed by different team members. The structure of the chief programmer team. The chief programmer provides an authority, and this structure is arguably more efficient than the democratic team for well-understood problems. However, the chief programmer team leads to lower team morale, since team-members work under the constant supervision of the chief programmer. This also inhibits their original thinking. The chief programmer team is subject to single point failure since too much responsibility and authority is assigned to the chief programmer.

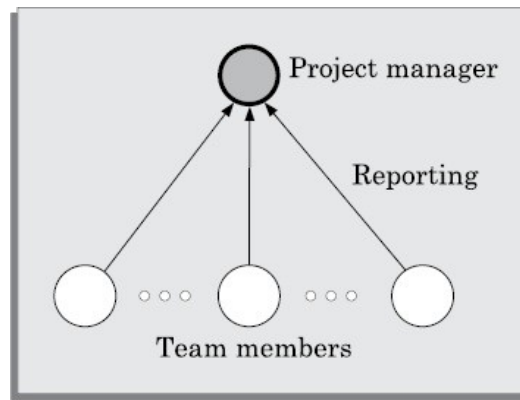


Figure: Chief programmer team structure.

The chief programmer team is probably the most efficient way of completing simple and small projects since the chief programmer can quickly work out a satisfactory design and ask the programmers to code different modules of his design solution.

Democratic team

The democratic team structure, as the name implies, does not enforce any formal team hierarchy. Typically, a manager provides the administrative leadership. At different times, different members of the group provide technical leadership.

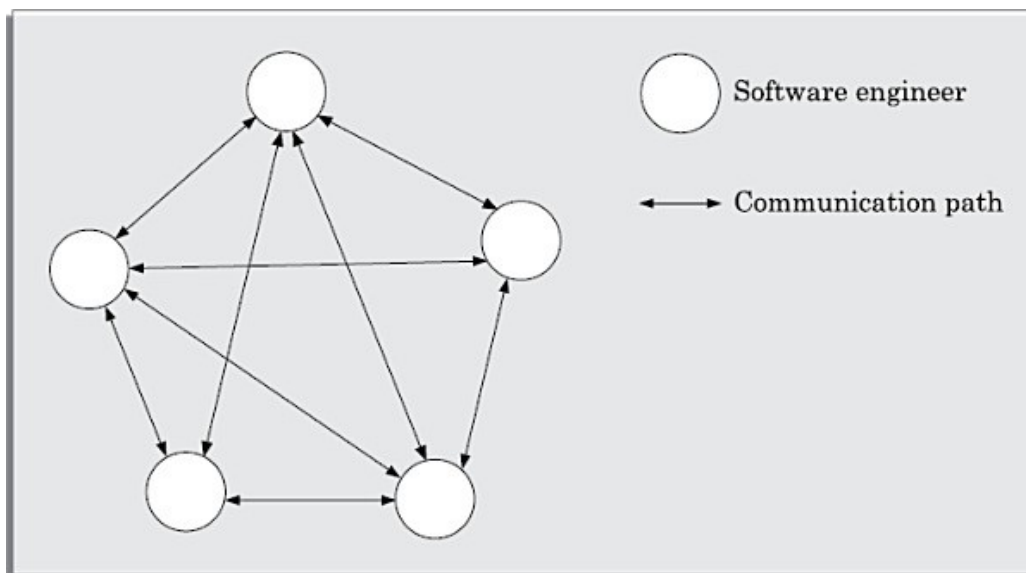


Figure: Democratic team structure.

In a democratic organisation, the team members have higher morale and job satisfaction. Consequently, it suffers from less manpower turnover. Though the democratic teams are less productive compared to the chief programmer team, the democratic team structure is appropriate for less understood problems, since a group of developers can invent better solutions than a single individual as in a chief programmer team. A democratic team structure is suitable for research-oriented projects requiring less than five or six developers. For large sized projects, a pure democratic organisation tends to become chaotic. The democratic team organisation encourages egoless programming as programmers can share and review each other's work. To appreciate the concept of egoless programming, we need to understand the concept of ego from a psychological perspective.

Mixed control team organisation

The mixed control team organisation, as the name implies, draws upon the ideas from both the democratic organisation and the chief-programmer organisation. The mixed control team organisation is shown pictorially. This team organisation incorporates both hierarchical reporting and democratic set up. the communication paths are shown as dashed lines and the reporting structure is shown using solid arrows. The mixed control team organisation is suitable for large team sizes. The democratic arrangement at the senior developers level is used to decompose the problem into small parts. Each democratic setup at the programmer level attempts solution to a single part. Thus, this team organisation is eminently suited to handle large and complex programs. This team structure is extremely popular and is being used in many software development companies.

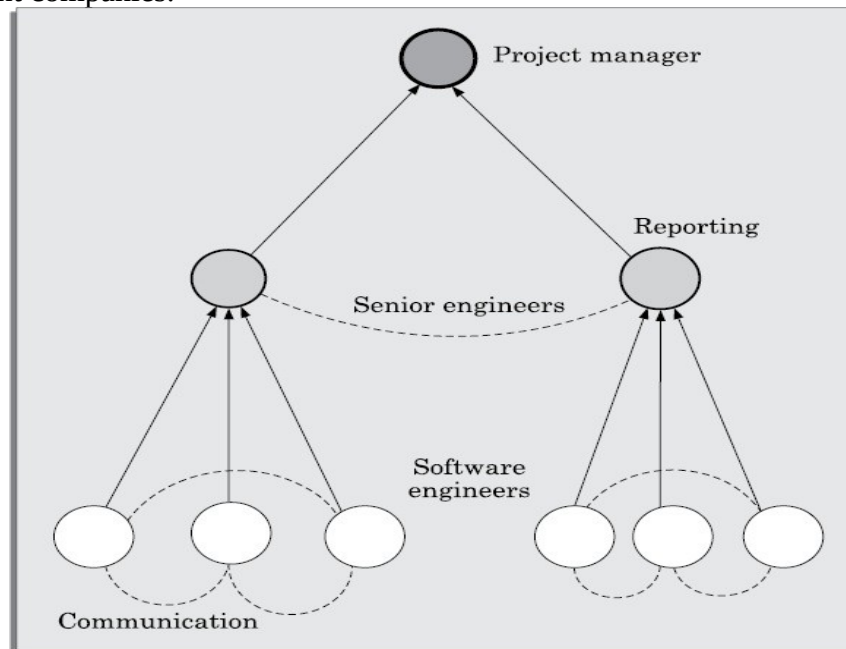


Figure: Mixed team structure.

Egoless programming technique

- Ordinarily, the human psychology makes an individual take pride in everything he creates using original thinking. Software development requires original thinking too, although of a different type. The human psychology makes one emotionally involved with his creation and hinders him from objective examination of his creations. Just like temperamental artists, programmers find it extremely difficult to locate bugs in their own programs or flaws in their own design. Therefore, the best way to find problems in a design or code is to have someone review it. Often, having to explain one's program to someone else gives a person enough objectivity to find out what might have gone wrong. This observation is the basic idea behind code walk throughs. An application of this, is to encourage democratic team to think that the design, code, and other deliverables to belong to the entire group. This is called egoless programming technique

STAFFING

Software project managers usually take the responsibility of choosing their team. Therefore, they need to identify good software developers for the success of the project. A common misconception held by managers as evidenced in their staffing, planning and scheduling practices, is the assumption that one software engineer is as productive as another. However, experiments have revealed that there exists a large variability of productivity between the worst and the best software developers in a scale of 1 to 30. In fact,

the worst developers may sometimes even reduce the overall productivity of the team, and thus in effect exhibit negative productivity. Therefore, choosing good software developers is crucial to the success of a project.

Who is a good software engineer?

In the past, several studies concerning the traits of a good software engineer have been carried out. All these studies roughly agree on the following attributes that good software developers should possess:

- Exposure to systematic techniques, i.e. familiarity with software engineering principles.
- Good technical knowledge of the project areas (Domain knowledge)
- Good programming abilities.
- Good communication skills. These skills comprise of oral, written, and interpersonal skills.
- High motivation.
- Sound knowledge of fundamentals of computer science
- Intelligence.
- Ability to work in a team.
- Discipline, etc.

Risk management

Every project is susceptible to a large number of risks. Without effective management of the risks, even the most meticulously planned project may go hay ware.

A risk is any anticipated unfavourable event or circumstance that can occur while a project is underway.

Risk Identification

A software project can be affected by a large variety of risks. In order to be able to systematically identify the important risks which might affect a software project, it is necessary to categorize risks into different classes. The project manager can then examine which risks from each class are relevant to the project. There are three main categories of risks which can affect a software project:

Project risks. Project risks concern varies forms of budgetary, schedule, personnel, resource, and customer-related problems. An important project risk is schedule slippage. Since, software is intangible, it is very difficult to monitor and control a software project. It is very difficult to control something which cannot be seen. For any manufacturing project, such as manufacturing of cars, the project manager can see the product taking shape. He can for instance, see that the engine is fitted, after that the doors are fitted, the car is getting painted, etc. Thus he can easily assess the progress of the work and control it. The invisibility of the product being developed is an important reason why many software projects suffer from the risk of schedule slippage.

Technical risks. Technical risks concern potential design, implementation, interfacing, testing, and maintenance problems. Technical risks also include ambiguous specification, incomplete specification, changing specification, technical uncertainty, and technical obsolescence. Most technical risks occur due to the development team's insufficient knowledge about the project.

Business risks: This type of risks includes the risk of building an excellent product that no one wants, losing budgetary commitments, etc.

Classification of risks in a project

Example : The project manager can identify several risks in this project. Let us classify them appropriately.

- What if the project cost escalates and overshoots what was estimated?:

Project risk.

- What if the mobile phones that are developed become too bulky in size to conveniently carry?:

Business risk.

- What if it is later found out that the level of radiation coming from the phones is harmful to human being?: **Business risk.**

- What if call hand-off between satellites becomes too difficult to implement?: **Technical risk.**

Risk assessment

The objective of risk assessment is to rank the risks in terms of their damage causing potential. For risk assessment, first each risk should be rated in two ways:

- The likelihood of a risk coming true (denoted as r).
- The consequence of the problems associated with that risk (denoted as s).

Based on these two factors, the priority of each risk can be computed:

$$p = r * s$$

Where, p is the priority with which the risk must be handled, r is the probability of the risk becoming true, and s is the severity of damage caused due to the risk becoming true. If all identified risks are prioritized, then the most likely and damaging risks can be handled first and more comprehensive risk abatement procedures can be designed for these risks.

Risk Mitigation

After all the identified risks of a project are assessed, plans must be made to contain the most damaging and the most likely risks. Different risks require different containment procedures. In fact, most risks require ingenuity on the part of the project manager in tackling the risk.

There are three main strategies to plan for risk containment:

Avoid the risk: This may take several forms such as discussing with the customer to change the requirements to reduce the scope of the work, giving incentives to the engineers to avoid the risk of manpower turnover, etc. The different categories of constraints that usually give rise to risks are:

Process-related risk: These risks arise due to aggressive work schedule, budget, and resource utilisation.

Product-related risks: These risks arise due to commitment to challenging product features (e.g. response time of one second, etc.), quality, reliability etc.

Technology-related risks: These risks arise due to commitment to use certain technology (e.g., satellite communication).

A few examples of risk avoidance can be the following: Discussing with the customer to change the requirements to reduce the scope of the work, giving incentives to the developers to avoid the risk of manpower turnover, etc.

Transfer the risk: This strategy involves getting the risky component developed by a third party, buying insurance cover, etc.

Risk reduction: This involves planning ways to contain the damage due to a risk. For example, if there is risk that some key personnel might leave, new recruitment may be planned.

$$\text{risk leverage} = \frac{\text{risk exposure before reduction} - \text{risk exposure after reduction}}{\text{cost of reduction}}$$

Risk related to schedule slippage

Even though there are three broad ways to handle any risk, but still risk handling requires a lot of ingenuity on the part of a project manager. As an example, it can be considered the options available to contain an important type of risk that occurs in many software projects – that of schedule slippage. Risks relating to schedule slippage arise primarily due to the intangible nature of software. Therefore, these can be dealt with by increasing the visibility of the software product. Visibility of a software product can be increased by producing relevant documents during the development process wherever meaningful and getting these documents reviewed by an appropriate team. Milestones should be placed at regular intervals through a software engineering process to provide a manager with regular indication of progress. Completion of a phase of the development process before followed need not be the only milestones. Every phase can be broken down to reasonable-sized tasks and milestones can be scheduled for these tasks too. A milestone is reached, once documentation produced as part of a software engineering task is produced and gets successfully reviewed. Milestones need not be placed for every activity. An approximate rule of thumb is to set a milestone every 10 to 15 days.

Software configuration management

The results (also called as the deliverables) of a large software development effort typically consist of a large number of objects, e.g. source code, design document, SRS document, test document, user's manual, etc. These objects are usually referred to and modified by a number of software engineers through out the life cycle of the software. The state of all these objects at any point of time is called the configuration of the software product. The state of each deliverable object changes as development progresses and also as bugs are detected and fixed.

Software revision versus version

A new version of a software is created when there is a significant change in functionality, technology, or the hardware it runs on, etc. On the other hand a new revision of a software refers to minor bug fix in that software. A new release is created if there is only a bug fix, minor enhancements to the functionality, usability, etc.

For example, one version of a mathematical computation package might run on Unix-based machines, another on Microsoft Windows and so on. As a software is released and used by the customer, errors are discovered that need correction. Enhancements to the functionalities of the software may also be needed. A new release of software is an improved system intended to replace an old one. Often systems are described as

version m, release n; or simple m.n. Formally, a history relation is version of can be defined between objects. This relation can be split into two sub relations *is revision of* and *is variant of*.

Necessity of software configuration management

There are several reasons for putting an object under configuration management. But, possibly the most important reason for configuration management is to control the access to the different deliverable objects. Unless strict discipline is enforced regarding updation and storage of different objects, several problems appear. The following are some of the important problems that appear if configuration management is not used.

Inconsistency problem when the objects are replicated. A scenario can be considered where every software engineer has a personal copy of an object (e.g. source code). As each engineer makes changes to his local copy, he is expected to intimate them to other engineers, so that the changes in interfaces are uniformly changed across all modules. However, many times an engineer makes changes to the interfaces in his local copies and forgets to intimate other teammates about the changes. This makes the different copies of the object inconsistent. Finally, when the product is integrated, it does not work. Therefore, when several team members work on developing an object, it is necessary for them to work on a single copy of the object, otherwise inconsistency may arise.

Problems associated with concurrent access. Suppose there is a single copy of a problem module, and several engineers are working on it. Two engineers may simultaneously carry out changes to different portions of the same module, and while saving overwrite each other. Though the problem associated with concurrent access to program code has been explained, similar problems occur for any other deliverable object.

Providing a stable development environment. When a project is underway, the team members need a stable environment to make progress. Suppose somebody is trying to integrate module A, with the modules B and C, he cannot make progress if developer of module C keeps changing C; this can be especially frustrating if a change to module C forces him to recompile A. When an effective configuration management is in place, the manager freezes the objects to form a base line. When anyone needs any of the objects under configuration control, he is provided with a copy of the base line item. The requester makes changes to his private copy. Only after the requester is through with all modifications to his private copy, the configuration is updated and a new base line gets formed instantly. This establishes a baseline for others to use and depend on. Also, configuration may be frozen periodically. Freezing a configuration may involve archiving everything needed to rebuild it. (Archiving means copying to a safe place such as a magnetic tape).

System accounting and maintaining status information. System accounting keeps track of who made a particular change and when the change was made.

Handling variants. Existence of variants of a software product causes some peculiar problems. Suppose somebody has several variants of the same module, and finds a bug in one of them. Then, it has to be fixed in all versions and revisions. To do it efficiently, he should not have to fix it in each and every version and revision of the software separately.

Software configuration management activities

Normally, a project manager performs the configuration management activity by using an automated configuration management tool. A configuration management tool provides automated support for overcoming all the problems mentioned above. In addition, a configuration management tool helps to keep track of various deliverable objects, so that the project manager can quickly and unambiguously determine the current state of the project. The configuration management tool enables the engineers to change the various components in a controlled manner.

Configuration management is carried out through two principal activities:

- Configuration identification involves deciding which parts of the system should be kept track of.
- Configuration control ensures that changes to a system happen smoothly

Configuration identification

The project manager normally classifies the objects associated with a software development effort into three main categories: controlled, precontrolled, and uncontrolled. Controlled objects are those which are already put under configuration control. One must follow some formal procedures to change them. Precontrolled objects are not yet under configuration control, but will eventually be under configuration control. Uncontrolled objects are not and will not be subjected to configuration control. Controllable objects include both controlled and precontrolled objects. Typical controllable objects include:

- Requirements specification document
- Design documents
- Tools used to build the system, such as compilers, linkers, lexical analyzers, parsers, etc.
- Source code for each module
- Test cases
- Problem reports

The configuration management plan is written during the project planning phase and it lists all controlled objects. The managers who develop the plan must strike a balance between controlling too much, and controlling too little. If too much is controlled, overheads due to configuration management increase to unreasonably high levels. On the other hand, controlling too little might lead to confusion when something changes.

Configuration control

Configuration control is the process of managing changes to controlled objects. Configuration control is the part of a configuration management system that most directly affects the day-to-day operations of developers. The configuration control system prevents unauthorized changes to any controlled objects. In order to change a controlled object such as a module, a developer can get a private copy of the module by a reserve operation as shown in fig. 12.5. Configuration management tools allow only one person to reserve a module at a time. Once an object is reserved, it does not allow any one else to reserve this module until the reserved module is restored as shown in fig. 12.5. Thus, by preventing more than one engineer to simultaneously reserve a module, the problems associated with concurrent access are solved.

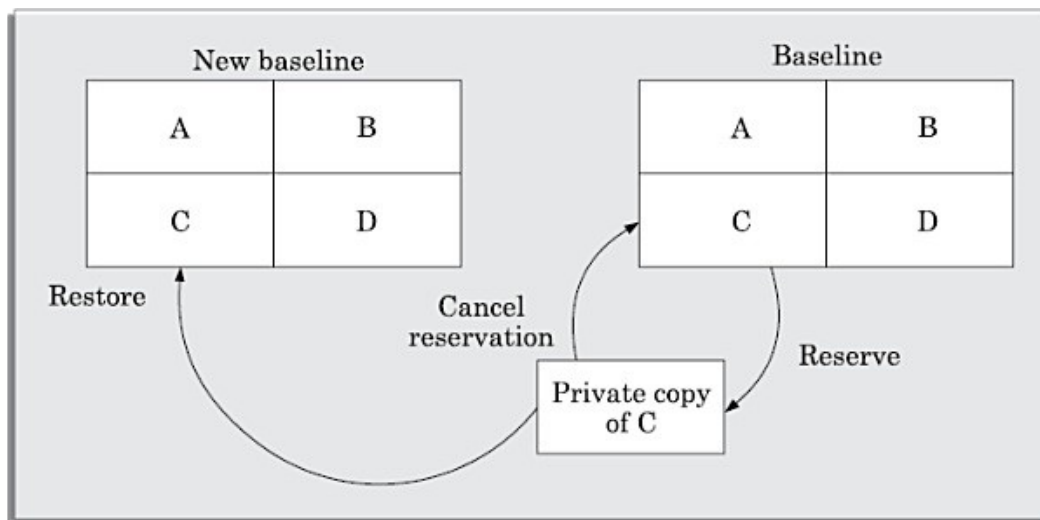


Fig: Reserve and restore operation in configuration control

It can be shown how the changes to any object that is under configuration control can be achieved. The engineer needing to change a module first obtains a private copy of the module through a reserve operation. Then, he carries out all necessary changes on this private copy. However, restoring the changed module to the system configuration requires the permission of a change control board (CCB). The CCB is usually constituted from among the development team members. For every change that needs to be carried out, the CCB reviews the changes made to the controlled object and certifies several things about the change:

- 1 Change is well-motivated.
- 2 Developer has considered and documented the effects of the change.
- 3 Changes interact well with the changes made by other developers.
- 4 Appropriate people (CCB) have validated the change, e.g. someone has tested the changed code, and has verified that the change is consistent with the requirement.

The change control board (CCB) sounds like a group of people. However, except for very large projects, the functions of the change control board are normally discharged by the project manager himself or some senior member of the development team. Once the CCB reviews the changes to the module, the project manager updates the old base line through a restore operation (as shown in fig. 12.5). A configuration control tool does not allow a developer to replace an object he has reserved with his local copy unless he gets an authorization from the CCB. By constraining the developers' ability to replace reserved objects, a stable environment is achieved. Since a configuration management tool allows only one engineer to work on one module at any one time, problem of accidental overwriting is eliminated. Also, since only the manager can update the baseline after the CCB approval, unintentional changes are eliminated.

Configuration management tools

SCCS and RCS are two popular configuration management tools available on most UNIX systems. SCCS or RCS can be used for controlling and managing different versions of text files. SCCS and RCS do not handle binary files (i.e. executable files, documents, files containing diagrams, etc.) SCCS and RCS provide an efficient way of storing versions that minimizes the amount of occupied disk space. Suppose, a module MOD is present in three versions MOD1.1, MOD1.2, and MOD1.3. Then, SCCS and RCS stores the original module MOD1.1 together with changes needed to transform MOD1.1 into MOD1.2 and MOD1.2 to MOD1.3. The changes needed to transform each base lined file to the next version are stored and are called

deltas. The main reason behind storing the deltas rather than storing the full version files is to save disk space.

The change control facilities provided by SCCS and RCS include the ability to incorporate restrictions on the set of individuals who can create new versions, and facilities for checking components in and out (i.e. reserve and restore operations). Individual developers check out components and modify them. After they have made all necessary changes to a module and after the changes have been reviewed, they check in the changed module into SCCS or RCS. Revisions are denoted by numbers in ascending order, e.g., 1.1, 1.2, 1.3 etc. It is also possible to create variants or revisions of a component by creating a fork in the development history.

SOFTWARE AND SOFTWARE ENGINEERING:

Software engineering stands for the term is made of two words, Software and Engineering.

Software is more than just a program code. A program is an executable code, which serves some computational purpose. Software is considered to be collection of executable programming code, associated libraries and documentations. Software, when made for a specific requirement is called software product.

Engineering on the other hand, is all about developing products, using well-defined, scientific principles and methods.

Software engineering is an engineering branch associated with development of software product using well-defined scientific principles, methods and procedures. The outcome of software engineering is an efficient and reliable software product.

1.THE NATURE OF SOFTWARE

Software takes Dual role of Software. It is a **Product** and at the same time a **Vehicle for delivering a product**.

Software delivers the most important product of our time is called **information**

Software is defined as

- 1. Instructions:** Programs that when executed provide desired function, features, and performance
- 2. Data structures:** Enable the programs to adequately manipulate information
- 3.Documents:** Descriptive information in both hard copy and virtual forms that describes the operation and use of the programs.

1.1.Characteristics of software:

Software has characteristics that are considerably different than those of hardware:

1.Software is developed or engineered, it is not manufactured in the Classical Sense.

Although some similarities exist between software development and hardware manufacture, the two activities are fundamentally different. In both the activities, high quality is achieved through good design, but the manufacturing phase for hardware can introduce quality problems that are nonexistent or easily corrected for software. Both the activities are dependent on people, but the relationship between people is totally varying. These two activities require the construction of a "**product**" but the approaches are different. Software costs are concentrated in engineering which means that software projects cannot be managed as if they were manufacturing.

2. Software doesn't "Wear Out"

The following figure shows the relationship between failure rate and time. Consider the failure rate as a function of time for hardware. The relationship is called **the bathtub curve**, indicates that hardware exhibits relatively high failure rates early in its life, defects are corrected and the failure rate drops to a steady-state level for some period of time. As time passes, however, the failure rate rises again as hardware components suffer from the cumulative effects of dust, vibration, abuse, temperature extremes, and many other environmental maladies. So, stated simply, the hardware begins to wear out. Software is not susceptible to the environmental maladies that cause **hardware to wear out**

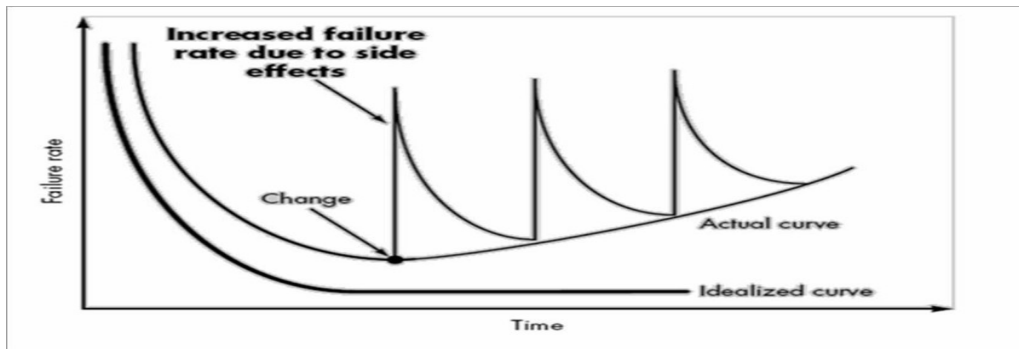


Fig: FAILURE CURVE FOR SOFTWARE

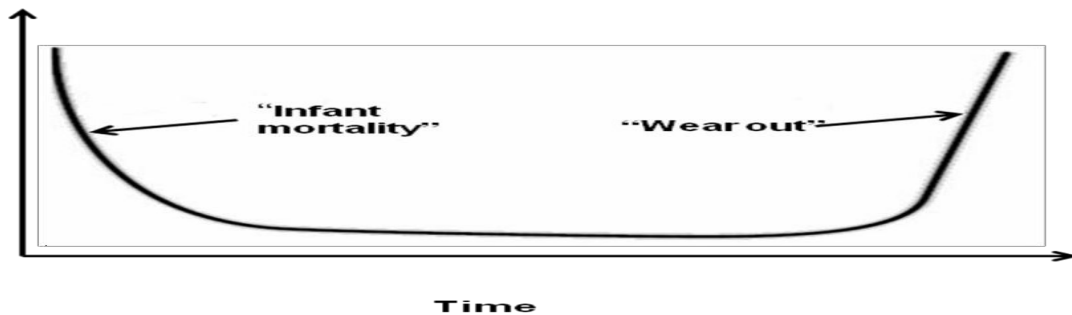


Fig: FAILURE CURVE FOR HARDWARE

3. Although the industry is moving toward component-based construction, most software continues to be custom built

A software component should be designed and implemented so that it can be reused in many Different programs . Modern reusable components encapsulate both data and the processing that is applied to the data, enabling the software engineer to create new applications from reusable parts

1.2.Types of Software:

1. System Software: It is a collection of programs written to service other programs. Some system software (example- compilers, editors and file management utilities). Some other system applications (operating system components, drivers, networking software, telecommunications processors). In either case, the system software area is characterized by heavy interaction with computer hardware, heavy usage by multiple users, scheduling, resource sharing, process management, complex data structure.

2. Application Software: It consists of standalone programs that solve a specific business need. Applications in this area process business or technical data in a way that facilitates business operations or management/technical decision-making. Application software is mainly used to control business functions in real-time.

3. Engineering/Scientific Software: Formerly characterized by “number crunching” algorithms, engineering and scientific software applications range from astronomy to volcanology. However, modern applications within the engineering/scientific area are moving away from conventional numerical algorithm Computer-Aided Design (CAD), System simulation has begun to take on real-time.

4. Embedded Software: Embedded Software resides within a product or system and it is used to implement and control features and functions for the end-user and for the system itself. Embedded Software can perform limited and esoteric functions. It also provides significant function and control capability (Example- Digital Function, Dashboard displays, cracking system, etc).

5. Product-line Software: Designed to provide a specific Capability for use by many different customers, product-line. The software can focus on a limited and esoteric marketplace (Example-inventory control products) or address mass consumer markets (Example – Word Processing, Spreadsheet, Computer Graphics, Multimedia, entertainment, database management, personal and business applications).

6. Web-Applications: “WebApps“, span a wide array of applications. In their simplest form, WebApps can be little more than a set of linked hypertext files that present information using text and limited graphics, such as- E-commerce and B2B Applications. WebApps are evolving into sophisticated computing functions and also are integrated with Corporate databases and business applications.

7. Artificial Intelligence Software: AI software makes use of non-numerical algorithms to solve complex problems that are not amenable to computations or straight-forward analysis. These applications within this area include robotics, expert systems, pattern recognition (image and voice), artificial neural networks and game playing.

New Software Challenges

- **Open-world computing** : Creating software to allow machines of all sizes to communicate with each other across vast networks (Distributed computing—wireless networks)
- **Netsourcing** : Architecting simple and sophisticated applications that benefit targeted end-user markets worldwide (the Web as a computing engine)
- **Open Source** : Distributing source code for computing applications so customers can make local modifications easily and reliably (“free” source code open to the computing community)

1.3.LEGACY SOFTWARE:

- Legacy software is older programs that are developed decades ago.
- The quality of legacy software is poor because it has inextensible design, convoluted code, poor and nonexistent documentation, test cases and results that are not achieved.As time passes legacy systems evolve due to following reasons:
- The software must be adapted to meet the needs of new computing environment or technology.
- The software must be enhanced to implement new business requirements.
- The software must be extended to make it interoperable with more modern systems or database
- The software must be re-architected to make it viable within a network environment.

2. UNIQUE NATURE OF WEB APPS

In the early days of the World Wide Web, websites consisted of little more than a set of linked hypertext files that presented information using text and limited graphics. As time passed, the augmentation of HTML by development tools (e.g., XML, Java) enabled Web engineers to provide computing capability along with informational content. *Web-based systems and applications (WebApps)* were born. Today, WebApps have evolved into sophisticated computing tools that not only provide stand-alone function to the end user, but also have been integrated with corporate databases and business applications.WebApps are one of a number of distinct software categories. Web-based systems and applications “involve a mixture between print publishing and software development, between marketing and computing, between internal communications and external relations, and between art and technology.”

The following attributes are encountered in the vast majority of WebApps.

- **Network intensiveness.** A WebApp resides on a network and must serve the needs of a diverse community of clients. The network may enable worldwide access and communication (i.e., the Internet) or more limited access and communication (e.g., a corporate Intranet).
- **Concurrency.** A large number of users may access the WebApp at one time. In many cases, the patterns of usage among end users will vary greatly.

- **Unpredictable load.** The number of users of the WebApp may vary by orders of magnitude from day to day. One hundred users may show up on Monday; 10,000 may use the system on Thursday.
- **Performance.** If a WebApp user must wait too long, he or she may decide to go elsewhere.
- **Availability.** Although expectation of 100 percent availability is unreasonable, users of popular WebApps often demand access on a 24/7/365 basis
- **Data driven.** The primary function of many WebApps is to use hypermedia to present text, graphics, audio, and video content to the end user. In addition, WebApps are commonly used to access information that exists on databases that are not an integral part of the Web-based environment (e.g., e-commerce or financial applications).
- **Content sensitive.** The quality and aesthetic nature of content remains an important determinant of the quality of a WebApp.
- **Continuous evolution.** Unlike conventional application software that evolves over a series of planned, chronologically spaced releases, Web applications evolve continuously.
- **Immediacy.** Although *immediacy*—the compelling need to get software to market quickly—is a characteristic of many application domains, WebApps often exhibit a time- to-market that can be a matter of a few days or weeks.
- **Security.** Because WebApps are available via network access, it is difficult, if not impossible, to limit the population of end users who may access the application. In order to protect sensitive content and provide secure modes
- **Aesthetics.** An undeniable part of the appeal of a WebApp is its look and feel. When an application has been designed to market or sell products or ideas, aesthetics may have as much to do with success as technical design.

3.SOFTWARE MYTHS:

Software Myths- beliefs about software and the process used to build it - can be traced to the earliest days of computing. Myths have a number of attributes that have made them insidious. For instance, myths appear to be reasonable statements of fact, they have an intuitive feel, and they are often promulgated by experienced practitioners who “know the score”.

We are having 3 types of myths

- **Management Myths**
- **Customer Myths**
- **Practitioner's Myths**

1.Management Myths :

Managers with software responsibility, like managers in most disciplines, are often under pressure to maintain budgets, keep schedules from slipping, and improve quality. Like a drowning person who grasps at a straw, a software manager often grasps at belief in a software myth.

Myth : We already have a book that's full of standards and procedures for building software. Won't That provide my people with everything they need to know?

Reality :

- The book of standards may very well exist, but is it used?
- Are software practitioners aware of its existence?

- Does it reflect modern software engineering practice?
- Is it complete?
- Is it adaptable?
- Is it streamlined to improve time to delivery while still maintaining a focus on Quality? In many cases, the answer to these entire question is NO.

Myth : *If we get behind schedule, we can add more programmers and catch up*

Reality : Software development is not a mechanistic process like manufacturing. “Adding people to a late software project makes it later.” At first, this statement may seem counterintuitive. However, as new people are added, people who were working must spend time educating the newcomers, thereby reducing the amount of time spent on productive development effort

Myth : If we decide to outsource the software project to a third party, I can just relax and let that firm build it.

Reality : If an organization does not understand how to manage and control software project internally, it will invariably struggle when it out sources software project.

2.Customer Myths

A customer who requests computer software may be a person at the next desk, technical group down the hall, the marketing /sales department, or an outside company that has requested software under contract. In many cases, the customer believes myths about software because software managers and practitioners do little to correct misinformation. Myths led to false expectations and ultimately, dissatisfaction with the developers.

Myth : A general statement of objectives is sufficient to begin writing programs - we can fill in details later.

Reality : Although a comprehensive and stable statement of requirements is not always possible, an ambiguous statement of objectives is a recipe for disaster. Unambiguous requirements are developed only through effective and continuous communication between customer and developer.

Myth : Project requirements continually change, but change can be easily accommodated because software is flexible.

Reality : It's true that software requirement change, but the impact of change varies with the time at which it is introduced. When requirement changes are requested early, cost impact is relatively small. However, as time passes, cost impact grows rapidly – resources have been committed, a design framework has been established, and change can cause upheaval that requires additional resources and major design modification.

3.Practitioner's myths.

Myths that are still believed by software practitioners have been fostered by 50 years of programming culture. During the early days of software, programming was viewed as an art form. Old ways and attitudes die hard.

Myth: Once we write the program and get it to work, our job is done.

Reality: Someone once said that "the sooner you begin 'writing code', the longer it'll take you to get done." Industry data indicate that between 60 and 80 percent of all effort expended on software will be expended after it is delivered to the customer for the first time.

Myth: Until I get the program "running" I have no way of assessing its quality.

Reality: One of the most effective software quality assurance mechanisms can be applied from the inception of a project—the formal technical review. Software reviews are a "quality filter" that have been found to be more effective than testing for finding certain classes of software defects.

Myth: The only deliverable work product for a successful project is the working program.

Reality: A working program is only one part of a software configuration that includes many elements. Documentation provides a foundation for successful engineering and, more important, guidance for software support.

Myth: Software engineering will make us create voluminous and unnecessary documentation and will invariably slow us down.

Reality: Software engineering is not about creating documents. It is about creating quality. Better quality leads to reduced rework. And reduced rework results in faster delivery times. Many software professionals recognize the fallacy of the myths just described. Regrettably, habitual attitudes and methods foster poor management and technical practices, even when reality dictates a better approach. Recognition of software realities is the first step toward formulation of practical solutions for software engineering.

REQUIREMENTS GATHERING AND ANALYSIS:

The complete set of requirements are almost never available in the form of a single document from the customer. In fact, it would be unrealistic to expect the customers to produce a comprehensive document containing a precise description of what he wants. Further, the complete requirements are rarely obtainable from any single customer representative. Therefore, the requirements have to be gathered by the analyst from several sources in bits and pieces. These gathered requirements need to be analysed to remove several types of problems that frequently occur in the requirements that have been gathered piecemeal from different sources.

For any type of software development project, availability of a good quality requirements document has been acknowledged to be a key factor in the successful completion of the project. A good requirements document not only helps to form a clear understanding of various features required from the software, but also serves as the basis for various activities carried out during later life cycle phases. After understanding the precise user requirements, the analysts analyse the requirements to weed out inconsistencies, anomalies and incompleteness. They then proceed to write the *software requirements specification* (SRS) document.

The project team to ensure that it accurately captures all the user requirements, and that it is understandable, consistent, unambiguous, and complete. The SRS document is then given to the customer for review. After the customer has reviewed the SRS document and agrees to it, it forms the basis for all future development activities and also serves as a contract document between the customer and the development organisation.

We can conceptually divide the requirements gathering and analysis activity into two separate tasks:

- Requirements gathering
- Requirements analysis

We discuss these two tasks in the following subsections.

Requirements Gathering

Requirements gathering is also popularly known as *requirements elicitation*. The primary objective of the requirements gathering task is to collect the requirements from the *stakeholders*.

A stakeholder is a source of the requirements and is usually a person, or a group of persons who either directly or indirectly are concerned with the software.

Requirements gathering may sound like a simple task. However, in practice it is very difficult to gather all the necessary information from a large number of stakeholders and from information scattered across several pieces of documents. Gathering requirements turns out to be especially challenging if there is no working model of the software being developed. Typically even before visiting the customer site, requirements gathering activity is started by studying the existing documents to collect all possible information about the system to be developed. During visit to the customer site, the analysts normally interview the end-users and customer representatives, carry out requirements gathering activities such as questionnaire surveys, task analysis, scenario analysis, and form analysis.

The goal of the requirements analysis and specification phase is to clearly understand the customer requirements and to systematically organise the requirements into a document called the Software Requirements Specification (SRS) document.

Good analysts share their experience and expertise with the customer and give his suggestions to define certain functionalities more comprehensively, make the functionalities more general and more complete. In the following, we briefly discuss the important ways in which an experienced analyst gathers requirements:

1.Studying existing documentation: The analyst usually studies all the available documents regarding the system to be developed before visiting the customer site. Customers usually provide statement of purpose (SoP) document to the developers. Typically these documents might discuss issues such as the context in which the software is required, the basic purpose, the stakeholders, features of any similar software developed elsewhere, etc.

2.Interview: Typically, there are many different categories of users of a software. Each category of users typically requires a different set of features from the software. Therefore, it is important for the analyst to first identify the different categories of users and then determine the requirements of each. For example, the different categories of users of a library automation software could be the library members, the librarians, and the accountants. The library members would like to use the software to query availability of books and issue and return books. The librarians might like to use the software to determine books that are overdue, create member accounts, delete member accounts, etc. The accounts personnel might use the software to invoke functionalities concerning financial aspects such as the total fee collected from the members, book procurement expenditures, staff salary expenditures, etc.

3.Task analysis: The users usually have a black-box view of a software and consider the software as something that provides a set of services (functionalities). A service supported by a software is also called a *task*. We can therefore say that the software performs various tasks of the users. In this context, the analyst tries to identify and understand the different tasks to be performed by the software. For each identified task, the analyst tries to formulate the different steps necessary to realise the required functionality in consultation with the users. For example, for the issue book service, the steps may be—authenticate user, check the number of books issued to the customer and determine if the maximum number of books that this member can borrow has been reached, check whether the book has been reserved, post the book issue details in the member's record, and finally print out a book issue slip that can be presented by the member at the security counter to take the book out of the library premises.

4.Scenario analysis: A task can have many scenarios of operation. The different scenarios of a task may take place when the task is invoked under different situations. For different types of scenarios of a task, the behaviour of the software can be different. For example, the possible scenarios for the book issue task of a library automation software may be:

- Book is issued successfully to the member and the book issue slip is printed.
- The book is reserved, and hence cannot be issued to the member.
- The maximum number of books that can be issued to the member is already reached, and no more books can be issued to the member.

- For various identified tasks, the possible scenarios of execution are identified and the details of each scenario is identified in consultation with the users. For each of the identified scenarios, details regarding system response, the exact conditions under which the scenario occurs, etc. are determined in consultation with the user.

5. Form analysis: Form analysis is an important and effective requirements gathering activity that is undertaken by the analyst, when the project involves automating an existing manual system. During the operation of a manual system, normally several forms are required to be filled up by the stakeholders, and in turn they receive several notifications (usually manually filled forms). In form analysis the existing forms and the formats of the notifications produced are analysed to determine the data input to the system and the data that are output from the system. For the different sets of data input to the system, how these input data would be used by the system to produce the corresponding output data is determined from the users.

Requirements Analysis:

After requirements gathering is complete, the analyst analyses the gathered requirements to form a clear understanding of the exact customer requirements and to weed out any problems in the gathered requirements. It is natural to expect that the data collected from various stakeholders to contain several contradictions, ambiguities, and incompleteness, since each stakeholder typically has only a partial and incomplete view of the software.

During requirements analysis, the analyst needs to identify and resolve three main types of problems in the requirements:

- Anomaly
- Inconsistency
- Incompleteness

Anomaly: It is an anomaly is an ambiguity in a requirement. When a requirement is anomalous, several interpretations of that requirement are possible. Any anomaly in any of the requirements can lead to the development of an incorrect system, since an anomalous requirement can be interpreted in the several ways during development.

Inconsistency: Two requirements are said to be inconsistent, if one of the requirements contradicts the other. The following are two examples of inconsistent requirements:

Incompleteness: An incomplete set of requirements is one in which some requirements have been overlooked. The lack of these features would be felt by the customer much later, possibly while using the software. Often, incompleteness is caused by the inability of the customer to visualise the system that is to be developed and to anticipate all the features that would be required. An experienced analyst can detect most of these missing features and suggest them to the customer for his consideration and approval for incorporation in the requirements.

Example: one of the clerks expressed the following—If a student secures a *grade point average* (GPA) of less than 6, then the parents of the student must be intimated about the regrettable performance through a (postal) letter as well as through e-mail. However, on an examination of all requirements, it was found that there is no provision by which either the postal or e-mail address of the parents of the students can be entered into the system. The feature that would allow entering the e-mail ids and postal addresses of the parents of the students was missing, thereby making the requirements incomplete.

Role of a system analyst

The analyst starts requirements gathering and analysis activity by collecting all information from the customer which could be used to develop the requirements of the system. He then analyzes the collected information to obtain a clear and thorough understanding of the product to be developed, with a view to removing all ambiguities and inconsistencies from the initial customer perception of the

problem. The following basic questions pertaining to the project should be clearly understood by the analyst in order to obtain a good grasp of the problem

- What is the problem?
- Why is it important to solve the problem?
- What are the possible solutions to the problem?
- What exactly are the data input to the system and what exactly are the data output by the system?
- What are the likely complexities that might arise while solving the problem?
- If there are external software or hardware with which the developed software has to interface, then what exactly would the data interchange formats with the external system be?

After the analyst has understood the exact customer requirements, he proceeds to identify and resolve the various requirements problems. The most important requirements problems that the analyst has to identify and eliminate are the problems of anomalies, inconsistencies, and incompleteness. When the analyst detects any inconsistencies, anomalies or incompleteness in the gathered requirements, he resolves them by carrying out further discussions with the end-users and the customers.

SOFTWARE REQUIREMENTS SPECIFICATION (SRS)

After the analyst has gathered all the required information regarding the software to be developed, and has removed all incompleteness, inconsistencies, and anomalies from the specification, he starts to systematically organise the requirements in the form of an SRS document.

Users of SRS Document

Usually a large number of different people need the SRS document for very different purposes. Some of the important categories of users of the SRS document and their needs for use are as follows:

- **Users, customers, and marketing personnel:** These stakeholders need to refer to the SRS document to ensure that the system as described in the document will meet their needs. Remember that the customer may not be the user of the software, but may be some one employed or designated by the user. For generic products, the marketing personnel need to understand the requirements that they can explain to the customers.
- **Software developers:** The software developers refer to the SRS document to make sure that they are developing exactly what is required by the customer.
- **Test engineers:** The test engineers use the SRS document to understand the functionalities, and based on this write the test cases to validate its working. They need that the required functionality should be clearly described, and the input and output data should have been identified precisely.
- **User documentation writers:** The user documentation writers need to read the SRS document to ensure that they understand the features of the product well enough to be able to write the users' manuals.
- **Project managers:** The project managers refer to the SRS document to ensure that they can estimate the cost of the project easily by referring to the SRS document and that it contains all the information required to plan the project.
- **Maintenance engineers:** The SRS document helps the maintenance engineers to understand the functionalities supported by the system. A clear knowledge of the functionalities can help them to understand the design and code.

A well-formulated SRS document finds a variety of usage other than the primary intended usage as a basis for starting the software development work. In the following subsection, we identify the important uses of a well-formulated SRS document:

- **Forms an agreement between the customers and the developers:** A good SRS document sets the stage for the customers to form their expectation about the software and the developers about what is expected from the software.
- **Reduces future reworks:** The process of preparation of the SRS document forces the stakeholders to rigorously think about all of the requirements before design and development get underway. This reduces later redesign, recoding, and retesting. Careful review of the SRS document can reveal omissions, misunderstandings, and inconsistencies early in the development cycle.
- **Provides a basis for estimating costs and schedules:** Project managers usually estimate the size of the software from an analysis of the SRS document. Based on this estimate they make other estimations such as the effort required to develop the software and the total cost of development. The SRS document also serves as a basis for price negotiations with the customer. The project manager also uses the SRS document for work scheduling.
- **Provides a baseline for validation and verification:** The SRS document provides a baseline against which compliance of the developed software can be checked. It is also used by the test engineers to create the *test plan*.
- **Facilitates future extensions:** The SRS document usually serves as a basis for planning future enhancements.

Traceability: Traceability means that it would be possible to identify (trace) the specific design component which implements a given requirement, the code part that corresponds to a given design component, and test cases that test a given requirement. Thus, any given code component can be traced to the corresponding design component, and a design component can be traced to a specific requirement that it implements and *vice versa*. Traceability analysis is an important concept and is frequently used during software development.

To achieve traceability, it is necessary that each functional requirement should be numbered uniquely and consistently. Proper numbering of the requirements makes it possible for different documents to uniquely refer to specific requirements.

Problems without a SRS document

- The important problems that an organization would face if it does not develop an SRS document are as follows:
- Without developing the SRS document, the system would not be implemented according to customer needs.
- Software developers would not know whether what they are developing is what exactly required by the customer.
- Without SRS document, it will be very much difficult for the maintenance engineers to understand the functionality of the system.
- It will be very much difficult for user document writers to write the users' manuals properly without understanding the SRS document.

Characteristics of a Good SRS Document

The skill of writing a good SRS document usually comes from the experience gained from writing SRS documents for many projects. However, the analyst should be aware of the desirable qualities that every good SRS document should possess. IEEE Recommended Practice for Software

Requirements Specifications[IEEE830] describes the content and qualities of a good software requirements specification (SRS).

Purpose: This section should describe where the software would be deployed and how the software would be used.

Project scope: This section should briefly describe the overall context within which the software is being developed. For example, the parts of a problem that are being automated and the parts that would need to be automated during future evolution of the software.

Environmental Characteristics: This section should briefly outline the environment (hardware and other software) with which the software will interact.

Some of the identified desirable qualities of an SRS document are the following:

- **Concise:** The SRS document should be concise and at the same time unambiguous, consistent, and complete. Verbose and irrelevant descriptions reduce readability and also increase the possibilities of errors in the document.
- **Implementation-independent:** The SRS should be free of design and implementation decisions unless those decisions reflect actual requirements. It should only specify what the system should do and refrain from stating how to do these. This means that the SRS document should specify the externally visible behaviour of the system and not discuss the implementation issues.

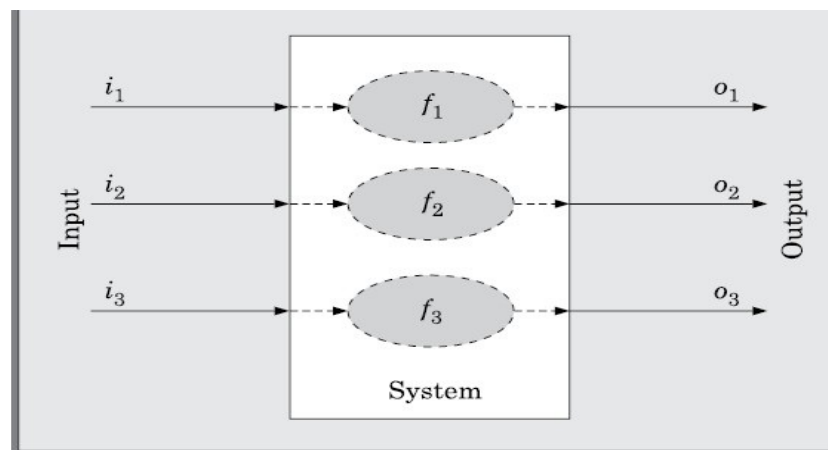


Fig: The black-box view of a system as performing a set of functions

The SRS document should describe the system to be developed as a black box, and should specify only the externally visible behaviour of the system. For this reason, the SRS document is also called the black-box specification of the software being

- **Traceable:** It should be possible to trace a specific requirement to the design elements that implement it and *vice versa*. Similarly, it should be possible to trace a requirement to the code segments that implement it and the test cases that test this requirement and *vice versa*. Traceability is also important to verify the results of a phase with respect to the previous phase and to analyse the impact of changing a requirement on the design elements and the code.
- **Modifiable:** Customers frequently change the requirements during the software development due to a variety of reasons. Therefore, in practice the SRS document undergoes several revisions during software development. Also, an SRS document is often modified after the project completes to accommodate future enhancements and evolution. To cope up with the requirements changes, the SRS document should be easily modifiable. For this, an SRS document should be well-structured. A well-structured document is easy to understand and modify. Having the description of a requirement scattered across many places in the SRS document may not be wrong—but it tends to make the requirement difficult to understand and also any modification to the

requirement would become difficult as it would require changes to be made at large number of places in the document.

- **Identification of response to undesired events:** The SRS document should discuss the system responses to various undesired events and exceptional conditions that may arise.
- **Verifiable:** All requirements of the system as documented in the SRS document should be verifiable. This means that it should be possible to design test cases based on the description of the functionality as to whether or not requirements have been met in an implementation. A requirement such as “the system should be user friendly” is not verifiable. On the other hand, the requirement —“When the name of a book is entered, the software should display whether the book is available for issue or it has been loaned out” is verifiable. Any feature of the required system that is not verifiable should be listed separately in the goals of the implementation section of the SRS document.

Attributes of Bad SRS Documents

SRS documents written by novices frequently suffer from a variety of problems. As discussed earlier, the most damaging problems are incompleteness, ambiguity, and contradictions.

- **Over-specification**
- **Forward references**
- **Wishful thinking**
- **Noise**

Important Categories of Customer Requirements

A good SRS document, should properly categorize and organise the requirements into different sections [IEEE830]. As per the IEEE 830 guidelines, the important categories of user requirements are the following. The IEEE 830 standard recommends that out of the various non-functional requirements, the external interfaces, and the design and implementation constraints should be documented in two different sections. The remaining non-functional requirements should be documented later in a section and these should include the performance and security requirements.

An SRS document should clearly document the following aspects of a software:

- Functional requirements
- Non-functional requirements
 - Design and implementation constraints
 - External interfaces required
 - Other non-functional requirements
- Goals of implementation.

Functional requirements:-

The functional requirements part discusses the functionalities required from the system. The system is considered to perform a set of high- level functions $\{f_i\}$. The functional view of the system is shown in fig. Each function f_i of the system can be considered as a transformation of a set of input data (ii) to the corresponding set of output data (O_i). The user can get some meaningful piece of work done using a high-level function.

Identifying functional requirements from a problem description:

The high-level functional requirements often need to be identified either from an informal problem description document or from a conceptual understanding of the problem. Each high-level requirement characterizes a way of system usage by some user to perform some meaningful piece of

work. There can be many types of users of a system and their requirements from the system may be very different. So, it is often useful to identify the different types of users who might use the system and then try to identify the requirements from each user's perspective.

Here we list all functions $\{f_i\}$ that the system performs. Each function f_i as shown in fig is considered as a transformation of a set of input data to some corresponding output data.



Fig: Function f_i

For documenting the functional requirements, we need to specify the set of functionalities supported by the system. A function can be specified by identifying the state at which the data is to be input to the system, its input data domain, the output data domain, and the type of processing to be carried on the input data to obtain the output data. Let us first try to document the withdraw-cash function of an ATM (Automated Teller Machine) system. The withdraw-cash is a high-level requirement. It has several sub-requirements corresponding to the different user interactions. These different interaction sequences capture the different scenarios.

Example: - Withdraw Cash from ATM R1:

Description: The withdraw cash function first determines the type of account that the user has and the account number from which the user wishes to withdraw cash. It checks the balance to determine whether the requested amount is available in the account. If enough balance is available, it outputs the required cash, otherwise it generates an error message.

R1.1 select withdraw amount option Input:

“withdraw amount” option

Output: user prompted to enter the account type

R1.2: select account type

Input: user option

Output: prompt to enter amount R1.3:

get required amount

Input: amount to be withdrawn in integer values greater than 100 and less than 10,000 in multiples of 100.

Output: The requested cash and printed transaction statement.

Processing: the amount is debited from the user's account if sufficient balance is available, otherwise an error message displayed.

Non-functional requirements:-

The non-functional requirements are non-negotiable obligations that must be supported by the software. The non-functional requirements capture those requirements of the customer that cannot be expressed as functions (i.e., accepting input data and producing output data). Non-functional requirements usually address aspects concerning external interfaces, user interfaces, maintainability, portability, usability, maximum number of concurrent users, timing, and throughput (transactions per second, etc.). The non-functional requirements can be critical in the sense that any failure by the developed software to achieve some minimum defined level in these requirements can be considered as a failure and make the software unacceptable by the customer. Nonfunctional requirements may include:

- # reliability issues,
- # accuracy of results,
- #performance issues
- # human - computer interface issues,
- #constraints on the system implementation,
- # security and maintainability of the system, etc.

In the following subsections, we discuss the different categories of non- functional requirements that are described under three different sections:

Design and implementation constraints: Design and implementation constraints are an important category of non-functional requirements describe any items or issues that will limit the options available to the developers. Some of the example constraints can be—corporate or regulatory policies that needs to be honoured; hardware limitations; interfaces with other applications; specific technologies, tools, and databases to be used; specific communications protocols to be used; security considerations; design conventions or programming standards to be followed, etc. Consider an example of a constraint that can be included in this section—Oracle DBMS needs to be used as this would facilitate easy interfacing with other applications that are already operational in the organisation.

External interfaces required: Examples of external interfaces are— hardware, software and communication interfaces, user interfaces, report formats, etc. To specify the user interfaces, each interface between the software and the users must be described. The description may include sample screen images, any GUI standards or style guides that are to be followed, screen layout constraints, standard buttons and functions(e.g., help) that will appear on every screen, keyboard shortcuts, error message display standards, and so on. One example of a user interface requirement of a software can be that it should be usable by factory shop floor workers who may not even have a high school degree. The details of the user interface design such as screen designs, menu structure, navigation diagram, etc. should be documented in a separate user interface specification document.

External interface requirements:

User interfaces: This section should describe a high-level description of various interfaces and various principles to be followed. The details of the user interface design should be documented in a separate user interface specification document.

Hardware interfaces: This section should describe the interface between the software and the hardware components of the system. This section may include the description of the supported device types, the nature of the data and control interactions between the software and the hardware, and the communication protocols to be used.

Software interfaces: This section should describe the connections between this software and other specific software components, including databases, operating systems, tools, libraries, and integrated commercial components, etc. Identify the data items that would be input to the software and the data that would be output should be identified and the purpose of each should be described.

Communications interfaces: This section should describe the requirements associated with any type of communications required by the software, such as e-mail, web access, network server communications protocols, etc. This section should define any pertinent message formatting to be used. It should also identify any communication standards that will be used, such as TCP sockets, FTP, HTTP, or SHTTP. Specify any communication security or encryption issues that may be relevant, and also the data transfer rates, and synchronisation mechanisms.

Other non-functional requirements: This section contains a description of non-functional requirements that are neither design constraints and nor are external interface requirements. An important example is a performance requirement such as the number of transactions completed per unit time. Besides performance requirements, the other non-functional requirements to be described in this section may include reliability issues, accuracy of results, and security issues.

Performance requirements: Aspects such as number of transaction to be completed per second should be specified here. Some performance requirements may be specific to individual functional requirements or features. These should also be specified here.

Safety requirements: Those requirements that are concerned with possible loss or damage that could result from the use of the software are specified here. For example, recovery after power failure, handling software and hardware failures, etc. may be documented here.

Security requirements: This section should specify any requirements regarding security or privacy requirements on data used or created by the software. Any user identity authentication requirements should be described here. It should also refer to any external policies or regulations concerning the security issues. Define any security or privacy certifications that must be satisfied.

Goals of implementation:-

The goals of implementation part documents some general suggestions regarding development. These suggestions guide trade-off among design goals. The goals of implementation section might document issues such as revisions to the system functionalities that may be required in the future, new devices to be supported in the future, reusability issues, etc. These are the items which the developers might keep in their mind during development so that the developed system may meet some aspects that are not required immediately.

The different scenarios occur depending on the amount entered for withdrawal. The different scenarios are essentially different behaviour exhibited by the system for the same high-level function. Typically, each user input and the corresponding system action may be considered as a sub-requirement of a high-level requirement. Thus, each high-level requirement can consist of several sub-requirements.

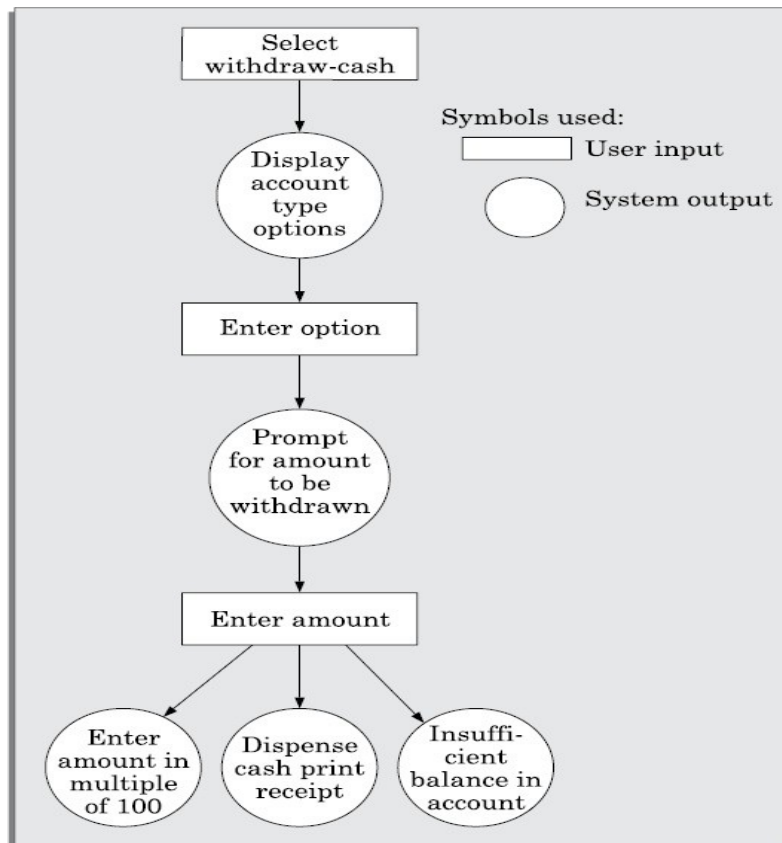


Figure: User and system interactions in high-level functional requirement.

Overall description of organisation of SRS document

Product perspective: This section needs to briefly state as to whether the software is intended to be a replacement for a certain existing systems, or it is a new software. If the software being developed would be used as a component of a larger system, a simple schematic diagram can be given to show the major components of the overall system, subsystem interconnections, and external interfaces can be helpful.

Product features: This section should summarize the major ways in which the software would be used. Details should be provided in Section 3 of the document. So, only a brief summary should be presented here.

User classes: Various user classes that are expected to use this software are identified and described here. The different classes of users are identified by the types of functionalities that they are expected to invoke, or their levels of expertise in using computers.

Operating environment: This section should discuss in some detail the hardware platform on which the software would run, the operating system, and other application software with which the developed software would interact.

Design and implementation constraints: In this section, the different constraints on the design and implementation are discussed. These might include—corporate or regulatory policies; hardware limitations (timing requirements, memory requirements); interfaces to other applications; specific technologies, tools, and databases to be used; specific programming language to be used; specific communication protocols to be used; security considerations; design conventions or programming standards.

User documentation: This section should list out the types of user documentation, such as user manuals, on-line help, and trouble-shooting manuals that will be delivered to the customer along with the software.

A good SRS document should properly characterise the conditions under which different scenarios of interaction occur. That is, a high-level function might involve different steps to be undertaken as a consequence of some decisions made after each step. Sometimes the conditions can be complex and numerous and several alternative interaction and processing sequences may exist depending on the outcome of the corresponding condition checking. A simple text description in such cases can be difficult to comprehend and analyse. In such situations, a decision tree or a decision table can be used to represent the logic and the processing involved. Also, when the decision making in a functional requirement has been represented as a decision table, it becomes easy to automatically or at least manually design test cases.

There are two main techniques available to analyse and represent complex processing logic—decision trees and decision tables. Once the decision making logic is captured in the form of trees or tables, the test cases to validate these logic can be automatically obtained. It should, however, be noted that decision trees and decision tables have much broader applicability than just specifying complex processing logic in an SRS document. For instance, decision trees and decision tables find applications in information theory and switching theory.

Decision tree:

A decision tree gives a graphic view of the processing logic involved in decision making and the corresponding actions taken. The edges of a decision tree represent conditions and the leaf nodes represent the actions to be performed depending on the outcome of testing the condition.

Example: -Consider Library Membership Automation Software (LMS) where it should support the following three options:

- New member
- Renewal
- Cancel membership

New member option-

Decision: When the 'new member' option is selected, the software asks details about the member like the member's name, address, phone number etc.

Action: If proper information is entered then a membership record for the member is created and a bill is printed for the annual membership charge plus the security deposit payable.

Renewal option-

Decision: If the 'renewal' option is chosen, the LMS asks for the member's name and his membership number to check whether he is a valid member or not.

Action: If the membership is valid then membership expiry date is updated and the annual membership bill is printed, otherwise an error message is displayed.

Cancel membership option-

Decision: If the 'cancel membership' option is selected, then the software asks for member's name and his membership number.

Action: The membership is cancelled, a cheque for the balance amount due to the member is printed and finally the membership record is deleted from the database.

Decision tree representation of the above example -

The following tree shows the graphical representation of the above example. After getting information from the user, the system makes a decision and then performs the corresponding actions.

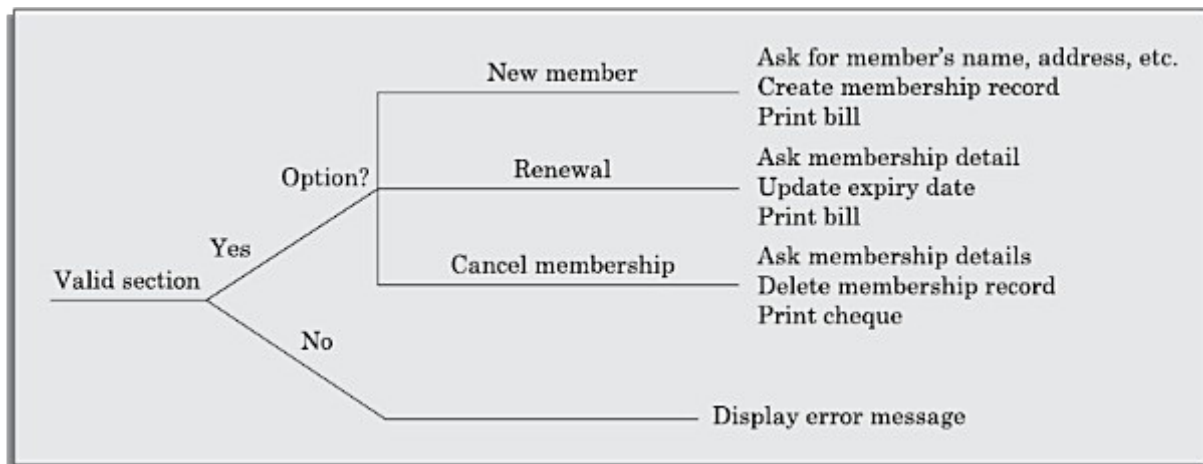


Fig: Decision tree for LMS

Decision table:

A decision table is used to represent the complex processing logic in a tabular or a matrix form. The upper rows of the table specify the variables or conditions to be evaluated. The lower rows of the table specify the actions to be taken when the corresponding conditions are satisfied. A column in a table is called a *rule*. A rule implies that if a condition is true, then the corresponding action is to be executed.

Example: -Consider the previously discussed LMS example. The following decision table (fig. 3.5) shows how to represent the LMS problem in a tabular form. Here the table is divided into two parts, the upper part shows the conditions and the lower part shows what actions are taken. Each column of the table is a rule.

Conditions				
Valid selection	No	Yes	Yes	Yes
New member	-	Yes	No	No
Renewal	-	No	Yes	No
Cancellation	-	No	No	Yes
Actions				
Display error message	x	-	-	-
Ask member's details	-	x	-	-
Build customer record	-	x	-	-
Generate bill	-	x	x	-
Ask member's name & membership number	-	-	x	x
Update expiry date	-	-	x	-
Print cheque	-	-	-	x
Delete record	-	-	-	x

Fig: Decision table for LMS

From the above table you can easily understand that, if the valid selection condition is false then the action taken for this condition is 'display error message'. Similarly, the actions taken for other conditions can be inferred from the table.

Decision table *versus* decision tree

Even though both decision tables and decision trees can be used to represent complex program logic, they can be distinguishable on the following three considerations:

Readability: Decision trees are easier to read and understand when the number of conditions are small. On the other hand, a decision table causes the analyst to look at every possible combination of conditions which he might otherwise omit.

Explicit representation of the order of decision making: In contrast to the decision trees, the order of decision making is abstracted out in decision tables. A situation where decision tree is more useful is when multilevel decision making is required. Decision trees can more intuitively represent multilevel decision making hierarchically, whereas decision tables can only represent a single decision to select the appropriate action for execution.

Representing complex decision logic: Decision trees become very complex to understand when the number of conditions and actions increase. It may even be to draw the tree on a single page. When very large number of decisions are involved, the decision table representation may be preferred.

Formal technique

A formal technique is a mathematical method to specify a hardware and/or software system, verify whether a specification is realizable, verify that an implementation satisfies its specification, prove properties of a system without necessarily running the system, etc. The mathematical basis of a formal method is provided by the specification language.

Formal specification language

A formal specification language consists of two sets syn and sem , and a relation sat between them. The set syn is called the syntactic domain, the set sem is called the semantic domain, and the relation sat is called the satisfaction relation. For a given specification syn , and model of the system sem , if $sat(syn, sem)$, as shown in figure then syn is said to be the specification of sem , and sem is said to be the specificand of syn .

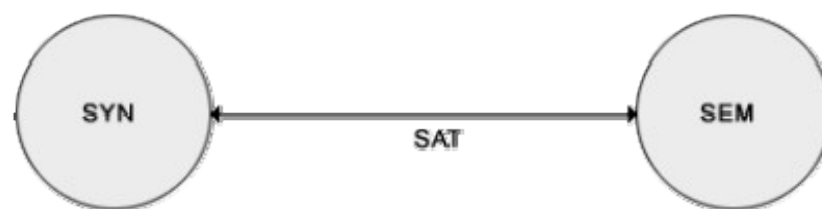


Fig: $sat(syn, sem)$

Syntactic Domains

The syntactic domain of a formal specification language consists of an alphabet of symbols and set of formation rules to construct well-formed formulas from the alphabet. The well-formed formulas are used to specify a system.

Semantic Domains

Formal techniques can have considerably different semantic domains. Abstract data type specification languages are used to specify algebras, theories, and programs. Programming languages are used to specify functions from input to output values. Concurrent and distributed system

specification languages are used to specify state sequences, event sequences, state-transition sequences, synchronization trees, partial orders, state machines, etc.

Satisfaction Relation

Given the model of a system, it is important to determine whether an element of the semantic domain satisfies the specifications. This satisfaction is determined by using a homomorphism known as semantic abstraction function. The semantic abstraction function maps the elements of the semantic domain into equivalent classes. There can be different specifications describing different aspects of a system model, possibly using different specification languages. Some of these specifications describe the system's behavior and the others describe the system's structure. Consequently, two broad classes of semantic abstraction functions are defined: those that preserve a system's behavior and those that preserve a system's structure.

Model-oriented vs. property-oriented approaches

Formal methods are usually classified into two broad categories – model-oriented and property – oriented approaches. In a model-oriented style, one defines a system's behavior directly by constructing a model of the system in terms of mathematical structures such as tuples, relations, functions, sets, sequences, etc. In the property-oriented style, the system's behavior is defined indirectly by stating its properties, usually in the form of a set of axioms that the system must satisfy.

Example:-

Let us consider a simple producer/consumer example. In a property- oriented style, it is probably started by listing the properties of the system like: the consumer can start consuming only after the producer has produced an item, the producer starts to produce an item only after the consumer has consumed the last item, etc. A good example of a producer-consumer problem is CPU-Printer coordination. After processing of data, CPU outputs characters to the buffer for printing. Printer, on the other hand, reads characters from the buffer and prints them. The CPU is constrained by the capacity of the buffer, whereas the printer is constrained by an empty buffer. Examples of property-oriented specification styles are axiomatic specification and algebraic specification.

In a model-oriented approach, we start by defining the basic operations, p (produce) and c (consume). Then we can state that $S_1 + p \rightarrow S$, $S + c \rightarrow S_1$. Thus the model-oriented approaches essentially specify a program by writing another, presumably simpler program. Examples of popular model-oriented specification techniques are Z, CSP, CCS, etc.

Model-oriented approaches are more suited to use in later phases of life cycle because here even minor changes to a specification may lead to drastic changes to the entire specification. They do not support logical conjunctions (AND) and disjunctions (OR).

Property-oriented approaches are suitable for requirements specification because they can be easily changed. They specify a system as a conjunction of axioms and you can easily replace one axiom with another one.

Operational semantics

Informally, the operational semantics of a formal method is the way computations are represented. There are different types of operational semantics according to what is meant by a single run of the system and how the runs are grouped together to describe the behavior of the system. Some commonly used operational semantics are as follows:

Linear Semantics:-

In this approach, a run of a system is described by a sequence (possibly infinite) of events or states. The concurrent activities of the system are represented by non-deterministic interleavings of the atomic actions. For example, a concurrent activity $a \parallel b$ is represented by the set of sequential activities $a;b$ and $b;a$. This is simple but rather unnatural representation of concurrency. The behavior of

a system in this model consists of the set of all its runs. To make this model realistic, usually justice and fairness restrictions are imposed on computations to exclude the unwanted interleavings.

Branching Semantics:-

In this approach, the behavior of a system is represented by a directed graph as shown in the figure. The nodes of the graph represent the possible states in the evolution of a system. The descendants of each node of the graph represent the states which can be generated by any of the atomic actions enabled at that state. An example involving the transactions in an ATM is shown in fig. Although this semantic model distinguishes the branching points in a computation, still it represents concurrency by interleaving.

Maximally parallel semantics:-

In this approach, all the concurrent actions enabled at any state are assumed to be taken together. This is again not a natural model of concurrency since it implicitly assumes the availability of all the required computational resources.

Partial order semantics:-

Under this view, the semantics ascribed to a system is a structure of states satisfying a partial order relation among the states (events). The partial order represents a precedence ordering among events, and constraints some events to occur only after some other events have occurred; while the occurrence of other events (called concurrent events) is considered to be incomparable. This fact identifies concurrency as a phenomenon not translatable to any interleaved representation.

For example, figure shows the semantics implied by a simplified beverage selling machine. From the figure, we can infer that beverage is dispensed only if an inserted coin is accepted by the machine (precedence). Similarly, preparation of ingredients and milk are done simultaneously (concurrency). Hence, node Ingredient can be compared with node Brew, but neither can it be compared with node Hot/Cold nor with node Accepted.

Merits of formal requirements specification

Formal methods possess several positive features, some of which are discussed below.

- Formal specifications encourage rigour. Often, the very process of construction of a rigorous specification is more important than the formal specification itself. The construction of a rigorous specification clarifies several aspects of system behavior that are not obvious in an informal specification.
- Formal methods usually have a well-founded mathematical basis. Thus, formal specifications are not only more precise, but also mathematically sound and can be used to reason about the properties of a specification and to rigorously prove that an implementation satisfies its specifications.
- Formal methods have well-defined semantics. Therefore, ambiguity in specifications is automatically avoided when one formally specifies a system.
- The mathematical basis of the formal methods facilitates automating the analysis of specifications. For example, a tableau-based technique has been used to automatically check the consistency of specifications. Also, automatic theorem proving techniques can be used to verify that an implementation satisfies its specifications. The possibility of automatic verification is one of the most important advantages of formal methods.
- Formal specifications can be executed to obtain immediate feedback on the features of the specified system. This concept of executable specifications is related to rapid prototyping. Informally, a prototype is a “toy” working model of a system that can provide immediate feedback on the behavior of the specified system, and is especially useful in checking the completeness of specifications.

Limitations of formal requirements specification

- It is clear that formal methods provide mathematically sound frameworks within large, complex systems can be specified, developed and verified in a systematic rather than in an ad hoc manner. However, formal methods suffer from several shortcomings, some of which are the following:
- Formal methods are difficult to learn and use.
- The basic incompleteness results of first-order logic suggest that it is impossible to check absolute correctness of systems using theorem proving techniques.
- Formal techniques are not able to handle complex problems. This shortcoming results from the fact that, even moderately complicated problems blow up the complexity of formal specification and their analysis. Also, a large unstructured set of mathematical formulas is difficult to comprehend.

Axiomatic specification

- In axiomatic specification of a system, first-order logic is used to write the pre and post-conditions to specify the operations of the system in the form of axioms. The pre-conditions basically capture the conditions that must be satisfied before an operation can successfully be invoked. In essence, the pre-conditions capture the requirements on the input parameters of a function. The post-conditions are the conditions that must be satisfied when a function completes execution for the function to be considered to have executed successfully. Thus, the post-conditions are essentially constraints on the results produced for the function execution to be considered successful.
- The following are the sequence of steps that can be followed to systematically develop the axiomatic specifications of a function:
- Establish the range of input values over which the function should behave correctly. Also find out other constraints on the input parameters and write it in the form of a predicate.
- Specify a predicate defining the conditions which must hold on the output of the function if it behaved properly.
- Establish the changes made to the function's input parameters after execution of the function. Pure mathematical functions do not change their input and therefore this type of assertion is not necessary for pure functions.
- Combine all of the above into pre and post conditions of the function.

Example1: -Specify the pre- and post-conditions of a function that takes a real number as argument and returns half the input value if the input is less than or equal to 100, or else returns double the value.

$f(x : \text{real}) : \text{real}$

pre : $x \in \mathbb{R}$

post : $\{(x \leq 100) \wedge (f(x) = x/2)\} \vee \{(x > 100) \wedge (f(x) = 2 * x)\}$

Example2: Axiomatically specify a function named search which takes an integer array and an integer key value as its arguments and returns the index in the array where the key value is present.

search($X : \text{IntArray}, \text{key} : \text{Integer}$) : Integer

pre : $\exists i \in [X_{\text{first}} \dots X_{\text{last}}], X[i] = \text{key}$

post : $\{(X'[\text{search}(X, \text{key})] = \text{key}) \wedge (X = X')\}$

Here the convention followed is: If a function changes any of its input parameters and if that parameter is named X, then it is referred to as X' after the function completes execution.

Algebraic specification

In the algebraic specification technique an object class or type is specified in terms of relationships existing between the operations defined on that type. It was first brought into prominence by Guttag [1980, 1985] in specification of abstract data types. Various notations of algebraic specifications have evolved, including those based on OBJ and Larch languages.

Representation of algebraic specification:

Essentially, algebraic specifications define a system as a heterogeneous algebra. A heterogeneous algebra is a collection of different sets on which several operations are defined. Traditional algebras are homogeneous. A homogeneous algebra consists of a single set and several operations; {I, +, -, *, /}. In contrast, alphabetic strings together with operations of concatenation and length {A, I, con, len}, is not a homogeneous algebra, since the range of the length operation is the set of integers.

Each set of symbols in the algebra, in turn, is called a *sort* of the algebra. To define a heterogeneous algebra, we first need to specify its signature, the involved operations, and their domains and ranges. Using algebraic specification, we define the meaning of a set of interface procedure by using equations. An algebraic specification is usually presented in four sections.

Types section: In this section, the sorts (or the data types) being used is specified.

Exception section: This section gives the names of the exceptional conditions that might occur when different operations are carried out. These exception conditions are used in the later sections of an algebraic specification. For example, in a queue, possible exceptions are novalue (empty queue), underflow (removal from an empty queue), etc.

Syntax section: This section defines the signatures of the interface procedures. The collection of sets that form input domain of an operator and the sort where the output is produced are called the *signature* of the operator. For example, PUSH takes a stack and an element as its input and returns a new stack that has been created.

append : queue x element → queue

Equations section: This section gives a set of *rewrite rules* (or equations) defining the meaning of the interface procedures in terms of each other. In general, this section is allowed to contain conditional expressions. By convention each equation is implicitly universally quantified over all possible values of the variables. Names not mentioned in the syntax section such as 'r' or 'e' are variables. The first step in defining an algebraic specification is to identify the set of required operations. After having identified the required operators, it is helpful to classify them as either basic constructor operators, extra constructor operators, basic inspector operators, or extra inspection operators. The definition of these categories of operators is as follows:

Basic construction operators. These operators are used to create or modify entities of a type. The basic construction operators are essential to generate all possible element of the type being specified. For example, 'create' and 'append' are basic construction operators for a FIFO *queue*.

Extra construction operators. These are the construction operators other than the basic construction operators. For example, the operator 'remove' is an extra construction operator for a FIFO queue because even without using 'remove', it is possible to generate all values of the type being specified

Basic inspection operators. These operators evaluate attributes of a type without modifying them, e.g., eval, get, etc. Let S be the set of operators whose range is not the data type being specified. The set of the basic operators S1 is a subset of S, such that each operator from S-S1 can be expressed in

terms of the operators from S1. For example, 'isempty' is a basic inspection operator because it does not modify the FIFO queue type.

Extra inspection operators. These are the inspection operators that are not basic inspectors.

A good rule of thumb while writing an algebraic specification, is to first establish which are the constructor (basic and extra) and inspection operators (basic and extra). Then write down an axiom for composition of each basic construction operator over each basic inspection operator and extra constructor operator. Also, write down an axiom for each of the extra inspector in terms of any of the basic inspectors. Thus, if there are m_1 basic constructors, m_2 extra constructors, n_1 basic inspectors, and n_2 extra inspectors, we should have $m_1 \times (m_2 + n_1) + n_2$ axioms are the minimum required and many more axioms may be needed to make the specification complete. Using a complete set of rewrite rules, it is possible to simplify an arbitrary sequence of operations on the interface procedures.

Develop algebraic specification of simple problems

- The first step in defining an algebraic specification is to identify the set of required operations. After having identified the required operators, it is helpful to classify them into different categories.
- A simple way to determine whether an operator is a constructor (basic or extra) or an inspector (basic or extra) is to check the syntax expression for the operator. If the type being specified appears on the right hand side of the expression then it is a constructor, otherwise it is an inspection operator. For example, in a FIFO queue, 'create' is a constructor because the data type specified 'queue' appears on the right hand side of the expression. But, 'first' and 'isempty' are inspection operators since they do not modify the *queue* data type.

Example 1:-

Let us specify a data type point supporting the operations create, xcoord, ycoord, isequal; where the operations have their usual meaning.

Types:

```
defines point
uses boolean, integer
```

Syntax:

- 1.create : integer \times integer \rightarrow point
- 2.xcoord : point \rightarrow integer
- 3.ycoord : point \rightarrow integer
- 4.isequal : point \times point \rightarrow boolean

Equations:

- 1.xcoord(create(x, y)) = x
- 2.ycoord(create(x, y)) = y
- 3.isequal(create(x1, y1), create(x2, y2)) = ((x1 = x2) and (y1 = y2))

In this example, we have only one basic constructor (create), and three basic inspectors (xcoord, ycoord, and isequal). Therefore, we have only 3 equations.

The rewrite rules let you determine the meaning of any sequence of calls on the point type. Consider the following expression: *isequal (create (xcoord (create(2, 3)), 5), create (ycoord (create(2, 3)), 5))*. By applying the rewrite rule 1, you can simplify the given expression as *isequal (create (2, 5), create (ycoord (create(2, 3)), 5))*. By using rewrite rule 2, you can further simplify this as *isequal (create (2, 5), create (3, 5))*. This is false by rewrite rule 3.

Example 2:

Let us specify a FIFO queue supporting the operations *create*, *append*, *remove*, *first*, and *isempty* where the operations have their usual meaning.

Types:

defines queue

uses boolean, integer

Exceptions:

underflow, novalue

Syntax:

- 1.create : * queue
- 2.append : queue x element * queue
- 3.remove : queue * queue + {underflow}
- 4.first : queue * element + {novalue}
- 5.isempty : queue * boolean

Equations:

- 1.isempty(create()) = true
- 2.isempty((append(q,e)) = false
- 3.first(create()) = novalue
- 4.first(append(q,e)) = if isempty(q) then e else first(q)
- 5.remove(create()) = underflow
- 6.remove(append(q,e)) = if isempty(q) then create() else append(remove(q),e)

In this example, there are two basic constructors (create and append), one extra construction operator (remove) and two basic inspectors (first and empty). Therefore, there are $2 \times (1+2) + 0 = 6$ equations.

Properties of algebraic specifications

Three important properties that every algebraic specification should possess are:

- **Completeness:** This property ensures that using the equations, it should be possible to reduce any arbitrary sequence of operations on the interface procedures. There is no simple procedure to ensure that an algebraic specification is complete.
- **Finite termination property:** This property essentially addresses the following question: Do applications of the rewrite rules to arbitrary expressions involving the interface procedures always terminate? For arbitrary algebraic equations, convergence (finite termination) is undecidable. But, if the right hand side of each rewrite rule has fewer terms than the left, then the rewrite process must terminate.
- **Unique termination property:** This property indicates whether application of rewrite rules in different orders always result in the same answer. Essentially, to determine this property, the answer to the following question needs to be checked: Can all possible sequence of choices in application of the rewrite rules to an arbitrary expression involving the interface procedures always give the same number? Checking the unique termination property is a very difficult problem.

Structured specification: Developing algebraic specifications is time consuming. Therefore efforts have been made to devise ways to ease the task of developing algebraic specifications. The following are some of the techniques that have successfully been used to reduce the effort in writing the specifications.

- **Incremental specification.** The idea behind incremental specification is to first develop the specifications of the simple types and then specify more complex types by using the specifications of the simple types.

- **Specification instantiation.** This involves taking an existing specification which has been developed using a generic parameter and instantiating it with some other sort.

Advantages and disadvantages of algebraic specifications

Algebraic specifications have a strong mathematical basis and can be viewed as heterogeneous algebra. Therefore, they are unambiguous and precise. Using an algebraic specification, the effect of any arbitrary sequence of operations involving the interface procedures can automatically be studied. A major shortcoming of algebraic specifications is that they cannot deal with side effects. Therefore, algebraic specifications are difficult to interchange with typical programming languages. Also, algebraic specifications are hard to understand.

Executable specification language (4GLs).

If the specification of a system is expressed formally or by using a programming language, then it becomes possible to directly execute the specification. However, executable specifications are usually slow and inefficient, 4GLs³ (4th Generation Languages) are examples of executable specification languages. 4GLs are successful because there is a lot of commonality across data processing applications. 4GLs rely on software reuse, where the common abstractions have been identified and parameterized. Careful experiments have shown that rewriting 4GL programs in higher level languages results in up to 50% lower memory usage and also the program execution time can reduce ten folds. Example of a 4GL is Structured Query Language (SQL).

Software design and its activities

Software design deals with transforming the customer requirements, as described in the SRS document, into a form (a set of documents) that is suitable for implementation in a programming language. A good software design is seldom arrived by using a single step procedure but rather through several iterations through a series of steps. Design activities can be broadly classified into two important parts:

- Preliminary (or high-level) design and
- Detailed design

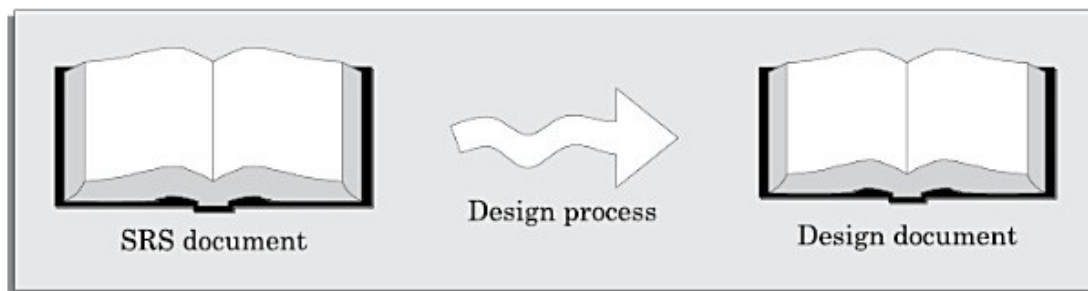


Fig: The design process.

Through high-level design, a problem is decomposed into a set of modules. The control relationships among the modules are identified, and also the interfaces among various modules are identified.

The following items are designed and documented during the design phase.

- **Different modules required:** The different modules in the solution should be clearly identified. Each module is a collection of functions and the data shared by the functions of the module. Each module should accomplish some well-defined task out of the overall responsibility of the software. Each module should be named according to the task it performs. For example, in an academic automation software, the module consisting of the functions and data necessary to accomplish the task of registration of the students should be named handle student registration.
- **Control relationships among modules:** A control relationship between two modules essentially arises due to function calls across the two modules. The control relationships existing among various modules should be identified in the design document.
- **Interfaces among different modules:** The interfaces between two modules identifies the exact data items that are exchanged between the two modules when one module invokes a function of the other module.
- **Data structures of the individual modules:** Each module normally stores some data that the functions of the module need to share to accomplish the overall responsibility of the module. Suitable data structures for storing and managing the data of a module need to be properly designed and documented.
- **Algorithms required to implement the individual modules:** Each function in a module usually performs some processing activity. The algorithms required to accomplish the processing activities of various modules need to be carefully designed and documented with due considerations given to the accuracy of the results, space and time complexities.

The outcome of high-level design is called the program structure or the software architecture. A notation that is widely being used for procedural development is a tree-like diagram called the structure chart. Another popular design representation techniques called UML that is being used to document object-oriented design, involves developing several types of diagrams to document the object-oriented design of a systems.

Characteristics of a good software design

The definition of “a good software design” can vary depending on the application being designed. For example, the memory size used by a program may be an important issue to characterize a good solution for embedded software development – since embedded applications are often required to be implemented using memory of limited size due to cost, space, or power consumption considerations. For embedded applications, one may sacrifice design comprehensibility to achieve code compactness. For embedded applications, factors like design comprehensibility may take a back seat while judging the goodness of design. Therefore, the criteria used to judge how good a given design solution is can vary widely depending upon the application. Not only is the goodness of design dependent on the targeted application, but also the notion of goodness of a design itself varies widely across software engineers and academicians. However, most researchers and software engineers agree on a few desirable characteristics that every good software design for general application must possess. The characteristics are listed below:

- **Correctness:** A good design should correctly implement all the functionalities identified in the SRS document.
- **Understandability:** A good design is easily understandable
- **Efficiency:** It should be efficient.
- **Maintainability:** It should be easily amenable to change.

A design solution is correct, understandability of a design is possibly the most important issue to be considered while judging the goodness of a design. A design that is easy to understand is also easy to develop, maintain and change. Thus, unless a design is easily understandable, it would require tremendous effort to implement and maintain it.

Features of a design document:

In order to facilitate understandability, the design should have the following features:

- It should use consistent and meaningful names for various design components.
- The design should be modular. The term modularity means that it should use a cleanly decomposed set of modules. It should neatly arrange the modules in a hierarchy, e.g. in a tree-like diagram.

Cohesion

Most researchers and engineers agree that a good software design implies clean decomposition of the problem into modules, and the neat arrangement of these modules in a hierarchy. The primary characteristics of neat module decomposition are high cohesion and low coupling. Cohesion is a measure of functional strength of a module. A module having high cohesion and low coupling is said to be functionally independent of other modules. By the term functional independence, we mean that a cohesive module performs a single task or function. A functionally independent module has minimal interaction with other modules.

Classification of cohesion

The different classes of cohesion that a module may possess are depicted

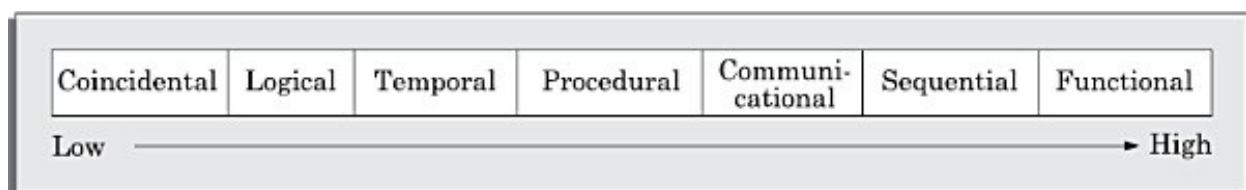


Fig:Classification of cohesion.

Coincidental cohesion: A module is said to have coincidental cohesion, if it performs a set of tasks that relate to each other very loosely, if at all. In this case, the module contains a random collection of functions. It is likely that the functions have been put in the module out of pure coincidence without any thought or design. For example, in a transaction processing system (TPS), the get-input, print-error, and summarize-members functions are grouped into one module. The grouping does not have any relevance to the structure of the problem.

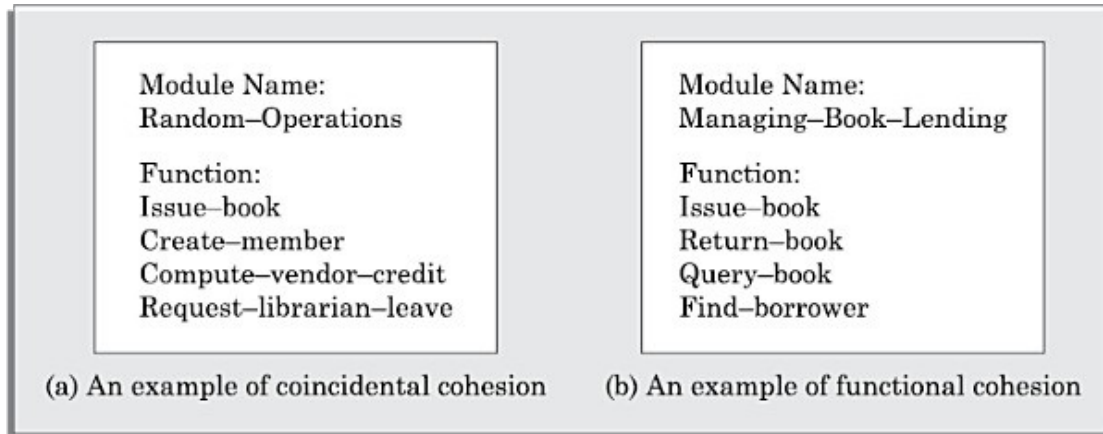


Fig: Examples of cohesion

Logical cohesion: A module is said to be logically cohesive, if all elements of the module perform similar operations, e.g. error handling, data input, data output, etc. An example of logical cohesion is the case where a set of print functions generating different output reports are arranged into a single module. As an example of logical cohesion, consider a module that contains a set of print functions to generate various types of output reports such as grade sheets, salary slips, annual reports, etc.

Temporal cohesion: When a module contains functions that are related by the fact that all the functions must be executed in the same time span, the module is said to exhibit temporal cohesion. The set of functions responsible for initialization, start-up, shutdown of some process, etc. exhibit temporal cohesion.

Procedural cohesion: A module is said to possess procedural cohesion, if the set of functions of the module are all part of a procedure (algorithm) in which certain sequence of steps have to be carried out for achieving an objective, e.g. the algorithm for decoding a message. The functions login(), place-order(), check-order(), print-bill(), place-order-on-vendor(), update-inventory(), and logout() all do different things and operate on different data.

Communicational cohesion: A module is said to have communicational cohesion, if all functions of the module refer to or update the same data structure, e.g. the set of functions defined on an array or a stack. As an example of procedural cohesion, consider a module named student in which the different functions in the module such as admitStudent, enterMarks, printGradeSheet, etc. access and manipulate data stored in an array named studentRecords defined within the module.

Sequential cohesion: A module is said to possess sequential cohesion, if the elements of a module form the parts of a sequence, where the output from one element of the sequence is input to the next. For example, in a TPS, the get-input, validate-input, sort-input functions are grouped into one module. As an example, consider the following situation. In an on-line store, consider that after a customer requests for some item, it is first determined if the item is in stock. In this case, if the functions create-order(), check-item-availability(), place-order-on-vendor() are placed in a single module, then the module would exhibit sequential cohesion.

Functional cohesion: Functional cohesion is said to exist, if different elements of a module cooperate to achieve a single function. For example, a module containing all the functions required to manage employees' pay-roll exhibits functional cohesion. Suppose a module exhibits functional cohesion and we are asked to describe what the module does, then we would be able to describe it using a single sentence.

Coupling

Coupling between two modules is a measure of the degree of interdependence or interaction between the two modules. A module having high cohesion and low coupling is said to be functionally independent of other modules. If two modules interchange large amounts of data, then they are highly interdependent. The degree of coupling between two modules depends on their interface complexity. The interface complexity is basically determined by the number of types of parameters that are interchanged while invoking the functions of the module.

Classification of Coupling

Even if there are no techniques to precisely and quantitatively estimate the coupling between two modules, classification of the different types of coupling will help to quantitatively estimate the degree of coupling between two modules. Five types of coupling can occur between any two modules.

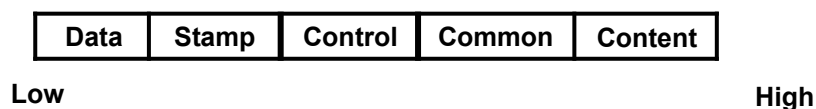


Fig: Classification of coupling

Data coupling: Two modules are data coupled, if they communicate through a parameter. An example is an elementary data item passed as a parameter between two modules, e.g. an integer, a float, a character, etc. This data item should be problem related and not used for the control purpose.

Stamp coupling: Two modules are stamp coupled, if they communicate using a composite data item such as a record in PASCAL or a structure in C.

Control coupling: Control coupling exists between two modules, if data from one module is used to direct the order of instructions execution in another. An example of control coupling is a flag set in one module and tested in another module.

Common coupling: Two modules are common coupled, if they share data through some global data items.

Content coupling: Content coupling exists between two modules, if they share code, e.g. a branch from one module into another module.

Control Hierarchy:

Layered Arrangement Of Modules:

In a layered design solution, the modules are arranged into several layers based on their call relationships. A module is allowed to call only the modules that are at a lower layer. That is, a module should not call a module that is either at a higher layer or even in the same layer. In a layered design, the top-most module in the hierarchy can be considered as a manager that only invokes the services of the lower level module to discharge its responsibility.

Superordinate and subordinate modules: In a control hierarchy, a module that controls another module is said to be superordinate to it. Conversely, a module controlled by another module is said to be subordinate to the controller.

Visibility: A module B is said to be visible to another module A, if A directly calls B. Thus, only the immediately lower layer modules are said to be visible to a module.

Control abstraction: In a layered design, a module should only invoke the functions of the modules that are in the layer immediately below it. In other words, the modules at the higher layers, should not be visible (that is, abstracted out) to the modules at the lower layers. This is referred to as control abstraction.

Depth and width: Depth and width of a control hierarchy provide an indication of the number of layers and the overall span of control respectively.

Fan-out: Fan-out is a measure of the number of modules that are directly controlled by a given module. In the fan-out of the module M1 is 3. A design in which the modules have very high fan-out numbers is not a good design. The reason for this is that a very high fan-out is an indication that the module lacks cohesion. A module having a large fan-out (greater than 7) is likely to implement several different functions and not just a single cohesive function.

Fan-in: Fan-in indicates the number of modules that directly invoke a given module. High fan-in represents code reuse and is in general, desirable in a good design. In the fan-in of the module M1 is 0, that of M2 is 1, and that of M5 is 2.

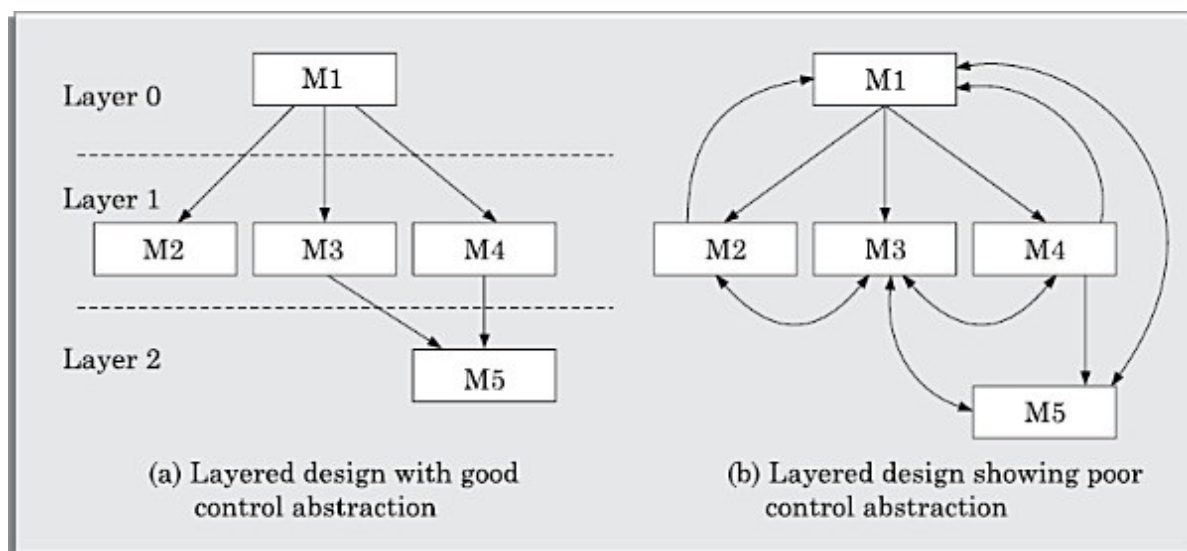


Figure: Examples of good and poor control abstraction.

Functional independence

A module having high cohesion and low coupling is said to be functionally independent of other modules. By the term functional independence, we mean that a cohesive module performs a single task or function. A functionally independent module has minimal interaction with other modules

Need for functional independence

Functional independence is a key to any good design due to the following reasons:

- **Error isolation:** Functional independence reduces error propagation. The reason behind this is that if a module is functionally independent, its degree of interaction with the other modules is less. Therefore, any error existing in a module would not directly effect the other modules.

- **Scope of reuse:** Reuse of a module becomes possible. Because each module does some well-defined and precise function, and the interaction of the module with the other modules is simple and minimal. Therefore, a cohesive module can be easily taken out and reused in a different program.
- **Understandability:** Complexity of the design is reduced, because different modules can be understood in isolation as modules are more or less independent of each other.

Function-oriented design

The following are the salient features of a typical function-oriented design approach:

Top-down decomposition: A system, to start with, is viewed as a black box that provides certain services (also known as high-level functions) to the users of the system. In top-down decomposition, starting at a high-level view of the system, each high-level function is successively refined into more detailed functions.

A system is viewed as something that performs a set of functions. For example, consider a function create-new-library-member which essentially creates the record for a new member, assigns a unique membership number to him, and prints a bill towards his membership charge. This function may consist of the following sub-functions:

- Assign-membership-number
- create-member-record
- print-bill

Centralised system state: The system state can be defined as the values of certain data items that determine the response of the system to a user action or external event. Each of these sub-functions may be split into more detailed subfunctions and so on.

The system state is centralized and shared among different functions, e.g. data such as member-records is available for reference and updation to several functions such as:

- create-new-member
- delete-member
- update-member-record

Object-oriented design

In the object-oriented design approach, the system is viewed as collection of objects (i.e. entities). The state is decentralized among the objects and each object manages its own state information. For example, in a Library Automation Software, each library member may be a separate object with its own data and functions to operate on these data. In fact, the functions defined for one object cannot refer or change data of other objects. Objects have their own internal data which define their state. Similar objects constitute a class. In other words, each object is a member of some class. Classes may inherit features from super class. Conceptually, objects communicate by message passing.

- **Data abstraction:** The principle of data abstraction implies that how data is exactly stored is abstracted away. This means that any entity external to the object (that is, an instance of an ADT) would have no knowledge about how data is exactly stored, organised, and manipulated inside the object.
- **Data structure:** A data structure is constructed from a collection of primitive data items. Just as a civil engineer builds a large civil engineering structure using primitive building materials such as bricks, iron rods, and cement; a programmer can construct a data structure as an organised collection of primitive data items such as integer, floating point numbers, characters, etc.

- **Data type:** A type is a programming language terminology that refers to anything that can be instantiated. For example, int, float, char etc., are the basic data types supported by C programming language. Thus, we can say that ADTs are user defined data types.

Function-oriented vs. object-oriented design approach

The following are some of the important differences between function-oriented and object-oriented design.

- Unlike function-oriented design methods, in OOD, the basic abstraction are not real-world functions such as sort, display, track, etc, but real-world entities such as employee, picture, machine, radar system, etc. For example in OOD, an employee pay-roll software is not developed by designing functions such as update-employee-record, get-employee-address, etc. but by designing objects such as employees, departments, etc. Grady Booch sums up this difference as “identify verbs if you are after procedural design and nouns if you are after object-oriented design”
- In OOD, state information is not represented in a centralized shared memory but is distributed among the objects of the system. For example, while developing an employee pay-roll system, the employee data such as the names of the employees, their code numbers, basic salaries, etc. are usually implemented as global data in a traditional programming system; whereas in an object-oriented system these data are distributed among different employee objects of the system. Objects communicate by message passing. Therefore, one object may discover the state information of another object by interrogating it. Of course, somewhere or other the real-world functions must be implemented. In OOD, the functions are usually associated with specific real-world entities (objects); they directly access only part of the system state information.
- Function-oriented techniques such as SA/SD group functions together if, as a group, they constitute a higher-level function. On the other hand, object-oriented techniques group functions together on the basis of the data they operate on.

Example: Fire-Alarm System

The owner of a large multi-storied building wants to have a computerized fire alarm system for his building. Smoke detectors and fire alarms would be placed in each room of the building. The fire alarm system would monitor the status of these smoke detectors. Whenever a fire condition is reported by any of the smoke detectors, the fire alarm system should determine the location at which the fire condition is reported by any of the smoke detectors, the fire alarm system should determine the location at which the fire condition has occurred and then sound the alarms only in the neighboring locations. The fire alarm system should also flash an alarm message on the computer console. Fire fighting personnel man the console round the clock. After a fire condition has been successfully handled, the fire alarm system should support resetting the alarms by the fire fighting personnel.

Function-Oriented Approach:

```
/* Global data (system state) accessible by various functions*/
BOOL detector_status[MAX_ROOMS];
int detector_locs[MAX_ROOMS];
BOOL alarm_status[MAX_ROOMS];
/* alarm activated when status is set */
```

```

int alarm_locs[MAX_ROOMS];
/* room number where alarm is located*/
int neighbor_alarm[MAX_ROOMS][10];
/* each detector has atmost 10 neighboring locations */

```

The functions which operate on the system state are:

```

interrogate_detectors();
get_detector_location();
determine_neighbor();
ring_alarm();
reset_alarm();
report_fire_location();

```

Object-Oriented Approach:

class detector

attributes:

status, location, neighbors

operations:

create, sense_status, get_location,

find_neighbors

class alarm

attributes:

location, status

operations:

create, ring_alarm, get_location, reset_alarm

- In the object oriented program, an appropriate number of instances of the class detector and alarm should be created. If the function-oriented and the object-oriented programs are examined, it can be seen that in the function-oriented program, the system state is centralized and several functions accessing this central data are defined. In case of the object-oriented program, the state information is distributed among various sensor and alarm objects.
- It is not necessary an object-oriented design be implemented by using an object-oriented language only. However, an object-oriented language such as C++ supports the definition of all the basic mechanisms of class, inheritance, objects, methods, etc. and also support all key object-oriented concepts that we have just discussed. Thus, an object-oriented language facilitates the implementation of an OOD. However, an OOD can as well be implemented using a conventional procedural language – though it may require more effort to implement an OOD using a procedural language as compared to the effort required for implementing the same design using an object-oriented language.
- Even though object-oriented and function-oriented approaches are remarkably different approaches to software design, yet they do not replace each other but complement each other in some sense. For example, usually one applies the top-down function-oriented techniques to design the

internal methods of a class, once the classes are identified. In this case, though outwardly the system appears to have been developed in an object-oriented fashion, but inside each class there may be a small hierarchy of functions designed in a top-down manner.

The SA/SD technique can be used to perform the high-level design of a software. The details of SA/SD technique are discussed further.

OVERVIEW OF SA/SD METHODOLOGY

As the name itself implies, SA/SD methodology involves carrying out two distinct activities:

- Structured analysis (SA)
- Structured design (SD)

The roles of structured analysis (SA) and structured design (SD) have been shown schematically in Figure. Observe the following from the figure:

----> During structured analysis, the SRS document is transformed into a data flow diagram (DFD) model.

-----> During structured design, the DFD model is transformed into a structure chart.

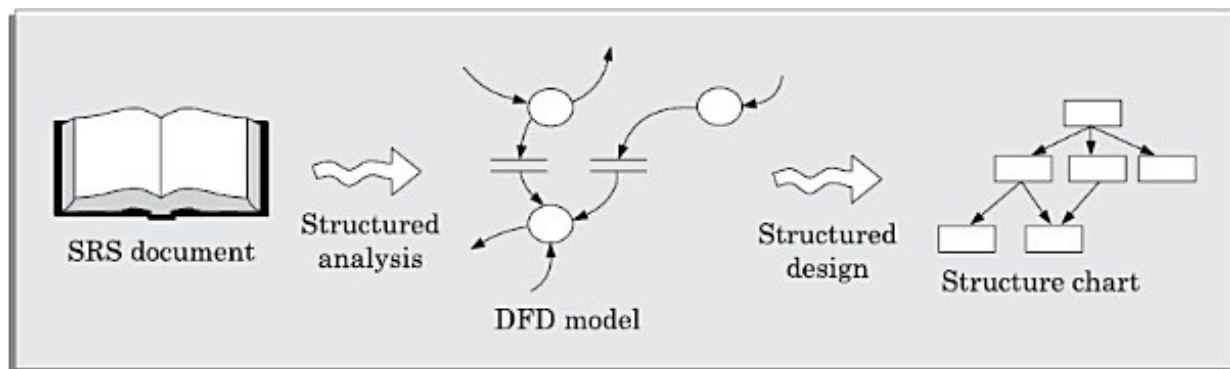


Figure: Structured analysis and structured design methodology.

The structured analysis activity transforms the SRS document into a graphic model called the DFD model. During structured analysis, functional decomposition of the system is achieved. Each function that the system needs to perform is analysed and hierarchically decomposed into more detailed functions. On the other hand, during structured design, all functions identified during structured analysis are mapped to a module structure. This module structure is also called the high-level design or the software architecture for the given problem. This is represented using a structure chart.

Structured Analysis

Structured analysis is used to carry out the top-down decomposition of a set of high-level functions depicted in the problem description and to represent them graphically. During structured analysis, functional decomposition of the system is achieved. That is, each function that the system performs is analyzed and hierarchically decomposed into more detailed functions. Structured analysis technique is based on the following essential underlying principles:

- Top-down decomposition approach.
- Divide and conquer principle. Each function is decomposed independently.
- Graphical representation of the analysis results using Data Flow Diagrams (DFDs).

Data Flow Diagram (DFD)

The DFD (also known as a bubble chart) is a hierarchical graphical model of a system that shows the different processing activities or functions that the system performs and the data interchange among these functions. Each function is considered as a processing station (or process) that consumes some input data and produces some output data. The system is represented in terms of the input data to the system, various processing carried out on these data, and the output data generated by the system. A DFD model uses a very limited number of primitive symbols to represent the functions performed by a system and the data flow among these functions.

Primitive symbols used for constructing DFDs

There are essentially five different types of symbols used for constructing DFDs. These primitive symbols are depicted in Figure. The meaning of these symbols are explained as follows:

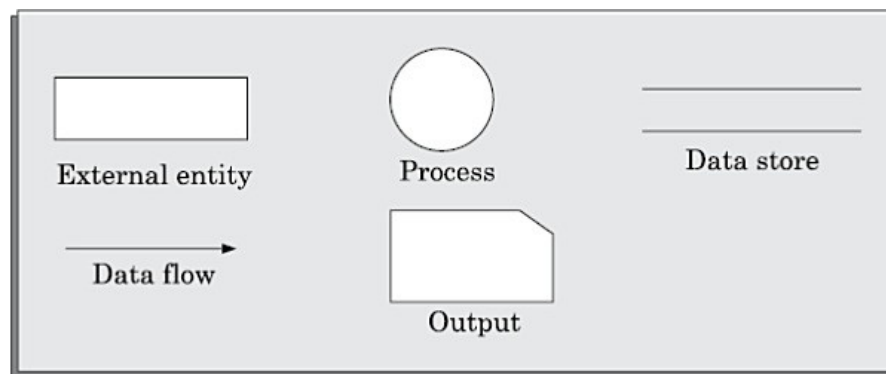


Figure: Symbols used for designing DFDs.

- **Function symbol:** A function is represented using a circle. This symbol is called a process or a bubble. Bubbles are annotated with the names of the corresponding functions.
- **External entity symbol:** An external entity such as a librarian, a library member, etc. is represented by a rectangle. The external entities are essentially those physical entities external to the software system which interact with the system by inputting data to the system or by consuming the data produced by the system. In addition to the human users, the external entity symbols can be used to represent external hardware and software such as another application software that would interact with the software being modelled.
- **Data flow symbol:** A directed arc (or an arrow) is used as a data flow symbol. A data flow symbol represents the data flow occurring between two processes or between an external entity and a process in the direction of the data flow arrow. Data flow symbols are usually annotated with the corresponding data names. For example the DFD. It shows three data flows—the data item number flowing from the process read-number to validate-number, data-item flowing into read-number, and valid-number flowing out of validate-number.
- **Data store symbol:** A data store is represented using two parallel lines. It represents a logical file. That is, a data store symbol can represent either a data structure or a physical file on disk. Each data store is connected to a process by means of a data flow symbol. The direction of the data flow arrow shows whether data is being read from or written into a data store. An arrow flowing in or out of a data store implicitly represents the entire data of the data store and hence arrows connecting to a data store need not be annotated with the name of the corresponding data items. As an example of a data store, number is a data store.

- **Output symbol:** The output symbol is as shown in Figure. The output symbol is used when a hard copy is produced.

Important concepts associated with constructing DFD models

Before we discuss how to construct the DFD model of a system, let us discuss some important concepts associated with DFDs:

Synchronous and asynchronous operations

If two bubbles are directly connected by a data flow arrow, then they are synchronous. This means that they operate at the same speed. An example of such an arrangement is shown here. Here, the `validate-number` bubble can start processing only after the `read-number` bubble has supplied data to it; and the `read-number` bubble has to wait until the `validate-number` bubble has consumed its data.

However, if two bubbles are connected through a data store, then the speed of operation of the bubbles are independent. This statement can be explained using the following reasoning. The data produced by a producer bubble gets stored in the data store. It is therefore possible that the producer bubble stores several pieces of data items, even before the consumer bubble consumes any of them.

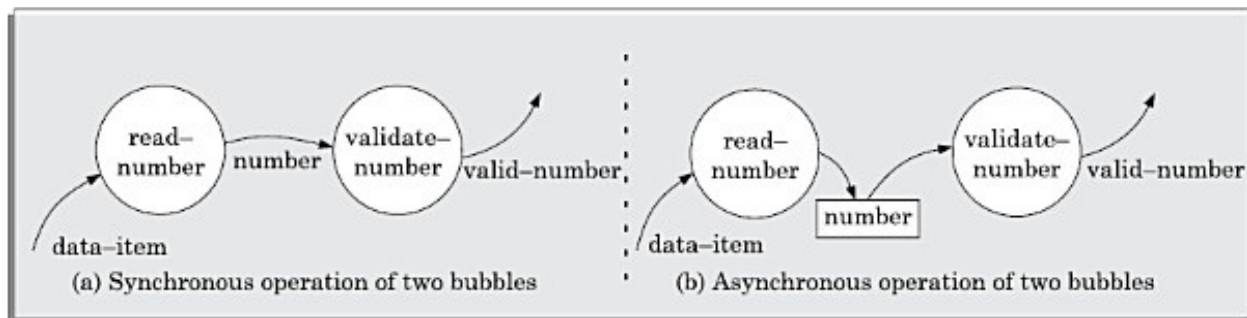


Figure: Synchronous and asynchronous data flow.

Data dictionary

A data dictionary lists all data items appearing in the DFD model of a system. The data items listed include all data flows and the contents of all data stores appearing on the DFDs in the DFD model of a system. A data dictionary lists the purpose of all data items and the definition of all composite data items in terms of their component data items. For example, a data dictionary entry may represent that the data **grossPay** consists of the components `regularPay` and `overtimePay`.

$$\text{grossPay} = \text{regularPay} + \text{overtimePay}$$

For the smallest units of data items, the data dictionary lists their name and their type. Composite data items can be defined in terms of primitive data items using the following data definition operators. The dictionary plays a very important role in any software development process, especially for the following reasons:

- A data dictionary provides a standard terminology for all relevant data for use by the developers working in a project. A consistent vocabulary for data items is very important, since in large projects different developers of the project have a tendency to use different terms to refer to the same data, which unnecessarily causes confusion.
- The data dictionary helps the developers to determine the definition of different data structures in terms of their component elements while implementing the design.

- The data dictionary helps to perform impact analysis. That is, it is possible to determine the effect of some data on various processing activities and vice versa. Such impact analysis is especially useful when one wants to check the impact of changing an input value type, or a bug in some functionality, etc.

Data definition

Composite data items can be defined in terms of primitive data items using the following data definition operators.

- **+**: denotes composition of two data items, e.g. **a+b** represents data **a** and **b**.
- **[, ,]**: represents selection, i.e. any one of the data items listed in the brackets can occur. For example, **[a,b]** represents either **a** occurs or **b** occurs.
- **()**: the contents inside the bracket represent optional data which may or may not appear. e.g. **a+(b)** represents either **a** occurs or **a+b** occurs.
- **{ }**: represents iterative data definition, e.g. **{name}5** represents five **name** data. **{name}*** represents zero or more instances of **name** data.
- **=:** represents equivalence, e.g. **a=b+c** means that **a** represents **b** and **c**.

/* */: Anything appearing within **/*** and ***/** is considered as a comment.

DEVELOPING THE DFD MODEL OF A SYSTEM

A DFD model of a system graphically represents how each input data is transformed to its corresponding output data through a hierarchy of DFDs.

The DFD model of a problem consists of many of DFDs and a single data dictionary.

The DFD model of a system is constructed by using a hierarchy of DFDs. The top level DFD is called the level 0 DFD or the context diagram. This is the most abstract (simplest) representation of the system (highest level). It is the easiest to draw and understand. At each successive lower level DFDs, more and more details are gradually introduced. To develop a higher-level DFD model, processes are decomposed into their subprocesses and the data flow among these subprocesses are identified.

To develop the data flow model of a system, first the most abstract representation (highest level) of the problem is to be worked out. Subsequently, the lower level DFDs are developed. Level 0 and Level 1 consist of only one DFD each. Level 2 may contain up to 7 separate DFDs, and level 3 up to 49 DFDs, and so on. However, there is only a single data dictionary for the entire DFD model. All the data names appearing in all DFDs are populated in the data dictionary and the data dictionary contains the definitions of all the data items.

Context diagram:

The context diagram is the most abstract data flow representation of a system. It represents the entire system as a single bubble. This bubble is labeled according to the main function of the system. The various external entities with which the system interacts and the data flow occurring between the system and the external entities are also represented. The data input to the system and the data output from the system are represented as incoming and outgoing arrows. These data flow arrows should be annotated with the corresponding data names. The name 'context diagram' is well justified because it represents the context in which the system is to exist, i.e. the external entities who would interact with the system and the specific data items they would be supplying the system and the data items they would be receiving from the system. The context diagram is also called as the level 0 DFD.

To develop the context diagram of the system, it is required to analyze the SRS document to identify the different types of users who would be using the system and the kinds of data they would be inputting to the system and the data they would be receiving from the system. Here, the term “users of the system” also includes the external systems which supply data to or receive data from the system.

The bubble in the context diagram is annotated with the name of the software system being developed (usually a noun). This is in contrast with the bubbles in all other levels which are annotated with verbs. This is expected since the purpose of the context diagram is to capture the context of the system rather than its functionality.

We can now describe how to go about developing the DFD model of a system more systematically.

Construction of context diagram: Examine the SRS document to determine:

- Different high-level functions that the system needs to perform.
- Data input to every high-level function.
- Data output from every high-level function.
- Interactions (data flow) among the identified high-level functions. Represent these aspects of the high-level functions in a diagrammatic form. This would form the top-level data flow diagram (DFD), usually called the DFD 0.

Construction of level 1 diagram: Examine the high-level functions described in the SRS document. If there are three to seven high-level requirements in the SRS document, then represent each of the high-level function in the form of a bubble. If there are more than seven bubbles, then some of them have to be combined. If there are less than three bubbles, then some of these have to be split.

Construction of lower-level diagrams: Decompose each high-level function into its constituent subfunctions through the following set of activities:

- Identify the different subfunctions of the high-level function.
- Identify the data input to each of these subfunctions.
- Identify the data output from each of these subfunctions.
- Identify the interactions (data flow) among these subfunctions. Represent these aspects in a diagrammatic form using a DFD.

Decomposition:-

Each bubble in the DFD represents a function performed by the system. The bubbles are decomposed into sub-functions at the successive levels of the DFD. Decomposition of a bubble is also known as factoring or exploding a bubble. Each bubble at any level of DFD is usually decomposed to anything between 3 to 7 bubbles. Too few bubbles at any level make that level superfluous. For example, if a bubble is decomposed to just one bubble or two bubbles, then this decomposition becomes redundant. Also, too many bubbles, i.e. more than 7 bubbles at any level of a DFD makes the DFD model hard to understand. Decomposition of a bubble should be carried on until a level is reached at which the function of the bubble can be described using a simple algorithm.

Numbering of Bubbles:-

It is necessary to number the different bubbles occurring in the DFD. These numbers help in uniquely identifying any bubble in the DFD by its bubble number. The bubble at the context level is usually assigned the number 0 to indicate that it is the 0 level DFD. Bubbles at level 1 are numbered, 0.1, 0.2, 0.3, etc, etc. When a bubble numbered x is decomposed, its children bubble are numbered x.1, x.2, x.3, etc. In this numbering scheme, by looking at the number of a bubble we can unambiguously determine its level, its ancestors, and its successors.

Balancing a DFD

The data that flow into or out of a bubble must match the data flow at the next level of DFD. This is known as balancing a DFD. The concept of balancing a DFD has been illustrated in fig. In the level 1 of the DFD, data items d1 and d3 flow out of the bubble 0.1 and the data item d2 flows into the bubble 0.1. In the next level, bubble 0.1 is decomposed. The decomposition is balanced, as d1 and d3 flow out of the level 2 diagram and d2 flows in.

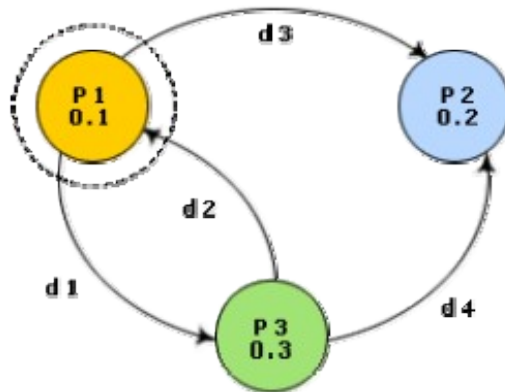


Fig:Level 1 DFD

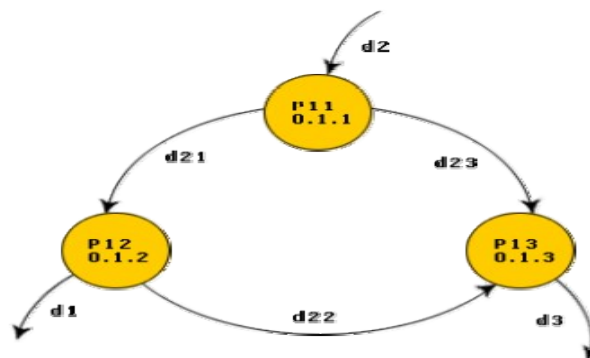


Fig:Level 2 DFD

some important shortcomings of the DFD model.

DFD models suffer from several shortcomings. The important shortcomings of the DFD models are the following:

- DFDs leave ample scope to be imprecise. In the DFD model, the function performed by a bubble is judged from its label. However, a short label may not capture the entire functionality of a bubble. For example, a bubble named find-book-position has only intuitive meaning and does not specify several things, e.g. what happens when some input information are missing or are incorrect. Further, the find-book-position bubble may not convey anything regarding what happens when the required book is missing.
- Control aspects are not defined by a DFD. For instance, the order in which inputs are consumed and outputs are produced by a bubble is not specified. A DFD model does not specify the order in which the different bubbles are executed. Representation of such aspects is very important for modeling real-time systems.
- The method of carrying out decomposition to arrive at the successive levels and the ultimate level to which decomposition is carried out are highly subjective and depend on the choice and judgment of the analyst. Due to this reason, even for the same problem, several alternative DFD representations are possible. Further, many times it is not possible to say which DFD representation is superior or preferable to another one.

- The data flow diagramming technique does not provide any specific guidance as to how exactly to decompose a given function into its sub- functions and we have to use subjective judgment to carry out decomposition.

Example.1(Tic-Tac-Toe Computer Game) Tic-tac-toe is a computer game in which a human player and the computer make alternate moves on a 3×3 square. A move consists of marking a previously unmarked square. The player who is first to place three consecutive marks along a straight line (i.e., along a row, column, or diagonal) on the square wins. As soon as either of the human player or the computer wins, a message congratulating the winner should be displayed. If neither player manages to get three consecutive marks along a straight line, and all the squares on the board are filled up, then the game is drawn. The computer always tries to win a game.

The context diagram and the level 1 DFD are shown in Figure.

Data dictionary for the DFD model of Example

move: integer /* number between 1 to 9 */

display: game+result

game: board

board: {integer}9

result: ["computer won", "human won", "drawn"]

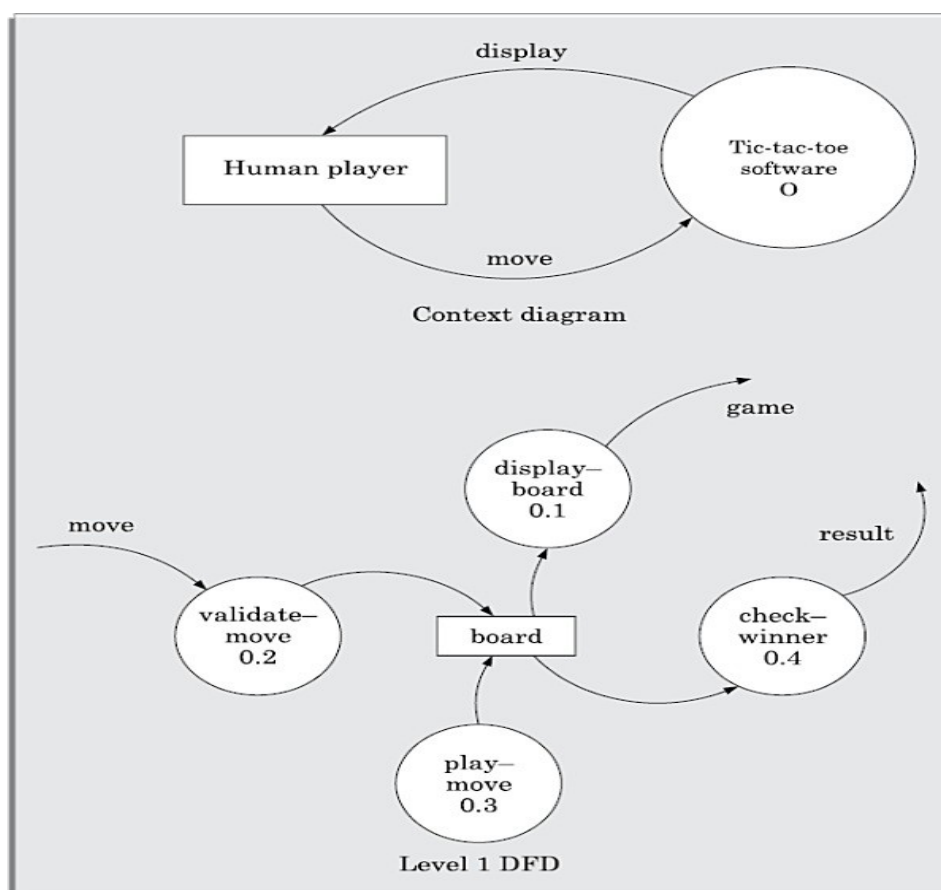


Figure: Context diagram and level 1 DFDs for Example 1

Example 2(Supermarket Prize Scheme) A super market needs to develop a software that would help it to automate a scheme that it plans to introduce to encourage regular customers. In this scheme, a customer would have first register by supplying his/her residence address, telephone number, and the driving license number. Each customer who registers for this scheme is assigned a unique customer number (CN) by the computer. A customer can present his CN to the check out staff when he makes any purchase. In this case, the value of his purchase is credited against his CN. At the end of each year, the supermarket intends to award surprise gifts to 10 customers who make the highest total purchase over the year. Also, it intends to award a 22 caret gold coin to every customer whose purchase exceeded Rs. 10,000. The entries against the CN are reset on the last day of every year after the prize winners' lists are generated.

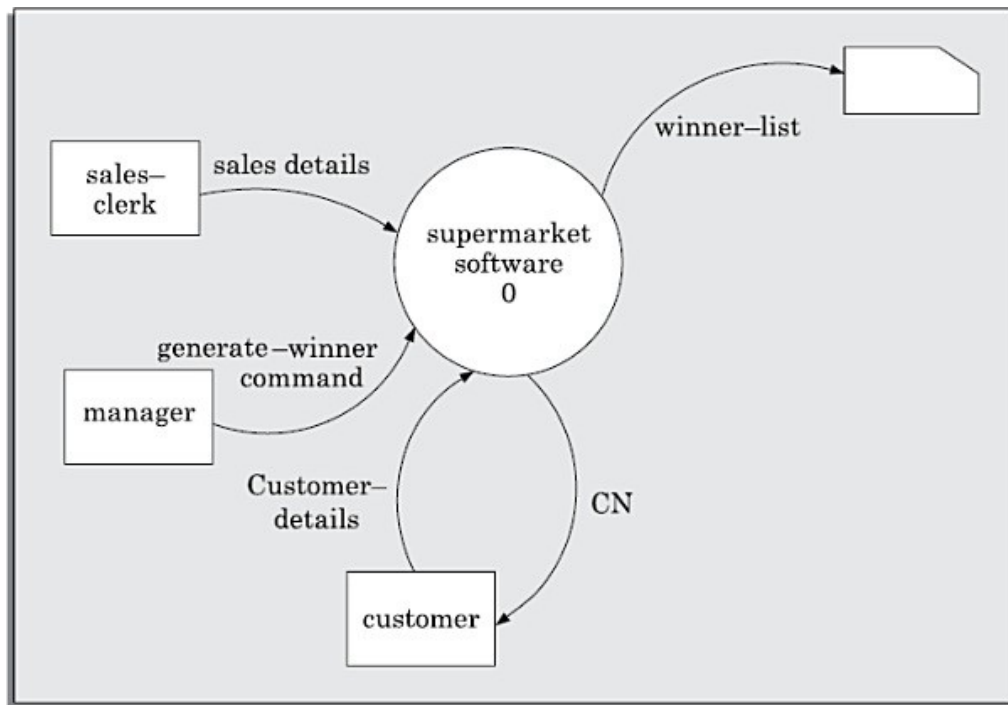


Figure: Context diagram for Example 2

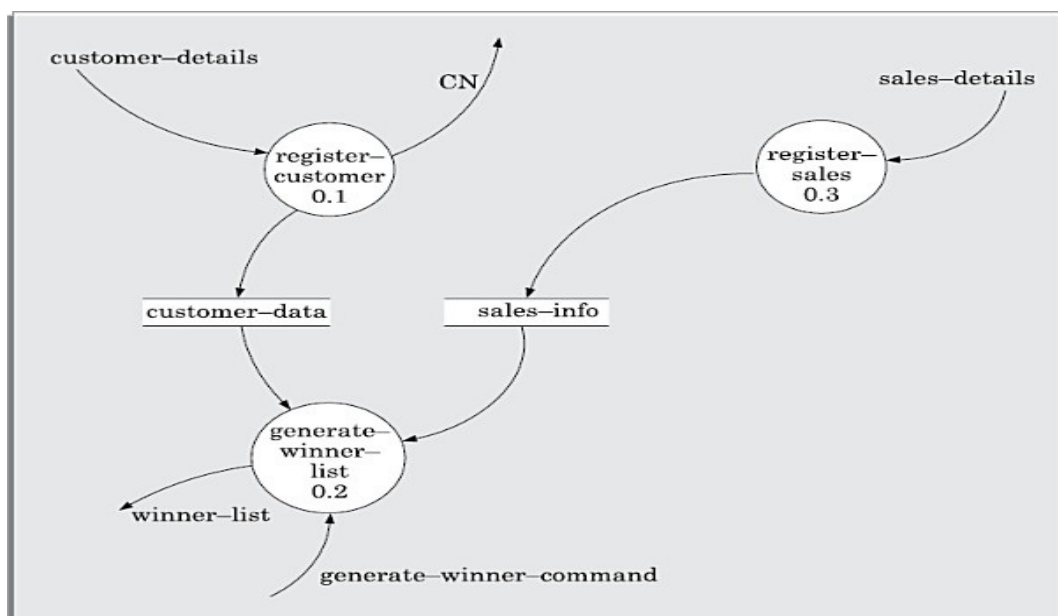


Figure: Level 1 diagram for Example 2

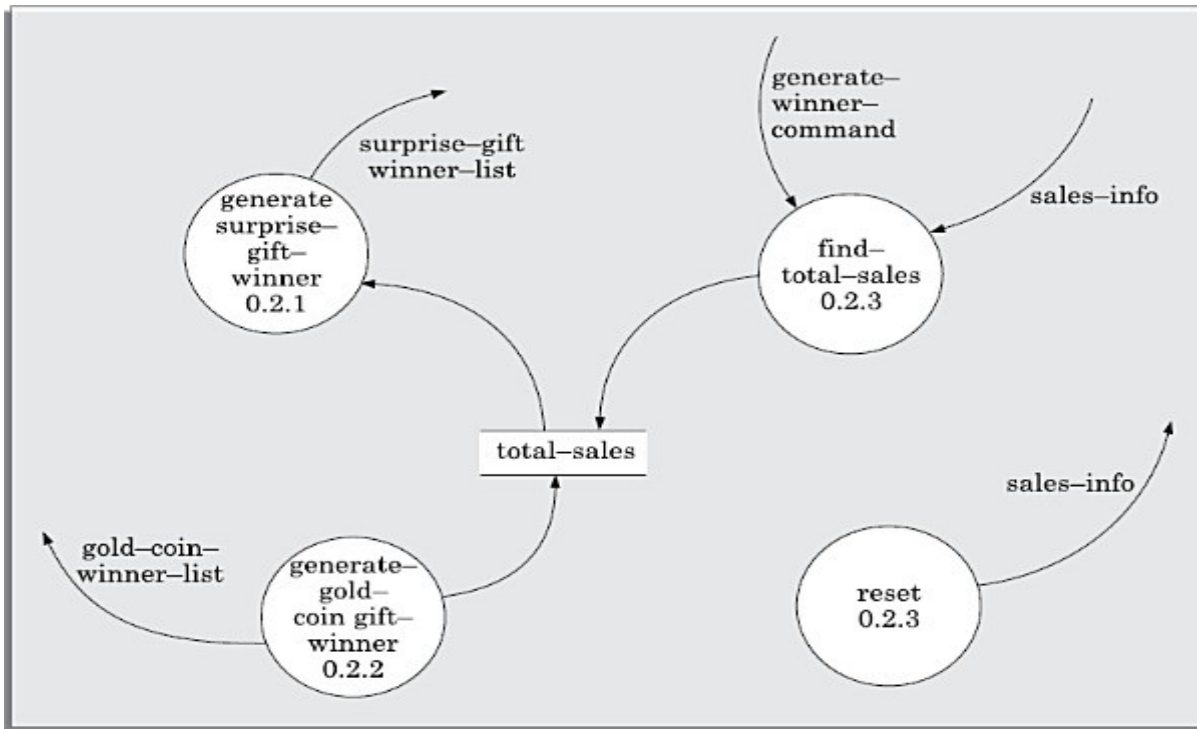


Figure: Level 2 diagram for Example 2

Data dictionary for the DFD model of Example 2

address: name+house#+street#+city+pin

sales-details: {item+amount}* + CN

CN: integer

customer-data: {address+CN}*

sales-info: {sales-details}*

winner-list: surprise-gift-winner-list + gold-coin-winner-list

surprise-gift-winner-list: {address+CN}*

gold-coin-winner-list: {address+CN}*

gen-winner-command: command

total-sales: {CN+integer}*

Extending DFD Technique to make it Applicable to Real-time Systems

In a real-time system, some of the high-level functions are associated with deadlines. Therefore, a function must not only produce correct results but also should produce them by some prespecified time. For real-time systems, execution time is an important consideration for arriving at a correct design. Therefore, explicit representation of control and event flow aspects are essential. One of the widely accepted techniques for extending the DFD technique to real-time system analysis is the Ward and Mellor technique [1985]. In the Ward and Mellor notation, a type of process that handles only control flows is introduced. These processes representing control processing are denoted using dashed bubbles. Control flows are shown using dashed lines/arrows.

To be able to separate the data processing and the control processing aspects, a control flow diagram (CFD) is defined. This reduces the complexity of the diagrams. In order to link the data

processing and control processing diagrams, a notational reference (solid bar) to a control specification is used. The CSPEC describes the following:

- The effect of an external event or control signal.
- The processes that are invoked as a consequence of an event.

Control specifications represent the behavior of the system in two different ways:

- It contains a state transition diagram (STD). The STD is a sequential specification of behaviour.
- It contains a program activation table (PAT). The PAT is a combinatorial specification of behaviour. PAT represents invocation sequence of bubbles in a DFD.

Structured Design

The aim of structured design is to transform the results of the structured analysis (i.e. a DFD representation) into a structure chart. Structured design provides two strategies to guide transformation of a DFD into a structure chart.

- Transform analysis
- Transaction analysis

Normally, one starts with the level 1 DFD, transforms it into module representation using either the transform or the transaction analysis and then proceeds towards the lower-level DFDs. At each level of transformation, it is important to first determine whether the transform or the transaction analysis is applicable to a particular DFD. These are discussed in the subsequent sub-sections.

Structure Chart

A structure chart represents the software architecture, i.e. the various modules making up the system, the dependency (which module calls which other modules), and the parameters that are passed among the different modules. Hence, the structure chart representation can be easily implemented using some programming language. Since the main focus in a structure chart representation is on the module structure of the software and the interactions among different modules, the procedural aspects (e.g. how a particular functionality is achieved) are not represented.

The basic building blocks using which structure charts are designed are as following:

Rectangular boxes: A rectangular box represents a module. Usually, every rectangular box is annotated with the name of the module it represents.

Module invocation arrows: An arrow connecting two modules implies that during program execution control is passed from one module to the other in the direction of the connecting arrow. However, just by looking at the structure chart, we cannot say whether a module calls another module just once or many times. Also, just by looking at the structure chart, we cannot tell the order in which the different modules are invoked.

Data flow arrows: These are small arrows appearing alongside the module invocation arrows. The data flow arrows are annotated with the corresponding data name. Data flow arrows represent the fact that the named data passes from one module to the other in the direction of the arrow.

Library modules: A library module is usually represented by a rectangle with double edges. Libraries comprise the frequently called modules. Usually, when a module is invoked by many other modules, it is made into a library module.

Selection: The diamond symbol represents the fact that one module of several modules connected with the diamond symbol is invoked depending on the outcome of the condition attached with the diamond symbol.

Repetition: A loop around the control flow arrows denotes that the respective modules are invoked repeatedly.

Structure Chart vs. Flow Chart

We are all familiar with the flow chart representation of a program. Flow chart is a convenient technique to represent the flow of control in a program. A structure chart differs from a flow chart in three principal ways:

- It is usually difficult to identify the different modules of the software from its flow chart representation.
- Data interchange among different modules is not represented in a flow chart.
- Sequential ordering of tasks inherent in a flow chart is suppressed in a structure chart.

Transform Analysis

Transform analysis identifies the primary functional components (modules) and the high level inputs and outputs for these components. The first step in transform analysis is to divide the DFD into 3 types of parts:

- Input
- Logical processing
- Output
- The input portion of the DFD includes processes that transform input data from physical (e.g. character from terminal) to logical forms (e.g. internal tables, lists, etc.). Each input portion is called an afferent branch.
- The output portion of a DFD transforms output data from logical to physical form. Each output portion is called an efferent branch. The remaining portion of a DFD is called the central transform.
- In the next step of transform analysis, the structure chart is derived by drawing one functional component for the central transform, and the afferent and efferent branches. These are drawn below a root module, which would invoke these modules.

Identifying the highest level input and output transforms requires experience and skill. One possible approach is to trace the inputs until a bubble is found whose output cannot be deduced from its inputs alone. Processes which validate input or add information to them are not central transforms. Processes which sort input or filter data from it are. The first level structure chart is produced by representing each input and output unit as boxes and each central transform as a single box.

In the third step of transform analysis, the structure chart is refined by adding sub-functions required by each of the high-level functional components. Many levels of functional components may be added. This process of breaking functional components into subcomponents is called factoring. Factoring includes adding read and write modules, error-handling modules, initialization and termination process, identifying customer modules, etc. The factoring process is continued until all bubbles in the DFD are represented in the structure chart.

Example: Draw the structure chart for the RMS software

By observing the level 1 DFD, we can identify validate-input as the afferent branch and write-output as the efferent branch. The remaining (i.e., compute-rms) as the central transform. By applying the step 2 and step 3 of transform analysis, we get the structure chart shown in Figure.

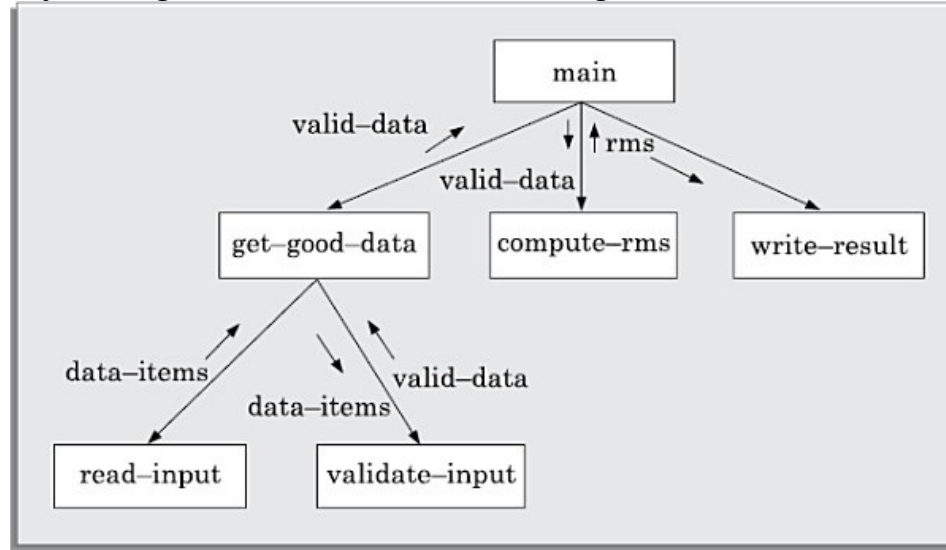


Figure: Structure chart for Example

Transaction Analysis

A transaction allows the user to perform some meaningful piece of work. Transaction analysis is useful while designing transaction processing programs. In a transaction-driven system, one of several possible paths through the DFD is traversed depending upon the input data item. This is in contrast to a transform centered system which is characterized by similar processing steps for each data item. Each different way in which input data is handled is a transaction. A simple way to identify a transaction is to check the input data. The number of bubbles on which the input data to the DFD are incident defines the number of transactions. However, some transaction may not require any input data. These transactions can be identified from the experience of solving a large number of examples.

For each identified transaction, trace the input data to the output. All the traversed bubbles belong to the transaction. These bubbles should be mapped to the same module on the structure chart. In the structure chart, draw a root module and below this module draw each identified transaction a module. Every transaction carries a tag, which identifies its type. Transaction analysis uses this tag to divide the system into transaction modules and a transaction-center module.

Example: Draw the structure chart for the personal library software of Example

The structure chart for the personal library software is shown in Figure

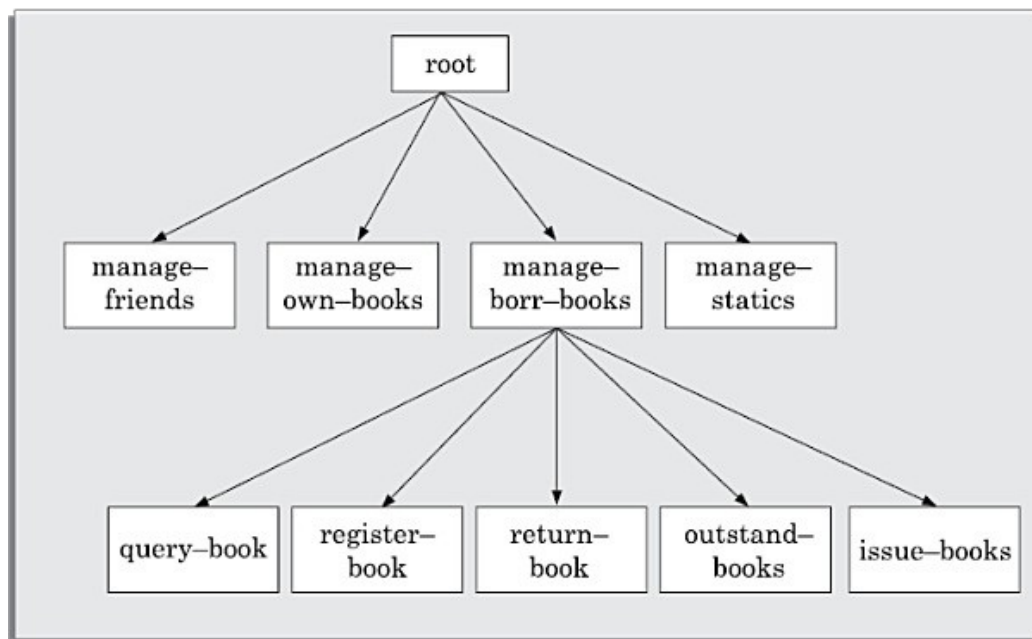


Figure: Structure chart for Example

DETAILED DESIGN

During detailed design the pseudo code description of the processing and the different data structures are designed for the different modules of the structure chart. These are usually described in the form of module specifications (MSPEC). MSPEC is usually written using structured English. The MSPEC for the non-leaf modules describe the different conditions under which the responsibilities are delegated to the lower-level modules. The MSPEC for the leaf-level modules should describe in algorithmic form how the primitive processing steps are carried out. To develop the MSPEC of a module, it is usually necessary to refer to the DFD model and the SRS document to determine the functionality of the module.

DESIGN REVIEW

After a design is complete, the design is required to be reviewed. The review team usually consists of members with design, implementation, testing, and maintenance perspectives, who may or may not be the members of the development team. Normally, members of the team who would code the design, and test the code, the analysts, and the maintainers attend the review meeting. The review team checks the design documents especially for the following aspects:

Traceability: Whether each bubble of the DFD can be traced to some module in the structure chart and vice versa. They check whether each functional requirement in the SRS document can be traced to some bubble in the DFD model and vice versa.

Correctness: Whether all the algorithms and data structures of the detailed design are correct.

Maintainability: Whether the design can be easily maintained in future.

Implementation: Whether the design can be easily and efficiently be implemented.

Model

A model is constructed by focusing only on a few aspects of the problem and ignoring the rest. The model of a problem is called an analysis model. On the other hand, the model of the solution (code) is called the design model. The design model is usually obtained by carrying out iterative refinements to the analysis model using a design methodology.

A model captures aspects important for some application while omitting (or abstracting) the rest. A model in the context of software development can be graphical, textual, mathematical, or program code-based. Models are very useful in documenting the design and analysis results. Models also facilitate the analysis and design procedures themselves. Graphical models are very popular because they are easy to understand and construct. UML is primarily a graphical modeling tool. However, it often requires text explanations to accompany the graphical models.

Need for a model

An important reason behind constructing a model is that it helps manage complexity. Once models of a system have been constructed, these can be used for a variety of purposes during software development, including the following:

- Analysis
- Specification
- Code generation
- Design
- Visualize and understand the problem and the working of a system
- Testing, etc.

In all these applications, the UML models can not only be used to document the results but also to arrive at the results themselves. Since a model can be used for a variety of purposes, it is reasonable to expect that the model would vary depending on the purpose for which it is being constructed. For example, a model developed for initial analysis and specification should be very different from the one used for design. A model that is being used for analysis and specification would not show any of the design decisions that would be made later on during the design stage. On the other hand, a model used for design purposes should capture all the design decisions. Therefore, it is a good idea to explicitly mention the purpose for which a model has been developed, along with the model.

Unified Modeling Language (UML)

UML, as the name implies, is a modeling language. It may be used to visualize, specify, construct, and document the artifacts of a software system. It provides a set of notations (e.g. rectangles, lines, ellipses, etc.) to create a visual model of the system. Like any other language, UML has its own syntax (symbols and sentence formation rules) and semantics (meanings of symbols and sentences). Also, we should clearly understand that UML is not a system design or development methodology, but can be used to document object-oriented and analysis results obtained using some methodology.

Origin of UML

In the late 1980s and early 1990s, there was a proliferation of object-oriented design techniques and notations. Different software development houses were using different notations to document their object-oriented designs. These diverse notations used to give rise to a lot of confusion.

UML was developed to standardize the large number of object-oriented modeling notations that existed and were used extensively in the early 1990s. The principles ones in use were:

- Object Management Technology [Rumbaugh 1991]
- Booch's methodology [Booch 1991]
- Object-Oriented Software Engineering [Jacobson 1992]
- Odell's methodology [Odell 1992]
- Shaler and Mellor methodology [Shaler 1992]

It is needless to say that UML has borrowed many concepts from these modeling techniques. Especially, concepts from the first three methodologies have been heavily drawn upon. UML was adopted by Object Management Group (OMG) as a *de facto* standard in 1997. OMG is an association of industries which tries to facilitate early formation of standards.

We shall see that UML contains an extensive set of notations and suggests construction of many types of diagrams. It has successfully been used to model both large and small problems. The elegance of UML, its adoption by OMG, and a strong industry backing have helped UML find widespread acceptance. UML is now being used in a large number of software development projects worldwide.

UML diagrams

UML can be used to construct nine different types of diagrams to capture five different views of a system. Just as a building can be modeled from several views (or perspectives) such as ventilation perspective, electrical perspective, lighting perspective, heating perspective, etc.; the different UML diagrams provide different perspectives of the software system to be developed and facilitate a comprehensive understanding of the system. Such models can be refined to get the actual implementation of the system.

The UML diagrams can capture the following five views of a system:

- User's view
- Structural view
- Behavioral view
- Implementation view
- Environmental view

It shows the different views that the UML diagrams can document. Observe that the users' view is shown as the central view. This is because based on the users' view, all other views are developed and all views need to conform to the user's view. Most of the object oriented analysis and design methodologies, including the one we are going to discuss in Chapter 8 require us to iterate among the different views several times to arrive at the final design. We first provide a brief overview of the different views of a system which can be documented using UML. In the subsequent sections, the diagrams used to realize the important views are discussed.

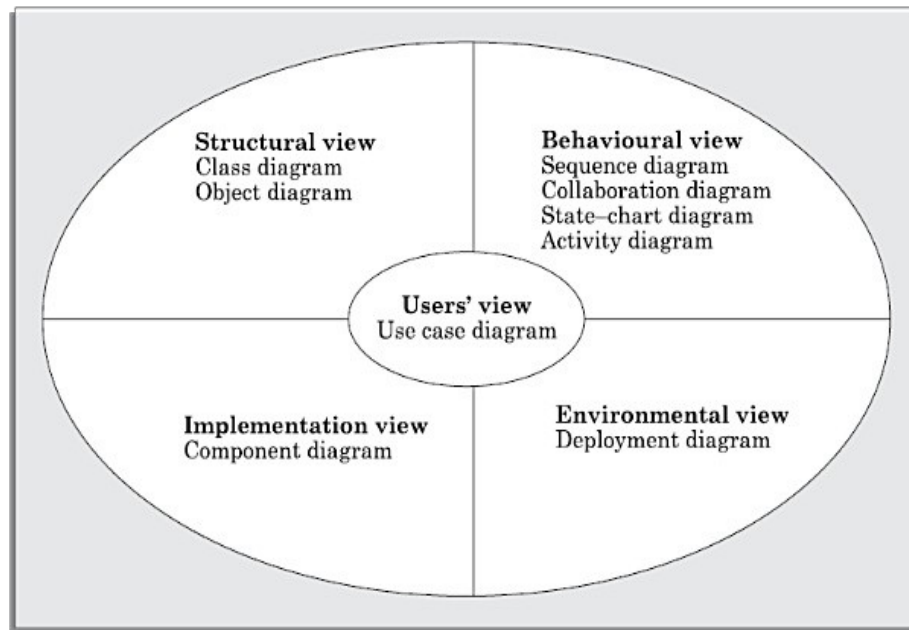


Figure: Different types of diagrams and views supported in UML.

Users' view: This view defines the functionalities (facilities) made available by the system to its users. The users' view captures the external users' view of the system in terms of the functionalities offered by the system. The users view is a black-box view of the system where the internal structure, the dynamic behavior of different system components, the implementation etc. are not visible. The users' view is very different from all other views in the sense that it is a functional model compared to the object model of all other views. The users' view can be considered as the central view and all other views are expected to conform to this view. This thinking is in fact the crux of any user centric development style.

Structural view: The structural view defines the kinds of objects (classes) important to the understanding of the working of a system and to its implementation. It also captures the relationships among the classes (objects). The structural model is also called the static model, since the structure of a system does not change with time.

Behavioral view: The behavioral view captures how objects interact with each other to realize the system behavior. The system behavior captures the time-dependent (dynamic) behavior of the system.

Implementation view: This view captures the important components of the system and their dependencies.

Environmental view: This view models how the different components are implemented on different pieces of hardware.

Use Case Model

The use case model for any system consists of a set of “use cases”. Intuitively, use cases represent the different ways in which a system can be used by the users. A simple way to find all the use cases of a system is to ask the question: “What the users can do using the system?” Thus for the Library Information System (LIS), the use cases could be:

- issue-book
- query-book

- return-book
- create-member
- add-book, etc

Use cases correspond to the high-level functional requirements. The use cases partition the system behavior into transactions, such that each transaction performs some useful action from the user's point of view. To complete each transaction may involve either a single message or multiple message exchanges between the user and the system to complete.

Purpose of use cases

The purpose of a use case is to define a piece of coherent behavior without revealing the internal structure of the system. The use cases do not mention any specific algorithm to be used or the internal data representation, internal structure of the software, etc. A use case typically represents a sequence of interactions between the user and the system. These interactions consist of one mainline sequence. The mainline sequence represents the normal interaction between a user and the system. The mainline sequence is the most occurring sequence of interaction. For example, the mainline sequence of the withdraw cash use case supported by a bank ATM drawn, complete the transaction, and get the amount. Several variations to the main line sequence may also exist. Typically, a variation from the mainline sequence occurs when some specific conditions hold.

For the bank ATM example, variations or alternate scenarios may occur, if the password is invalid or the amount to be withdrawn exceeds the amount balance. The variations are also called alternative paths. A use case can be viewed as a set of related scenarios tied together by a common goal. The mainline sequence and each of the variations are called scenarios or instances of the use case. Each scenario is a single path of user events and system activity through the use case.

Representation of use cases

Use cases can be represented by drawing a use case diagram and writing an accompanying text elaborating the drawing. In the use case diagram, each use case is represented by an ellipse with the name of the use case written inside the ellipse. All the ellipses (i.e. use cases) of a system are enclosed within a rectangle which represents the system boundary. The name of the system being modeled (such as Library Information System) appears inside the rectangle.

The different users of the system are represented by using the stick person icon. Each stick person icon is normally referred to as an actor. An actor is a role played by a user with respect to the system use. It is possible that the same user may play the role of multiple actors. Each actor can participate in one or more use cases. The line connecting the actor and the use case is called the communication relationship. It indicates that the actor makes use of the functionality provided by the use case. Both the human users and the external systems can be represented by stick person icons. When a stick person icon represents an external system, it is annotated by the stereotype <<external system>>.

Example: The use case model for the Tic-tac-toe game software is shown. This software has only one use case, namely, “play move”. Note that we did not name the use case “get-user-move”, as “get-user-move” would be inappropriate because this would represent the developer’s perspective of the use case. The use cases should be named from the users’ perspective.

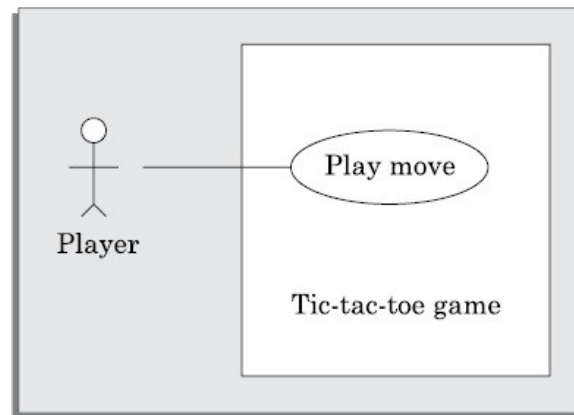


Figure: Use case model for Example

Text description

Each ellipse on the use case diagram should be accompanied by a text description. The text description should define the details of the interaction between the user and the computer and other aspects of the use case. It should include all the behavior associated with the use case in terms of the mainline sequence, different variations to the normal behavior, the system responses associated with the use case, the exceptional conditions that may occur in the behavior, etc. The behavior description is often written in a conversational style describing the interactions between the actor and the system. The text description may be informal, but some structuring is recommended. The following are some of the information which may be included in a use case text description in addition to the mainline sequence, and the alternative scenarios.

Contact persons: This section lists the personnel of the client organization with whom the use case was discussed, date and time of the meeting, etc.

Actors: In addition to identifying the actors, some information about actors using this use case which may help the implementation of the use case may be recorded.

Pre-condition: The preconditions would describe the state of the system before the use case execution starts.

Post-condition: This captures the state of the system after the use case has successfully completed.

Non-functional requirements: This could contain the important constraints for the design and implementation, such as platform and environment conditions, qualitative response time requirements, etc.

Exceptions, error situations: This contains only the domain-related errors such as lack of user’s access rights, invalid entry in the input fields, etc. Obviously, errors that are not domain related, such as software errors, need not be discussed here.

Sample dialogs: These serve as examples illustrating the use case.

Specific user interface requirements: These contain specific requirements for the user interface of the use case. For example, it may contain forms to be used, screen shots, interaction style, etc.

Document references: This part contains references to specific domain- related documents which may be useful to understand the system operation.

Example:The use case diagram of the Super market prize scheme described in example

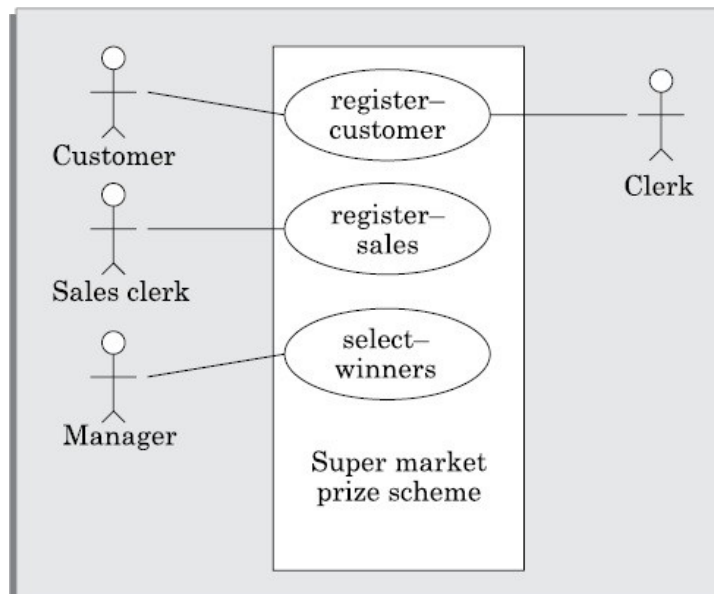


Figure : Use case model for Example

Text description

U1: register-customer: Using this use case, the customer can register himself by providing the necessary details.

Scenario 1: Mainline sequence

- 1.Customer: select register customer option
- 2 . System: display prompt to enter name, address, and telephone number.
3. Customer: enter the necessary values
- 4 : System: display the generated id and the message that the customer has successfully been registered.

Scenario 2: At step 4 of mainline sequence

- 4 : System: displays the message that the customer has already registered.

Scenario 3: At step 4 of mainline sequence

- 4 : System: displays message that some input information have not been entered. The system displays a prompt to enter the missing values.

U2: register-sales: Using this use case, the clerk can register the details of the purchase made by a customer.

Scenario 1: Mainline sequence

1. Clerk: selects the register sales option.
2. System: displays prompt to enter the purchase details and the id of the customer.
3. Clerk: enters the required details.
4. System: displays a message of having successfully registered the sale.

U3: select-winners. Using this use case, the manager can generate the winner list.

Scenario 2: Mainline sequence

1. Manager: selects the select-winner option.
2. System: displays the gold coin and the surprise gift winner list.

Utility of use case diagrams: From use case diagram, it is obvious that the utility of the use cases are represented by ellipses. They along with the accompanying text description serve as a type of requirements specification of the system and form the core model to which all other models must conform. But, what about the actors (stick person icons)? One possible use of identifying the different types of users (actors) is in identifying and implementing a security mechanism through a login system, so that each actor can involve only those functionalities to which he is entitled to. Another possible use is in preparing the documentation (e.g. users' manual) targeted at each category of user. Further, actors help in identifying the use cases and understanding the exact functioning of the system.

Factoring of use cases

It is often desirable to factor use cases into component use cases. Actually, factoring of use cases are required under two situations.

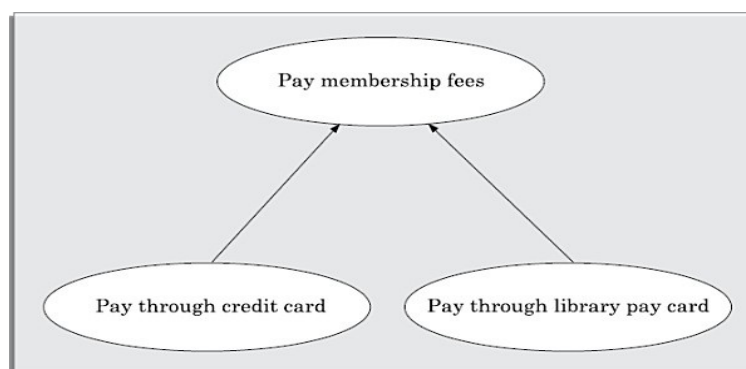
- First, complex use cases need to be factored into simpler use cases. This would not only make the behavior associated with the use case much more comprehensible, but also make the corresponding interaction diagrams more tractable. Without decomposition, the interaction diagrams for complex use cases may become too large to be accommodated on a single sized (A4) paper.
- Secondly, use cases need to be factored whenever there is common behavior across different use cases. Factoring would make it possible to define such behavior only once and reuse it whenever required. It is desirable to factor out common usage such as error handling from a set of use cases. This makes analysis of the class design much simpler and elegant. However, a word of caution here. Factoring of use cases should not be done except for achieving the above two objectives. From the design point of view, it is not advantageous to break up a use case into many smaller parts just for the shake of it.

UML offers three mechanisms for factoring of use cases as follows:

Generalization

Use case generalization can be used when one use case that is similar to another, but does something slightly differently or something more. Generalization works the same way with use cases as it does with classes. The child use case inherits the behavior and meaning of the parent use case. The notation is the same too. It is important to remember that the base and the derived use cases are separate use cases and should have separate text description

Figure: Representation of use case generalisation.



Includes

The includes relationship in the older versions of UML (prior to UML) was known as the uses relationship. The includes relationship involves one use case including the behavior of another use case in its sequence of events and actions. The includes relationship occurs when a chunk of behavior that is similar across a number of use cases. The factoring of such behavior will help in not repeating the specification and implementation across different use cases. Thus, the includes relationship explores the issue of reuse by factoring out the commonality across use cases. It can also be gainfully employed to decompose a large and complex use cases into more manageable parts. As shown in fig, the includes relationship is represented using a predefined stereotype `<<include>>`. In the includes relationship, a base use case compulsorily and automatically includes the behavior of the common use cases. As shown in example fig. issue-book and renew-book both include check-reservation use case. The base use case may include several use cases. In such cases, it may interleave their associated common use cases together. The common use case becomes a separate use case and the independent text description should be provided for it.



Figure: Representation of use case inclusion.

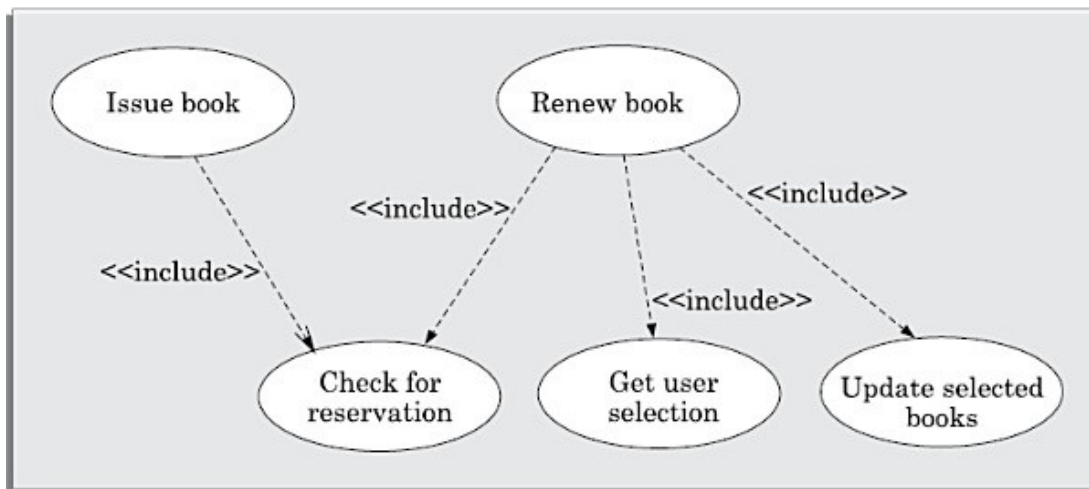


Figure: Example of use case inclusion

Extends

The main idea behind the extends relationship among the use cases is that it allows you to show optional system behavior. An optional system behavior is extended only under certain conditions. This relationship among use cases is also predefined as a stereotype as shown in fig. The extends relationship is similar to generalization. But unlike generalization, the extending use case can add additional behavior only at an extension point only when certain conditions are satisfied. The extension points are points within the use case where variation to the mainline (normal) action sequence may occur. The extends relationship is normally used to capture alternate paths or scenarios.

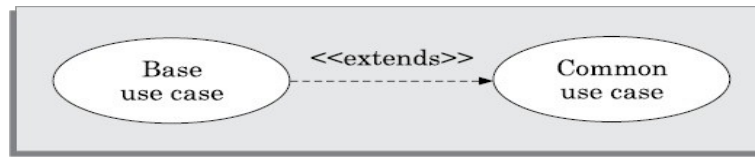


Figure: Example of use case extension.

The extends relationship is similar to generalisation. But unlike generalisation, the extending use case can add additional behaviour only at an extension point only when certain conditions are satisfied. The extension points are points within the use case where variation to the mainline (normal) action sequence may occur. The extends relationship is normally used to capture alternate paths or scenarios.

Organization of use cases

When the use cases are factored, they are organized hierarchically. The high-level use cases are refined into a set of smaller and more refined use cases as shown in fig. Top-level use cases are super-ordinate to the refined use cases. The refined use cases are sub-ordinate to the top-level use cases. Note that only the complex use cases should be decomposed and organized in a hierarchy. It is not necessary to decompose simple use cases. The functionality of the super-ordinate use cases is traceable to their sub-ordinate use cases. Thus, the functionality provided by the super-ordinate use cases is composite of the functionality of the sub-ordinate use cases. In the highest level of the use case model, only the fundamental use cases are shown. The focus is on the application context. Therefore, this level is also referred to as the context diagram. In the context diagram, the system limits are emphasized. In the top-level diagram, only those use cases with which external users of the system. The subsystem-level use cases specify the services offered by the subsystems. Any number of levels involving the subsystems may be utilized. In the lowest level of the use case hierarchy, the class-level use cases specify the functional fragments or operations offered by the classes.

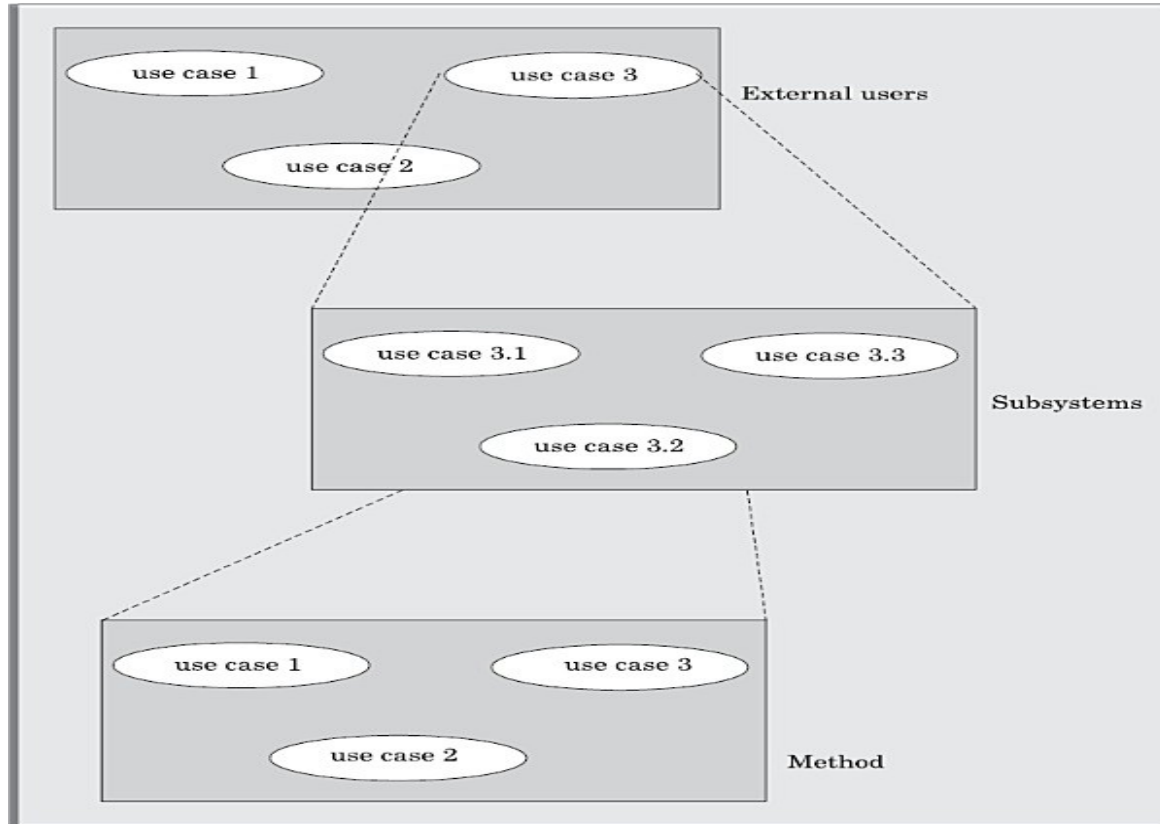


Figure: Hierarchical organisation of use cases.

Class diagrams

A class diagram describes the static structure of a system. It shows how a system is structured rather than how it behaves. The static structure of a system comprises of a number of class diagrams and their dependencies. The main constituents of a class diagram are classes and their relationships: generalization, aggregation, association, and various kinds of dependencies.

Classes

The classes represent entities with common features, i.e. attributes and operations. Classes are represented as solid outline rectangles with compartments. Classes have a mandatory name compartment where the name is written centered in boldface. The class name is usually written using mixed case convention and begins with an uppercase. The class names are usually chosen to be singular nouns. An example of a class is shown in fig.

Classes have optional attributes and operations compartments. A class may appear on several diagrams. Its attributes and operations are suppressed on all but one diagram.

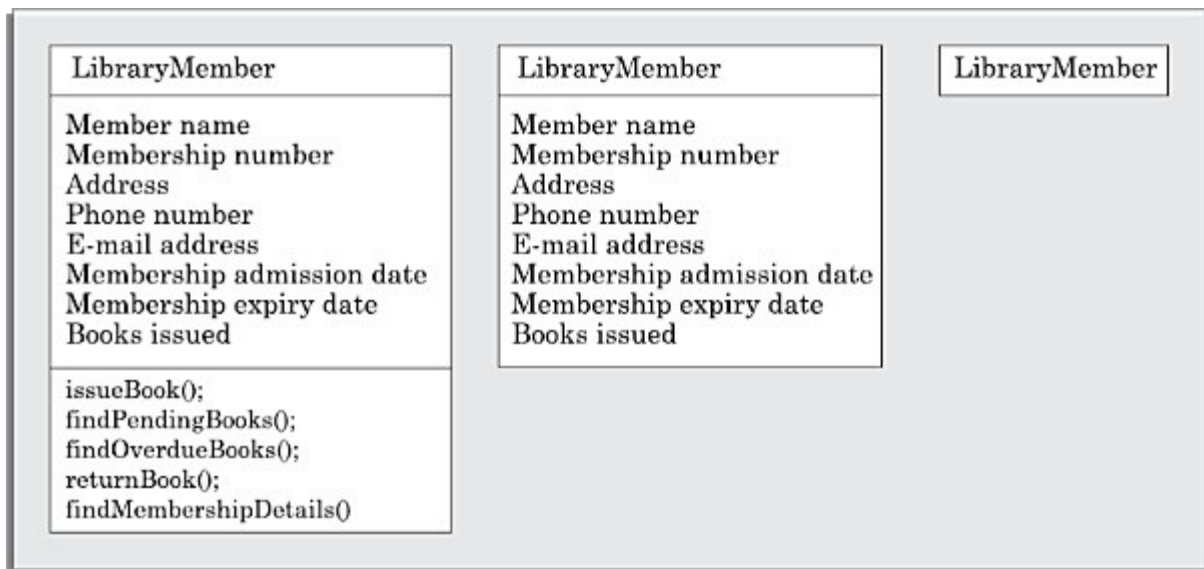


Figure: Different representations of the LibraryMember class.

Attributes

An attribute is a named property of a class. It represents the kind of data that an object might contain. Attributes are listed with their names, and may optionally contain specification of their type,(that is, their class, e.g., Int, Book, Employee, etc.), an initial value, and constraints. The type of the attribute is written by appending a colon and the type name after the attribute name. Typically, the first letter of a class name is a small letter. An example for an attribute is given. e.g. sensorStatus[1]:Int=0.

bookName : String

Operation

Operation is the implementation of a service that can be requested from any object of the class to affect behaviour. An object's data or state can be changed by invoking an operation of the object. A class may have any number of operations or no operation at all. Typically, the first letter of an operation name is a small letter. Abstract operations are written in italics. The parameters of an operation (if any), may have a kind

specified, which may be 'in', 'out' or 'inout'. An operation may have a return type consisting of a single return type expression. An example for an operation is given.

issueBook(in bookName) : Boolean

Association

Associations are needed to enable objects to communicate with each other. An association describes a connection between classes. The association relation between two objects is called object connection or link. Links are instances of associations. A link is a physical or conceptual connection between object instances. Mathematically, a link can be considered to be a tuple, i.e. an ordered list of object instances. An association describes a group of links with a common structure and common semantics. For example, consider the statement that Library Member borrows Books. Here, borrows is the association between the class LibraryMember and the class Book. Usually, an association is a binary relation (between two classes). However, three or more different classes can be involved in an association. A class can have an association relationship with itself (called recursive association). In this case, it is usually assumed that two different objects of the class are linked by the association relationship.

Association between two classes is represented by drawing a straight line between the concerned classes. It illustrates the graphical representation of the association relation. The name of the association is written along side the association line. An arrowhead may be placed on the association line to indicate the reading direction of the association. The arrowhead should not be misunderstood to be indicating the direction of a pointer implementing an association. On each side of the association relation, the multiplicity is noted as an individual number or as a value range. The multiplicity indicates how many instances of one class are associated with each other. Value ranges of multiplicity are noted by specifying the minimum and maximum value, separated by two dots, e.g. An asterisk is a wild card and means many (zero or more). The association of fig. should be read as "Many books may be borrowed by a Library Member". Observe that associations (and links) appear as verbs in the problem statement.

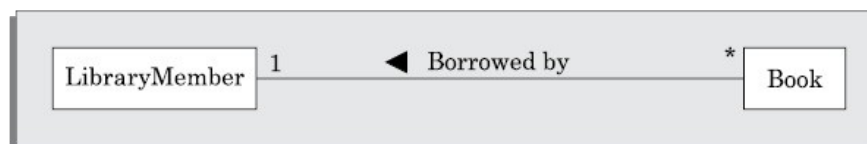


Figure: Association between two classes.

Associations are usually realized by assigning appropriate reference attributes to the classes involved. Thus, associations can be implemented using pointers from one object class to another. Links and associations can also be implemented by using a separate class that stores which objects of a class are linked to which objects of another class. Some CASE tools use the role names of the association relation for the corresponding automatically generated attribute.

Aggregation

Aggregation is a special type of association where the involved classes represent a whole-part relationship. The aggregate takes the responsibility of forwarding messages to the appropriate parts. Thus, the aggregate takes the responsibility of delegation and leadership. When an instance of one object contains instances of some other objects, then aggregation (or composition) relationship exists between the composite

object and the component object. Aggregation is represented by the diamond symbol at the composite end of a relationship. The number of instances of the component class aggregated can also be shown as in fig.

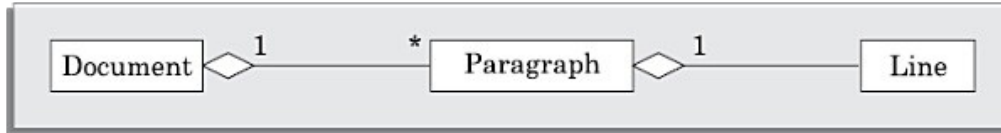


Fig: Representation of aggregation

Aggregation relationship cannot be reflexive (i.e. recursive). That is, an object cannot contain objects of the same class as itself. Also, the aggregation relation is not symmetric. That is, two classes A and B cannot contain instances of each other. However, the aggregation relationship can be transitive. In this case, aggregation may consist of an arbitrary number of levels.

Composition

Composition is a stricter form of aggregation, in which the parts are existence-dependent on the whole. This means that the life of the parts closely ties to the life of the whole. When the whole is created, the parts are created and when the whole is destroyed, the parts are destroyed. A typical example of composition is an invoice object with invoice items. As soon as the invoice object is created, all the invoice items in it are created and as soon as the invoice object is destroyed, all invoice items in it are also destroyed. The composition relationship is represented as a filled diamond drawn at the composite-end. An example of the composition relationship is shown in fig.



Fig: Representation of composition

Dependency:

A dependency relationship is shown as a dotted arrow that is drawn from the dependent class to the independent class.



Figure:Representation of dependence between classes.

Association vs. Aggregation vs. Composition

- Association is the most general (m:n) relationship. Aggregation is a stronger relationship where one is a part of the other. Composition is even stronger than aggregation, ties the lifecycle of the part and the whole together.
- Association relationship can be reflexive (objects can have relation to itself), but aggregation cannot be reflexive. Moreover, aggregation is anti-symmetric (If B is a part of A, A can not be a part of B).

- Composition has the property of exclusive aggregation i.e. an object can be a part of only one composite at a time. For example, a **Frame** belongs to exactly one **Window** whereas in simple aggregation, a part may be shared by several objects. For example, a **Wall** may be a part of one or more **Room** objects.
- In addition, in composition, the whole has the responsibility for the disposition of all its parts, i.e. for their creation and destruction.
 - in general, the lifetime of parts and composite coincides
 - parts with non-fixed multiplicity may be created after composite itself
 - parts might be explicitly removed before the death of the composite

For example, when a **Frame** is created, it has to be attached to an enclosing **Window**. Similarly, when the **Window** is destroyed, it must in turn destroy its **Frame** parts

Inheritance vs. Aggregation/Composition

- Inheritance describes '*is a*' / '*is a kind of*' relationship between classes (base class - derived class) whereas aggregation describes '*has a*' relationship between classes. Inheritance means that the object of the derived class inherits the properties of the base class; aggregation means that the object of the whole has objects of the part. For example, the relation "cash payment *is a kind of* payment" is modeled using inheritance; "purchase order has a few items" is modeled using aggregation.

Inheritance is used to model a "generic-specific" relationship between classes whereas aggregation/composition is used to model a "whole-part" relationship between classes.

- Inheritance means that the objects of the subclass can be used anywhere the super class may appear, but not the reverse; i.e. wherever we could use instances of 'payment' in the system, we could substitute it with instances of 'cash payment', but the reverse can not be done.
- Inheritance is defined statically. It can not be changed at run-time. Aggregation is defined dynamically and can be changed at run-time. Aggregation is used when the type of the object can change over time.

For example, consider this situation in a business system. A **BusinessPartner** might be a **Customer** or a **Supplier** or both. Initially we might be tempted to model it as in Fig 7.12(a). But in fact, during its lifetime, a business partner might become a customer as well as a supplier, or it might change from one to the other. In such cases, we prefer aggregation instead (see Fig 7.12(b)). Here, a business partner is a **Customer** if it has an aggregated **Customer** object, a **Supplier** if it has an aggregated **Supplier** object and a "**Customer_Supplier**" if it has both. Here, we have only two types. Hence, we are able to model it as inheritance. But what if there were several different types and combinations there of? The inheritance tree would be absolutely incomprehensible.

Also, the aggregation model allows the possibility for a business partner to be neither - i.e. has neither a customer nor a supplier object aggregated with it.

- The advantage of aggregation is the integrity of encapsulation. The operations of an object are the interfaces of other objects which imply low implementation dependencies. The significant disadvantage of aggregation is the increase in the number of objects and their relationships. On the other hand, inheritance allows for an easy way to modify implementation for reusability. But the significant disadvantage is that it breaks encapsulation, which implies implementation dependence.

Object diagrams

Object diagrams shows the snapshot of the objects in a system at a point in time. Since it shows instances of classes, rather than the classes themselves, it is often called as an instance diagram. The objects are drawn using rounded rectangles

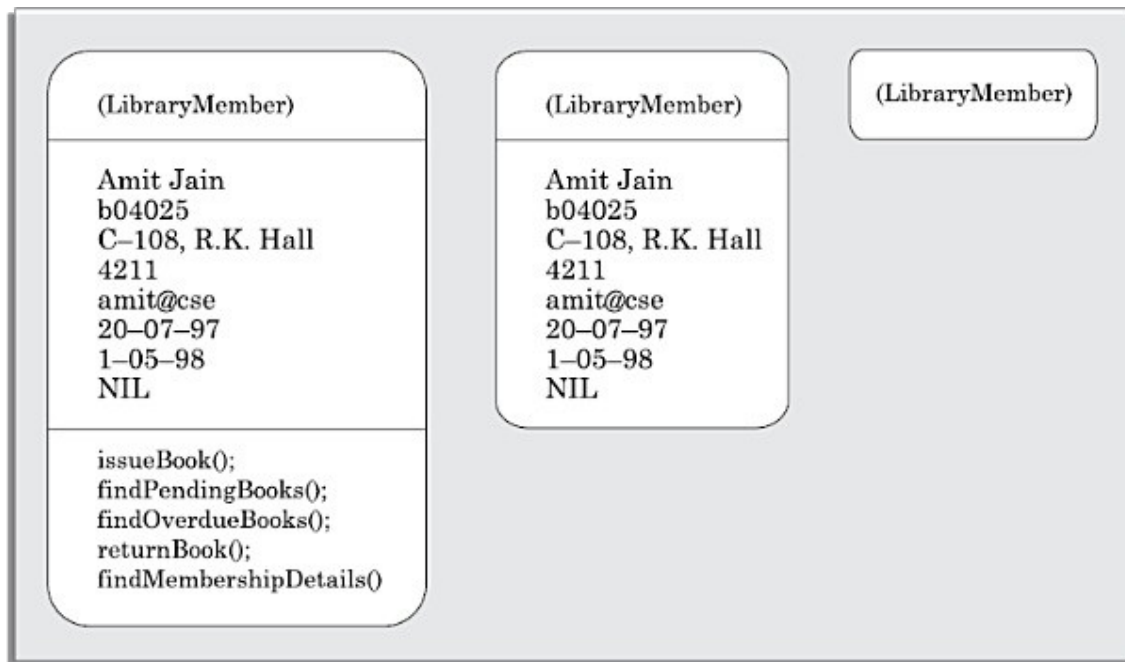


Figure: Different representations of a `LibraryMember` object.

An object diagram may undergo continuous change as execution proceeds. For example, links may get formed between objects and get broken. Objects may get created and destroyed, and so on. Object diagrams are useful to explain the working of a system.

Interaction Diagrams

Interaction diagrams are models that describe how group of objects collaborate to realize some behavior. Typically, each interaction diagram realizes the behavior of a single use case. An interaction diagram shows a number of example objects and the messages that are passed between the objects within the use case.

There are two kinds of interaction diagrams: sequence diagrams and collaboration diagrams. These two diagrams are equivalent in the sense that any one diagram can be derived automatically from the other. However, they are both useful. These two actually portray different perspectives of behavior of the system and different types of inferences can be drawn from them. The interaction diagrams can be considered as a major tool in the design methodology.

Sequence Diagram

A sequence diagram shows interaction among objects as a two dimensional chart. The chart is read from top to bottom. The objects participating in the interaction are shown at the top of the chart as boxes attached to a vertical dashed line. Inside the box the name of the object is written with a colon separating it from the name of the class and both the name of the object and the class are underlined. The objects appearing at the top signify that the object already existed when the use case execution was initiated. However, if some object is created during the execution of the use case and participates in the interaction

(e.g. a method call), then the object should be shown at the appropriate place on the diagram where it is created. The vertical dashed line is called the object's lifeline. The lifeline indicates the existence of the object at any particular point of time. The rectangle drawn on the lifetime is called the activation symbol and indicates that the object is active as long as the rectangle exists. Each message is indicated as an arrow between the lifeline of two objects. The messages are shown in chronological order from the top to the bottom. That is, reading the diagram from the top to the bottom would show the sequence in which the messages occur. Each message is labeled with the message name. Some control information can also be included. Two types of control information are particularly valuable.

- A condition (e.g. [invalid]) indicates that a message is sent, only if the condition is true.
- An iteration marker shows the message is sent many times to multiple receiver objects as would happen when a collection or the elements of an array are being iterated. The basis of the iteration can also be indicated e.g. [for every book object].

The sequence diagram for the book renewal use case for the Library Automation Software is shown in fig The development of the sequence diagram in the development methodology would help us in determining the responsibilities of the different classes; i.e. what methods should be supported by each class.

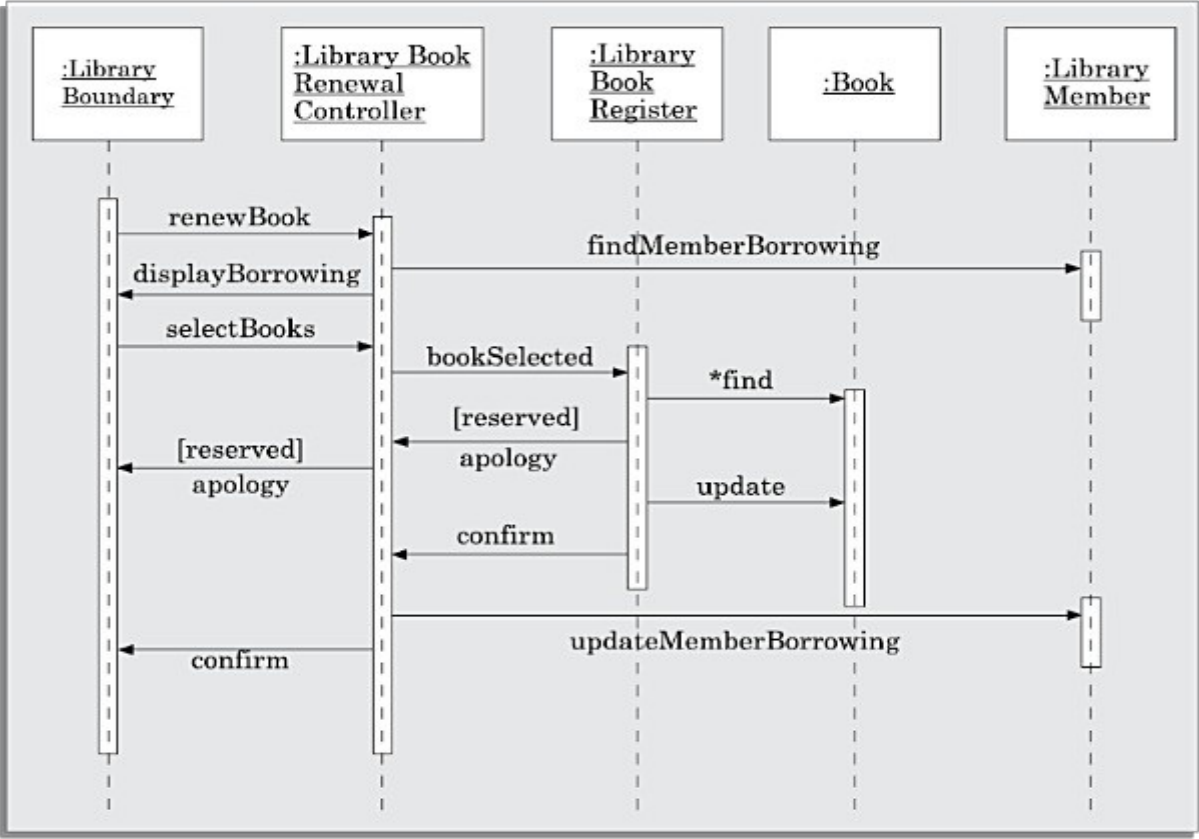


Figure: Sequence diagram for the renew book use case

Collaboration Diagram

A collaboration diagram shows both structural and behavioral aspects explicitly. This is unlike a sequence diagram which shows only the behavioral aspects. The structural aspect of a collaboration diagram consists of objects and the links existing between them. In this diagram, an object is also called a collaborator. The behavioral aspect is described by the set of messages exchanged among the different collaborators. The link between objects is shown as a solid line and can be used to send messages between two objects. The message is shown as a labeled arrow placed near the link. Messages are prefixed with sequence numbers because they are only way to describe the relative sequencing of the messages in this diagram. The collaboration diagram for the example of fig. The use of the collaboration diagrams in our development process would be to help us to determine which classes are associated with which other classes.

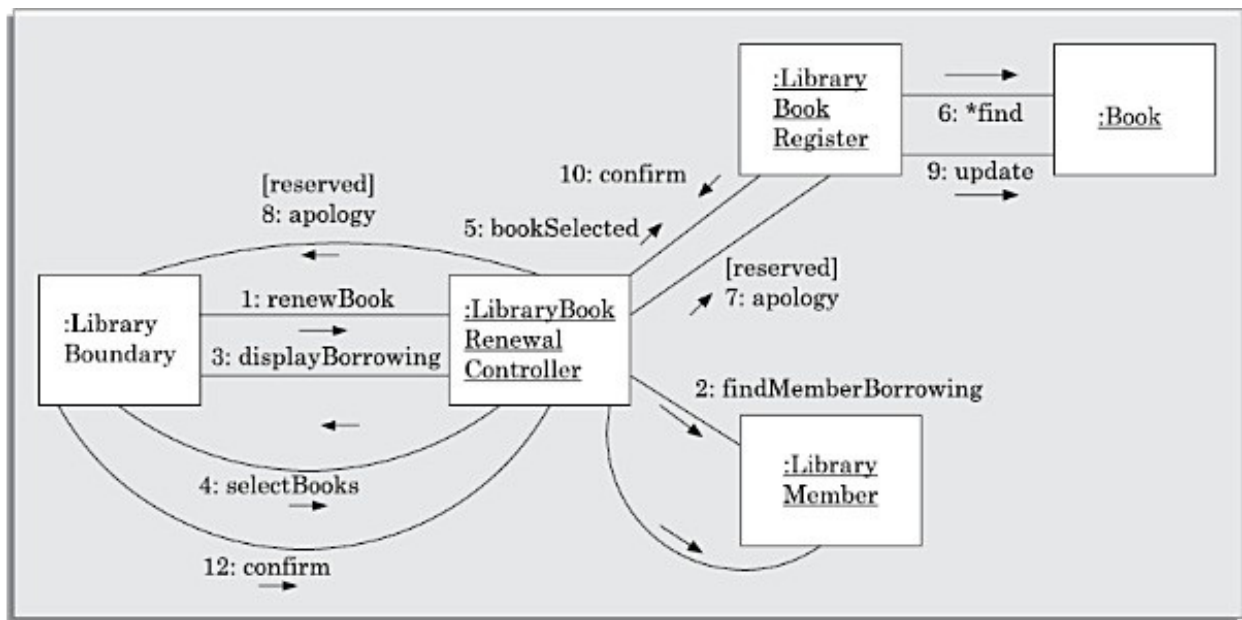


Figure: Collaboration diagram for the renew book use case.

Activity diagrams

The activity diagram is possibly one modeling element which was not present in any of the predecessors of UML. The activity diagram focuses on representing activities or chunks of processing which may or may not correspond to the methods of classes. An activity is a state with an internal action and one or more outgoing transitions which automatically follow the termination of the internal activity. If an activity has more than one outgoing transitions, then these must be identified through conditions. An interesting feature of the activity diagrams is the swim lanes. Swim lanes enable you to group activities based on who is performing them, e.g. academic department vs. hostel office. Thus swim lanes subdivide activities based on the responsibilities of some components. The activities in a swim lane can be assigned to some model elements, e.g. classes or some component, etc.

Activity diagrams are normally employed in business process modeling. This is carried out during the initial stages of requirements analysis and specification. Activity diagrams can be very useful to understand complex processing activities involving many components. Later these diagrams can be used to develop interaction diagrams which help to allocate activities (responsibilities) to classes.

The student admission process in IIT is shown as an activity diagram. This shows the part played by different components of the Institute in the admission procedure. After the fees are received at the account section, parallel activities start at the hostel office, hospital, and the Department. After all these activities complete (this synchronization is represented as a horizontal line), the identity card can be issued to a student by the Academic section.

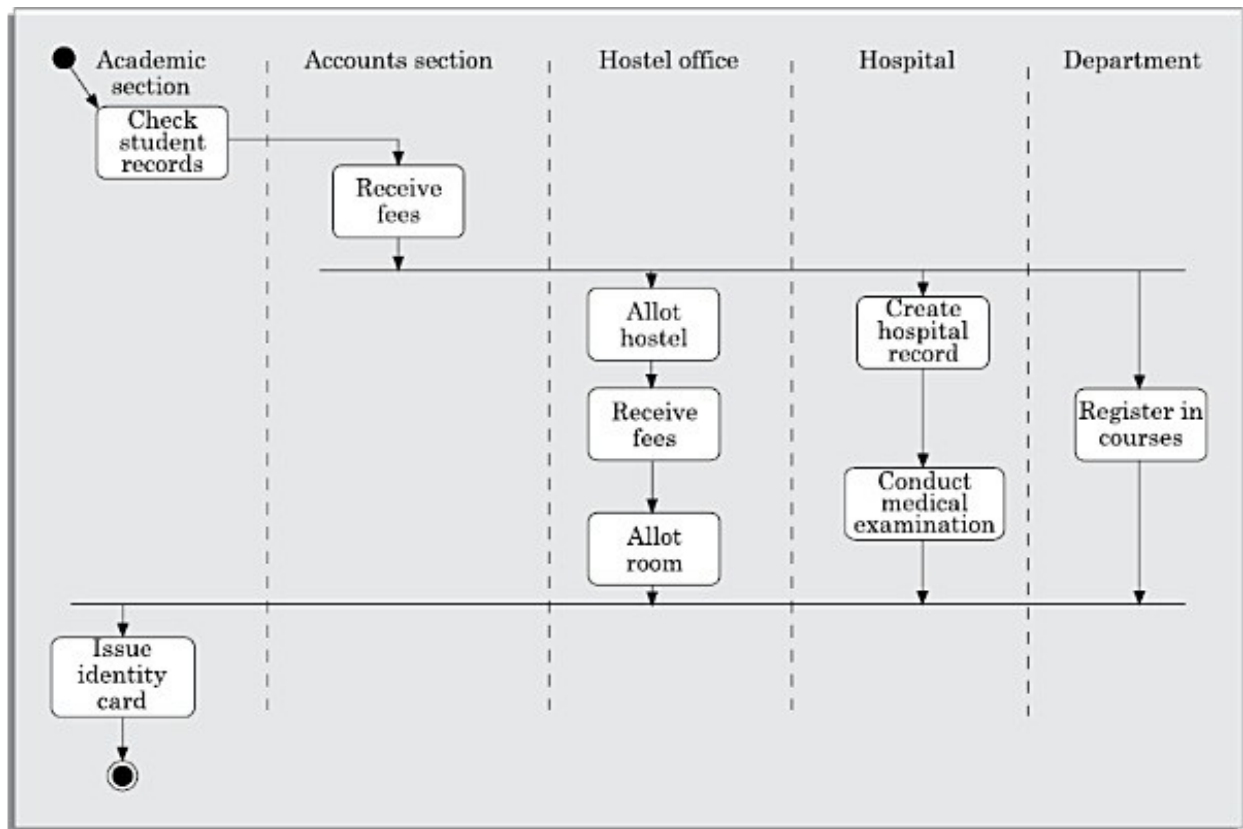


Figure: Activity diagram for student admission procedure at IIT.

Activity diagrams are normally employed in business process modelling. This is carried out during the initial stages of requirements analysis and specification. Activity diagrams can be very useful to understand complex processing activities involving the roles played by many components. Besides helping the developer to understand the complex processing activities, these diagrams can also be used to develop interaction diagrams which help to allocate activities (responsibilities) to classes.

State chart diagram

A state chart diagram is normally used to model how the state of an object changes in its lifetime. State chart diagrams are good at describing how the behavior of an object changes across several use case executions. However, if we are interested in modeling some behavior that involves several objects collaborating with each other, state chart diagram is not appropriate. State chart diagrams are based on the finite state machine (FSM) formalism.

An FSM consists of a finite number of states corresponding to those of the object being modeled. The object undergoes state changes when specific events occur. The FSM formalism existed long before the object-oriented technology and has been used for a wide variety of applications. Apart from modeling, it has even been used in theoretical computer science as a generator for regular languages.

A major disadvantage of the FSM formalism is the state explosion problem. The number of states becomes too many and the model too complex when used to model practical systems. This problem is overcome in UML by using state charts. The state chart formalism was proposed by David Harel [1990]. A state chart is a hierarchical model of a system and introduces the concept of a composite state (also called nested state).

Actions are associated with transitions and are considered to be processes that occur quickly and are not interruptible. Activities are associated with states and can take longer. An activity can be interrupted by an event.

The basic elements of the state chart diagram are as follows:

Initial state. This is represented as a filled circle.

Final state. This is represented by a filled circle inside a larger circle.

State. These are represented by rectangles with rounded corners.

Transition. A transition is shown as an arrow between two states. Normally, the name of the event which causes the transition is placed along side the arrow. A guard to the transition can also be assigned. A guard is a Boolean logic condition. The transition can take place only if the guard evaluates to true. The syntax for the label of the transition is shown in 3 parts: event[guard]/action.

An example state chart for the order object of the Trade House Automation software is shown in fig.

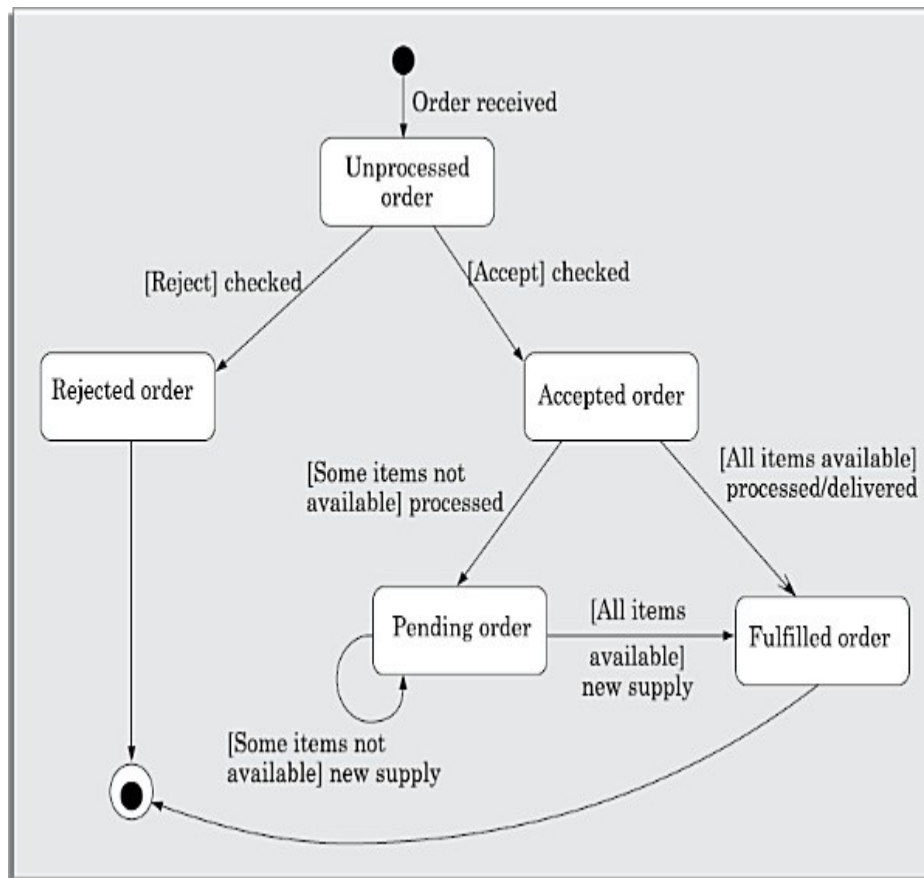


Figure: State chart diagram for an order object.

Package diagram

A package is a grouping of several classes. In fact, a package diagram can be used to group any UML artifacts. We have already discussed packaging of use cases. Packages are a popular way of organising source code files. Java packages are a good example which can be modelled using a package diagram. Such package diagrams show the different class groups (packages) and their inter dependencies. These are very useful to document organisation of source files for large projects that have a large number of program files. An example of a package diagram has been shown in Figure

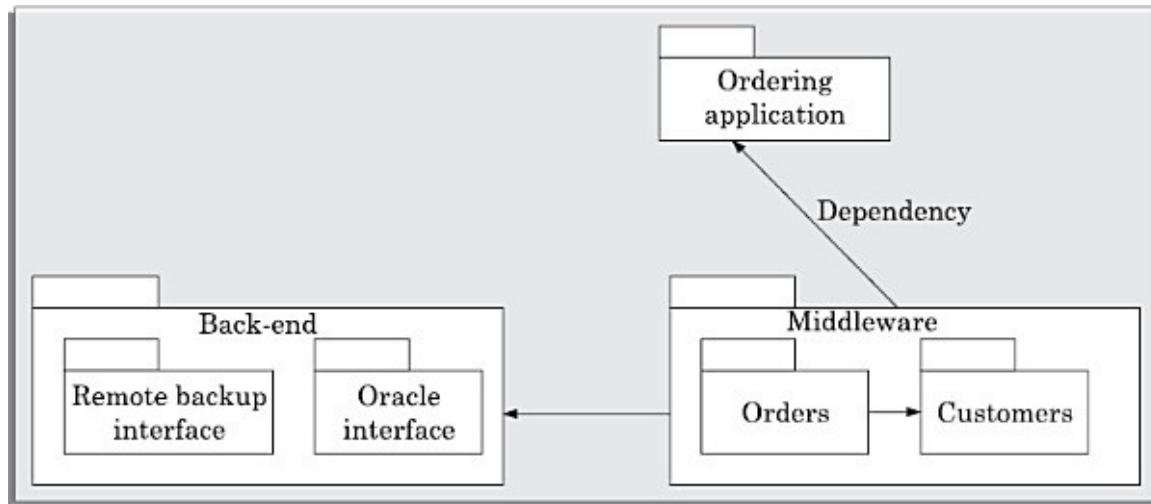


Figure: An example package diagram.

Component diagram

A component represents a piece of software that can be independently purchased, upgraded, and integrated into an existing software. A component diagram can be used to represent the physical structure of an implementation in terms of the various components of the system. A component diagram is typically used to achieve the following purposes:

- Organise source code to be able to construct executable releases.
- Specify dependencies among different components.
- A package diagram can be used to provide a high-level view of each component in terms of the different classes it contains.

Deployment diagram

The deployment diagram shows the environmental view of a system. That is, it captures the environment in which the software solution is implemented. In other words, a deployment diagram shows how a software system will be physically deployed in the hardware environment. That is, which component will execute on which hardware component and how they will communicate with each other. Since the diagram models the run time architecture of an application, this diagram can be very useful to the system's operation staff.

- The environmental view provided by the deployment diagram is important for complex and large software solutions that run on hardware systems comprising multiple components. In this case, deployment diagram provides an overview of how the different components are distributed among the different hardware components of the system.

Composite structure diagram

The composite structure diagram lets you define how a class is defined by a further structure of classes and the communication paths between these parts. Some new core constructs such as parts, ports and connectors are introduced.

Part: The concept of parts makes possible the description of the internal structure of a class.

Port: The concept of a port makes it possible to describe connection points formally. These are addressable, which means that signals can be sent to them.

Connector: Connectors can be used to specify the communication links between two or more parts.

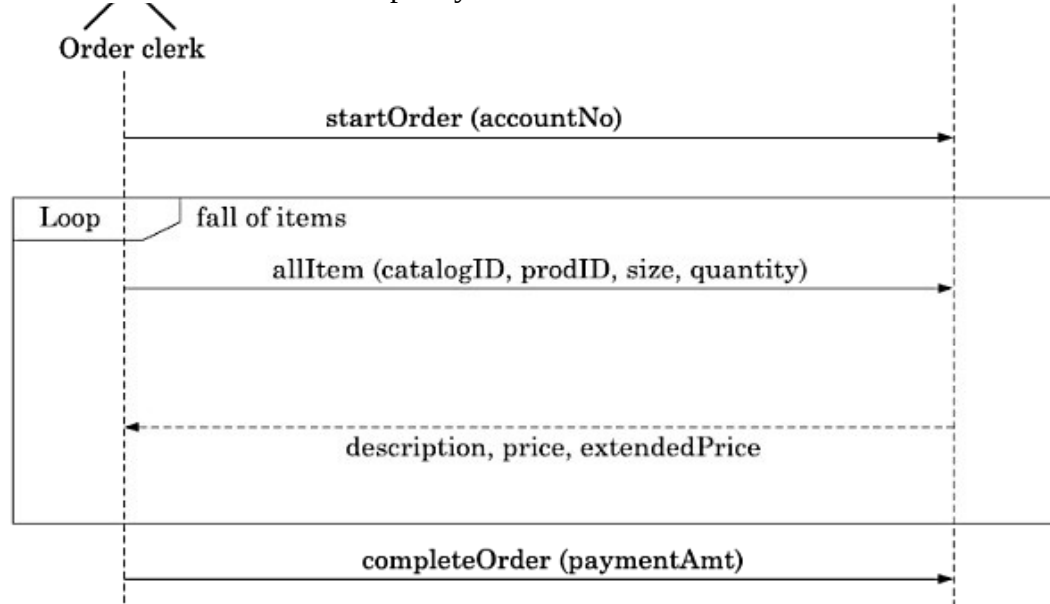


Figure: An example sequence diagram showing a combined fragment in UML 2.0.

UML defines thirteen types of diagrams, divided into three categories as follows:

Structure diagrams: These include the class diagram, object diagram, component diagram, composite structure diagram, package diagram, and deployment diagram.

Behaviour diagrams: These diagrams include the usecase diagram, activity diagram, and state machine diagram.

Interaction diagrams: These diagrams include the sequence diagram, communication diagram, timing diagram, and interaction overview diagram. The collaboration diagram of UML 1.X has been renamed in UML 2.0 as communication diagram. This renaming was necessary as the earlier name was somewhat misleading, it shows the communications among the classes during the execution of a use case rather than showing collaborative problem solving.

Characteristics of a user interface

It is very important to identify the characteristics desired of a good user interface. Because unless we are aware of these, it is very much difficult to design a good user interface. A few important characteristics of a good user interface are the following:

- **Speed of learning.** A good user interface should be easy to learn. Speed of learning is hampered by complex syntax and semantics of the command issue procedures. A good user interface should not require its users to memorize commands. Neither should the user be asked to remember information from one screen to another while performing various tasks using the interface. Besides, the following three issues are crucial to enhance the speed of learning:
- **Use of Metaphors and intuitive command names.** Speed of learning an interface is greatly facilitated if these are based on some day-to-day real-life examples or some physical objects with which the users are familiar. The abstractions of real-life objects or concepts used in user interface design are called metaphors. If the user interface of a text editor uses concepts similar to the tools used by a writer for text editing such as cutting lines and paragraphs and pasting it at other places, users can immediately relate to it. Another popular metaphor is a shopping cart. Everyone knows how a shopping cart is used to make choices while purchasing items in a supermarket. If a user interface uses the shopping cart metaphor for designing the interaction style for a situation where similar types of choices have to be made, then the users can easily understand and learn to use the interface. Yet another example of a metaphor is the trashcan. To delete a file, the user may drag it to the trashcan. Also, learning is facilitated by intuitive command names and symbolic command issue procedures.
- **Consistency.** Once a user learns about a command, he should be able to use the similar commands in different circumstances for carrying out similar actions. This makes it easier to learn the interface

since the user can extend his knowledge about one part of the interface to the other parts. For example, in a word processor, “Control-b” is the short-cut key to embolden the selected text. The same short-cut should be used on the other parts of the interface, for example, to embolden text in graphic objects also - circle, rectangle, polygon, etc. Thus, the different commands supported by an interface should be consistent.

- **Component-based interface.** Users can learn an interface faster if the interaction style of the interface is very similar to the interface of other applications with which the user is already familiar. This can be achieved if the interfaces of different applications are developed using some standard user interface components. This, in fact, is the theme of the component-based user interface. Examples of standard user interface components are: radio button, check box, text field, slider, progress bar, etc.

The speed of learning characteristic of a user interface can be determined by measuring the training time and practice that users require before they can effectively use the software.

- **Speed of use.** Speed of use of a user interface is determined by the time and user effort necessary to initiate and execute different commands. This characteristic of the interface is some times referred to as productivity support of the interface. It indicates how fast the users can perform their intended tasks. The time and user effort necessary to initiate and execute different commands should be minimal. This can be achieved through careful design of the interface. For example, an interface that requires users to type in lengthy commands or involves mouse movements to different areas of the screen that are wide apart for issuing commands can slow down the operating speed of users. The most frequently used commands should have the smallest length or be available at the top of the menu to minimize the mouse movements necessary to issue commands.
- **Speed of recall.** Once users learn how to use an interface, the speed with which they can recall the command issue procedure should be maximized. This characteristic is very important for intermittent users. Speed of recall is improved if the interface is based on some metaphors, symbolic command issue procedures, and intuitive command names.

- **Error prevention.** A good user interface should minimize the scope of committing errors while initiating different commands. The error rate of an interface can be easily determined by monitoring the errors committed by average users while using the interface. This monitoring can be automated by instrumenting the user interface code with monitoring code which can record the frequency and types of user error and later display the statistics of various kinds of errors committed by different users.

Moreover, errors can be prevented by asking the users to confirm any potentially destructive actions specified by them, for example, deleting a group of files. Consistency of names, issue procedures, and behavior of similar commands and the simplicity of the command issue procedures minimize error possibilities. Also, the interface should prevent the user from entering wrong values.

- **Aesthetic and attractive.** A good user interface should be attractive to use. An attractive user interface catches user attention and fancy. In this respect, graphics-based user interfaces have a definite advantage over text-based interfaces.
- **Consistency.** The commands supported by a user interface should be consistent. The basic purpose of consistency is to allow users to generalize the knowledge about aspects of the interface from one part to another. Thus, consistency facilitates speed of learning, speed of recall, and also helps in reduction of error rate.
- **Feedback.** A good user interface must provide feedback to various user actions. Especially, if any user request takes more than few seconds to process, the user should be informed about the state of the processing of his request. In the absence of any response from the computer for a long time, a novice user might even start recovery/shutdown procedures in panic. If required, the user should be periodically informed about the progress made in processing his command.
For example, if the user specifies a file copy/file download operation, a progress bar can be displayed to display the status. This will help the user to monitor the status of the action initiated.
- **Support for multiple skill levels.** A good user interface should support multiple levels of sophistication of command issue procedure for different categories of users. This is necessary because users with different levels of experience in using an application prefer different types of user interfaces. Experienced users are more concerned about the efficiency of the command issue procedure, whereas novice users pay importance to usability aspects. Very cryptic and complex commands discourage a novice, whereas elaborate command sequences make the command issue procedure very slow and therefore put off experienced users. When someone uses an application for the first time, his primary concern is speed of learning. After using an application for extended periods of time, he becomes familiar with the operation of the software. As a user becomes more and more familiar with an interface, his focus shifts from usability aspects to speed of command issue aspects. Experienced users look for options such as “hot-keys”, “macros”, etc. Thus, the skill level of users improves as they keep using a software product and they look for commands to suit their skill levels.
- **Error recovery (undo facility).** While issuing commands, even the expert users can commit errors. Therefore, a good user interface should allow a user to undo a mistake committed by him while using the interface. Users are put to inconvenience, if they cannot recover from the errors they commit while using the software.
- **User guidance and on-line help.** Users seek guidance and on-line help when they either forget a command or are unaware of some features of the software. Whenever users need guidance or seek help from the system, they should be provided with the appropriate guidance and help.

User guidance and online help

Users may seek help about the operation of the software any time while using the software. This is provided by the on-line help system. This is different from the guidance and error messages which are flashed automatically without the user asking for them. The guidance messages prompt the user regarding the options he has regarding the next command, and the status of the last command, etc.

- **On-line Help System.** Users expect the on-line help messages to be tailored to the context in which they invoke the “help system”. Therefore, a good on-line help system should keep track of what a user is doing while invoking the help system and provide the output message in a context-dependent way. Also, the help messages should be tailored to the user’s experience level. Further, a good on-line help system should take advantage of any graphics and animation characteristics of the screen and should not just be a copy of the user’s manual.gives a snapshot of a typical on-line help provided by a user interface.

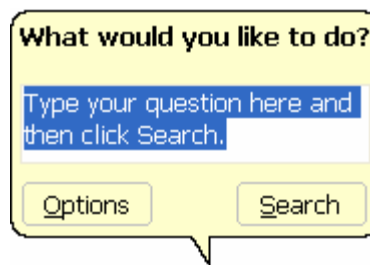


Fig. Example of an on-line help interface

- **Guidance Messages.** The guidance messages should be carefully designed to prompt the user about the next actions he might pursue, the current status of the system, the progress made so far in processing his last command, etc. A good guidance system should have different levels of sophistication for different categories of users. For example, a user using a command language interface might need a different type of guidance compared to a user using a menu or iconic interface. Also, users should have an option to turn off detailed messages.

Mode-based interface vs. modeless interface

- - A mode is a state or collection of states in which only a subset of all user interaction tasks can be performed. In a modeless interface, the same set of commands can be invoked at any time during the running of the software. Thus, a modeless interface has only a single mode and all the commands are available all the time during the operation of the software. On the other hand, in a mode-based interface, different set of commands can be invoked depending on the mode in which the system is, i.e. the mode at any instant is determined by the sequence of commands already issued by the user.
- A mode-based interface can be represented using a state transition diagram, where each node of the state transition diagram would represent a mode. Each state of the state transition diagram can be annotated with the commands that are meaningful in that state.

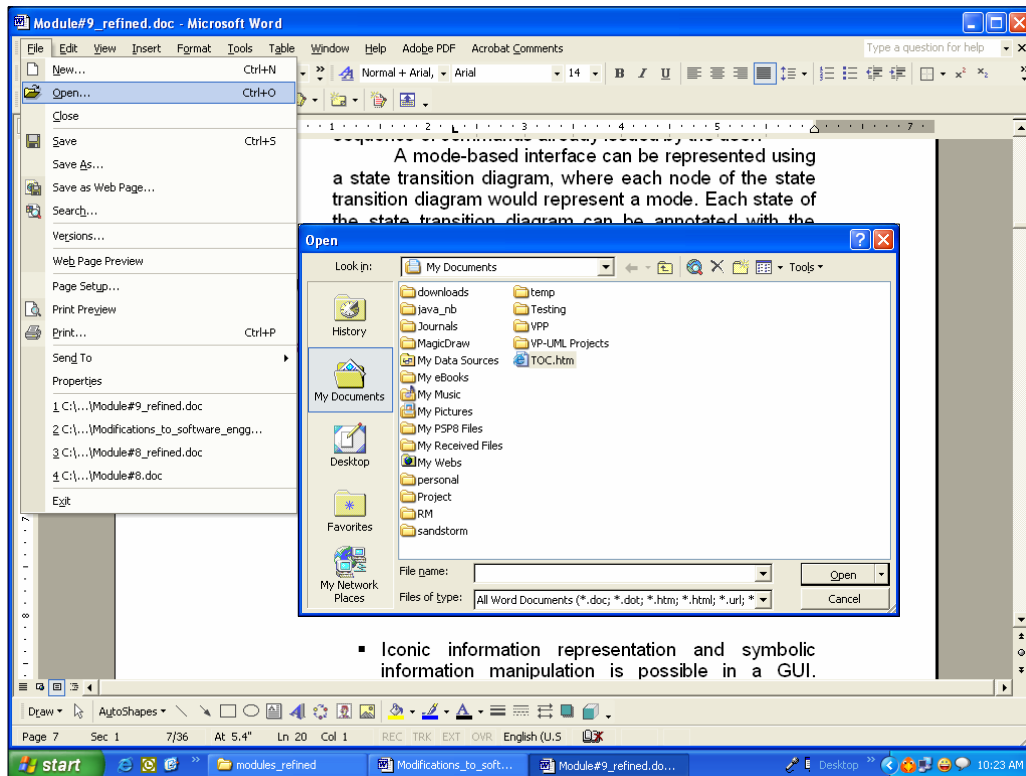


Fig:An example of mode-based interface

It shows the interface of a word processing program. The top-level menu provides the user with a gamut of operations like file open, close, save, etc. When the user chooses the open option, another frame is popped up which limits the user to select a name from one of the folders.

Types of user interfaces

User interfaces can be classified into the following three categories:

- Command language based interfaces
- Menu-based interfaces
- Direct manipulation interfaces

1.Command Language-based Interface

A command language-based interface – as the name itself suggests, is based on designing a command language which the user can use to issue the commands. The user is expected to frame the appropriate commands in the language and type them in appropriately whenever required. A simple command language-based interface might simply assign unique names to the different commands. However, a more sophisticated command language-based interface may allow users to compose complex commands by using a set of primitive commands. Such a facility to compose commands dramatically reduces the number of command names one would have to remember. Thus, a command language-based interface can be made concise requiring minimal typing by the user. Command language-based interfaces allow fast interaction with the computer and simplify the input of complex commands.

2.Menu-based Interface

An important advantage of a menu-based interface over a command language-based interface is that a menu-based interface does not require the users to remember the exact syntax of the commands. A menu-based interface is based on recognition of the command names, rather than recollection. Further, in a menu-based interface the typing effort is minimal as most interactions are carried out through menu selections using a pointing device. This factor is an important consideration for the occasional user who cannot type fast.

However, experienced users find a menu-based user interface to be slower than a command language-based interface because an experienced user can type fast and can get speed advantage by composing different primitive commands to express complex commands. Composing commands in a menu-based interface is not possible. This is because of the fact that actions involving logical connectives (and, or, etc.) are awkward to specify in a menu-based system. Also, if the number of choices is large, it is difficult to select from the menu. In fact, a major challenge in the design of a menu-based interface is to structure large number of menu choices into manageable forms. When the menu choices are large, they can be structured as the following way:

Scrolling menu

When a full choice list can not be displayed within the menu area, scrolling of the menu items is required. This would enable the user to view and select the menu items that cannot be accommodated on the screen. However, in a scrolling menu all the commands should be highly correlated, so that the user can easily locate a command that he needs. This is important since the user cannot see all the commands at any one time. An example situation where a scrolling menu is frequently used is font size selection in a document processor. Here, the user knows that the command list contains only the font sizes that are arranged in some order and he can scroll up and down to find the size he is looking for. However, if the commands do not have any definite ordering relation, then the user would have to in the worst case, scroll through all the commands to find the exact command he is looking for, making this organization inefficient.

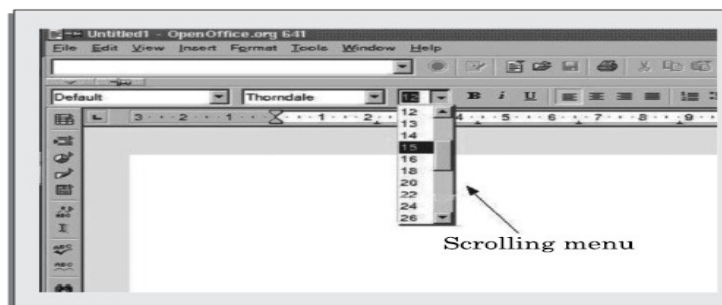


Fig: Font size selection using scrolling menu

Walking menu

Walking menu is very commonly used to structure a large collection of menu items. In this technique, when a menu item is selected, it causes further menu items to be displayed adjacent to it in a sub-menu. An example of a walking menu. A walking menu can successfully be used to

structure commands only if there are tens rather than hundreds of choices since each adjacently displayed menu does take up screen space and the total screen area is after limited.

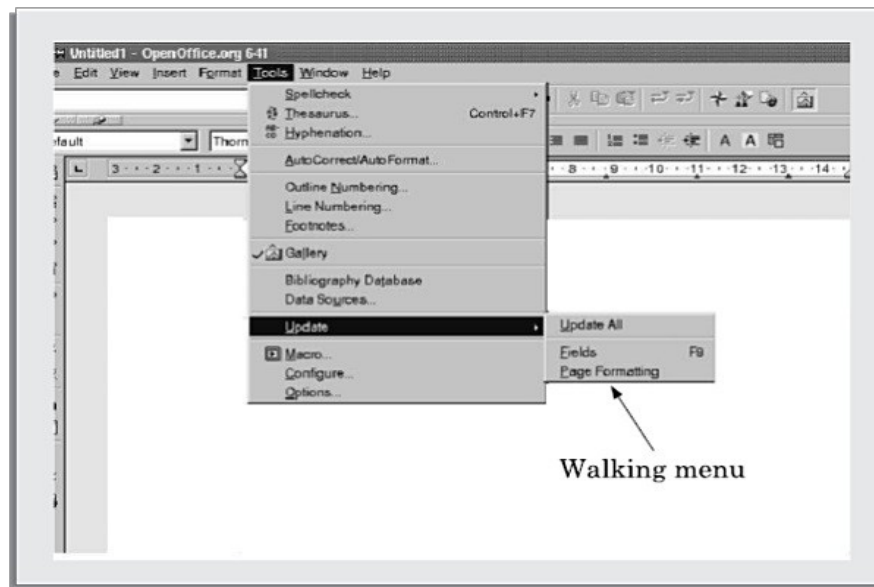


Fig.: Example of walking menu

Hierarchical menu

In this technique, the menu items are organized in a hierarchy or tree structure. Selecting a menu item causes the current menu display to be replaced by an appropriate sub-menu. Thus in this case, one can consider the menu and its various sub-menus to form a hierarchical tree-like structure. Walking menu can be considered to be a form of hierarchical menu which is practicable when the tree is shallow. Hierarchical menu can be used to manage large number of choices, but the users are likely to face navigational problems because they might lose track of where they are in the menu tree. This probably is the main reason why this type of interface is very rarely used.

3.Direct Manipulation Interfaces

Direct manipulation interfaces present the interface to the user in the form of visual models (i.e. icons or objects). For this reason, direct manipulation interfaces are sometimes called as iconic interface. In this type of interface, the user issues commands by performing actions on the visual representations of the objects, e.g. pull an icon representing a file into an icon representing a trash box, for deleting the file. Important advantages of iconic interfaces include the fact that the icons can be recognized by the users very easily, and that icons are language-independent. However, direct manipulation interfaces can be considered slow for experienced users. Also, it is difficult to give complex commands using a direct manipulation interface. For example, if one has to drag an icon representing the file to a trash box icon for deleting a file, then in order to delete all the files in the directory one has to perform this operation individually for all files – which could be very easily done by issuing a command like delete *.*.

Characteristics of command language-based interface have been discussed earlier.

Disadvantages of command language-based interface

- Command language-based interfaces suffer from several drawbacks. Usually, command language-based interfaces are difficult to learn and require the user to memorize the set of primitive commands. Also, most users make errors while formulating commands in the command language and also while typing them in. Further, in a command language-based interface, all interactions with the system is through a key-board and can not take advantage of effective interaction devices such as a mouse. Obviously, for casual and inexperienced users, command language-based interfaces are not suitable.

Issues in designing a command language-based interface

- Two overbearing command design issues are to reduce the number of primitive commands that a user has to remember and to minimize the total typing required while issuing commands. These can be elaborated as follows:
- The designer has to decide what mnemonics are to be used for the different commands. The designer should try to develop meaningful mnemonics and yet be concise to minimize the amount of typing required. For example, the shortest mnemonic should be assigned to the most frequently used commands.
- The designer has to decide whether the users will be allowed to redefine the command names to suit their own preferences. Letting a user define his own mnemonics for various commands is a useful feature, but it increases the complexity of user interface development.
- The designer has to decide whether it should be possible to compose primitive commands to form more complex commands. A sophisticated command composition facility would require the syntax and semantics of the various command composition options to be clearly and unambiguously specified. The ability to combine commands is a powerful facility in the hands of experienced users, but quite unnecessary for inexperienced users.

Types of menus and their features

Three main types of menus are scrolling menu, walking menu, and hierarchical menu. The features of scrolling menu, walking menu, and hierarchical menu have been discussed earlier.

Iconic interface

- Direct manipulation interfaces present the interface to the user in the form of visual models (i.e. icons or objects). For this reason, direct manipulation interfaces are sometimes called iconic interfaces. In this type of interface, the user issues commands by performing actions on the visual representations of the objects, e.g. pull an icon representing a file into an icon representing a trash box, for deleting the file.

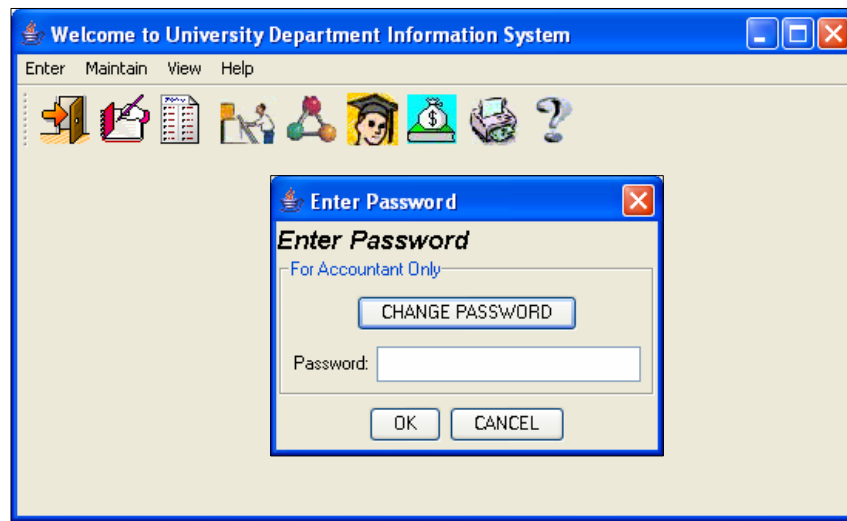


Fig:Example of an iconic interface

It shows an iconic interface. Here, the user is presented with a set of icons at the top of the frame for performing various activities. On clicking on any of the icons, either the user is prompted with a sub menu or the desired activity is performed.

Component-based GUI development

A development style based on widgets (window objects) is called component- based (or widget-based) GUI development style. There are several important advantages of using a widget-based design style. One of the most important reasons to use widgets as building blocks is because they help users learn an interface fast. In this style of development, the user interfaces for different applications are built from the same basic components. Therefore, the user can extend his knowledge of the behavior of the standard components from one application to the other. Also, the component-based user interface development style reduces the application programmer's work significantly as he is more of a user interface component integrator than a programmer in the traditional sense.

Need for component-based GUI development

The current style of user interface development is component-based. It recognizes that every user interface can easily be built from a handful of predefined components such as menus, dialog boxes, forms, etc. Besides the standard components, and the facilities to create good interfaces from them, one of the basic support available to the user interface developers is the window system. The window system lets the application programmer create and manipulate windows without having to write the basic windowing functions.

Visual Programming

Visual programming is the drag and drop style of program development. In this style of user interface development, a number of visual objects (icons) representing the GUI components are provided by the programming environment. The application programmer can easily develop the user interface by dragging the required component types (e.g. menu, forms, etc.) from the displayed icons and placing them wherever required. Thus, visual programming can be considered as program

development through manipulation of several visual objects. Reuse of program components in the form of visual objects is an important aspect of this style of programming. Though popular for user interface development, this style of programming can be used for other applications such as Computer-Aided Design application (e.g. factory design), simulation, etc. User interface development using a visual programming language greatly reduces the effort required to develop the interface.

Examples of popular visual programming languages are Visual Basic, Visual C++, etc. Visual C++ provides tools for building programs with window-based user interfaces for Microsoft Windows environments. In Visual C++, menu bars, icons, and dialog boxes, etc. can be designed easily before adding them to program. These objects are called as resources. Shape, location, type, and size of the dialog boxes can be designed before writing any C++ code for the application.

Window

A window is a rectangular area on the screen. A window can be considered to be a virtual screen, in the sense that it provides an interface to the user for carrying out independent activities, e.g. one window can be used for editing a program and another for drawing pictures, etc.

A window can be divided into two parts: client part, and non-client part. The client area makes up the whole of the window, except for the borders and scroll bars. The client area is the area available to a client application for display. The non-client part of the window determines the look and feel of the window. The look and feel defines a basic behavior for all windows, such as creating, moving, resizing, and iconifying the windows. A basic window with its different parts.

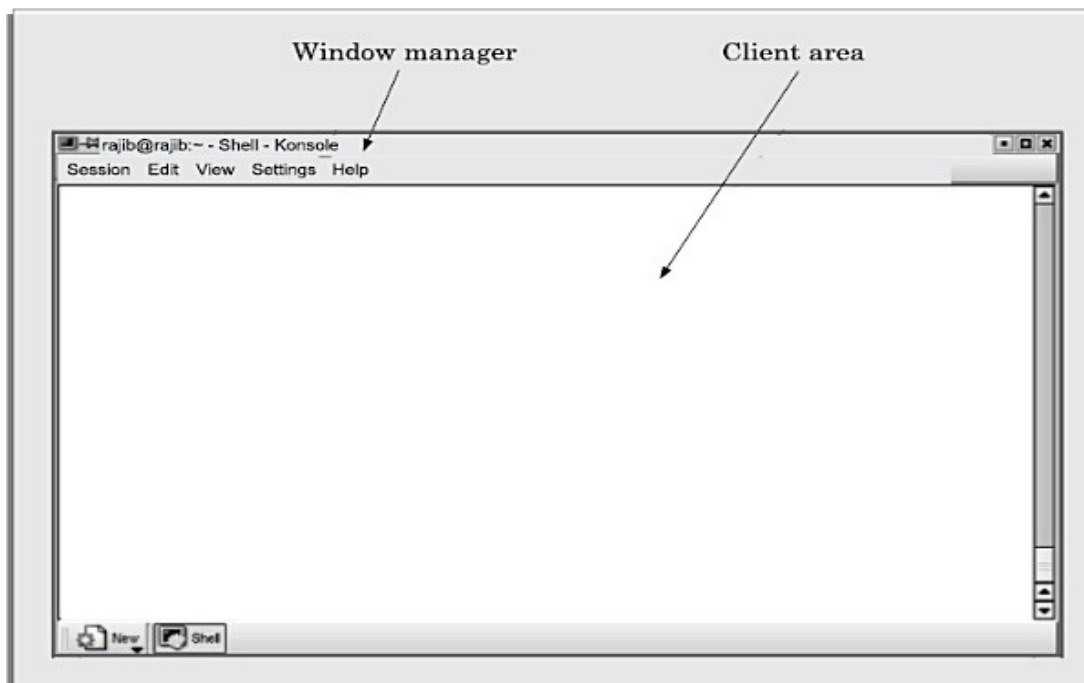


Fig: Window with client and user areas marked

Window Management System(WMS)

A graphical user interface typically consists of a large number of windows. Therefore, it is necessary to have some systematic way to manage these windows. Most graphical user interface development environments do this through a window management system (WMS). A window management system is primarily a resource manager. It keeps track of the screen area resource and allocates it to the different windows that seek to use the screen. From a broader perspective, a WMS can be considered as a user interface management system (UIMS) – which not only does resource management, but also provides the basic behavior to the windows and provides several utility routines to the application programmer for user interface development. A WMS simplifies the task of a GUI designer to a great extent by providing the basic behavior to the various windows such as move, resize, iconify, etc. as soon as they are created and by providing the basic routines to manipulate the windows from the application such as creating, destroying, changing different attributes of the windows, and drawing text, lines, etc.

A WMS consists of two parts:

- a window manager, and
- a window system.

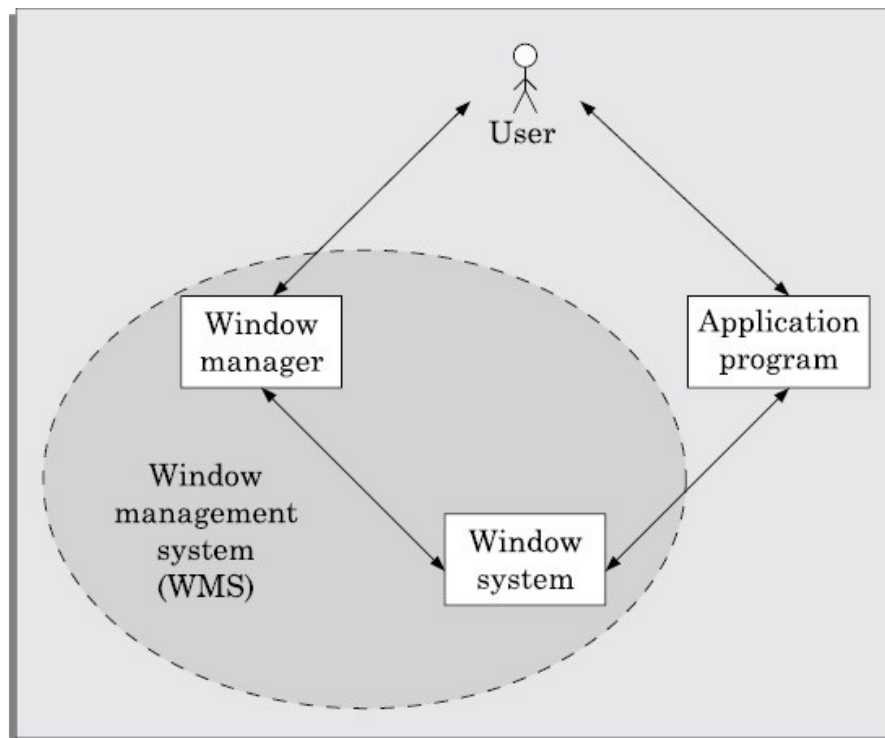


Figure: Window management system.

Window Manager and Window System

Window manager is the component of WMS with which the end user interacts to do various window-related operations such as window repositioning, window resizing, iconification, etc. The window manager is built on the top of the window system in the sense that it makes use of various services provided by the window system. The window manager and not the window system determines how the windows look and behave. In fact, several kinds of window managers can be developed based

on the same window system. The window manager can be considered as a special kind of client that makes use of the services (function calls) supported by the window system. The application programmer can also directly invoke the services of the window system to develop the user interface. The relationship between the window manager, window system, and the application program is shown in [fig.](#) This figure shows that the end-user can either interact with the application itself or with the window manager (resize, move, etc.) and both the application and the window manager invoke services of the window manager.

Window Manager

The window manager is responsible for managing and maintaining the non-client area of a window. Window manager manages the real-estate policy, provides look and feel of each individual window.

Types of widgets (window objects)

Different interface programming packages support different widget sets. However, a surprising number of them contain similar kinds of widgets, so that one can think of a generic widget set which is applicable to most interfaces. The following widgets are representatives of this generic class.

Label widget. This is probably one of the simplest widgets. A label widget does nothing except to display a label, i.e. it does not have any other interaction capabilities and is not sensitive to mouse clicks. A label widget is often used as a part of other widgets.

Container widget. These widgets do not stand by themselves, but exist merely to contain other widgets. Other widgets are created as children of the container widget. When the container widget is moved or resized, its children widget also get moved or resized. A container widget has no callback routines associated with it.

Pop-up menu. These are transient and task specific. A pop-up menu appears upon pressing the mouse button, irrespective of the mouse position.

Pull-down menu. These are more permanent and general. You have to move the cursor to a specific location and pull down this type of menu.

Dialog boxes. We often need to select multiple elements from a selection list. A dialog box remains visible until explicitly dismissed by the user. A dialog box can include areas for entering text as well as values. If an apply command is supported in a dialog box, the newly entered values can be tried without dismissing the box. Through most dialog boxes ask you to enter some information, there are some dialog boxes which are merely informative, alerting you to a problem with your system or an error you have made. Generally, these boxes ask you to read the information presented and then click **OK** to dismiss the box.

Push button. A push button contains key words or pictures that describe the action that is triggered when you activate the button. Usually, the action related to a push button occurs immediately when you click a push button unless it contains an ellipsis (...). A push button with an ellipsis generally indicates that another dialog box will appear.

Radio buttons. A set of radio buttons is used when only one option has to be selected out of many options. A radio button is a hollow circle followed by text describing the option it stands for. When a radio button is selected, it appears filled and the previously selected radio button from the group is unselected. Only one radio button from a group can be selected at any time. This operation is similar to that of the band selection buttons that were available in old radios.

Combo boxes. A combo box looks like a button until the user interacts with it. When the user presses or clicks it, the combo box displays a menu of items to choose from. Normally a combo box is used to display either one-of-many choices when space is limited, the number of choices is large, or when the menu items are computed at run-time.

X-Window.

The X-window functions are low-level functions written in C language which can be called from application programs. But only the very serious application designer would program directly using the X-windows library routines. Built on top of X-windows are higher-level functions collectively called Xtoolkit. Xtoolkit consists of a set of basic widgets and a set of routines to manipulate these widgets. One of the most widely used widget sets is X/Motif. Digital Equipment Corporation (DEC) used the basic X-window functions to develop its own look and feel for interface designs called DECWindows.

Popularity of X-Window

One of the important reasons behind the extreme popularity of the X-window system is probably due to the fact that it allows development of portable GUIs. Applications developed using X-window system are device-independent. Also, applications developed using the X-window system become network independent in the sense that the interface would work just as well on a terminal connected anywhere on the same network as the computer running the application is. Network-independent GUI operation has been schematically represented in the fig. 9.8. Here, “A” is the computer application in which the application is running. “B” can be any computer on the network from where interaction with the application can be made. Network-independent GUI was pioneered by the X-window system in the mid-eighties at MIT (Massachusetts Institute of Technology) with support from DEC (Digital Equipment Corporation). Now-a-days many user interface development systems support network-independent GUI development, e.g. the AWT and Swing components of Java.

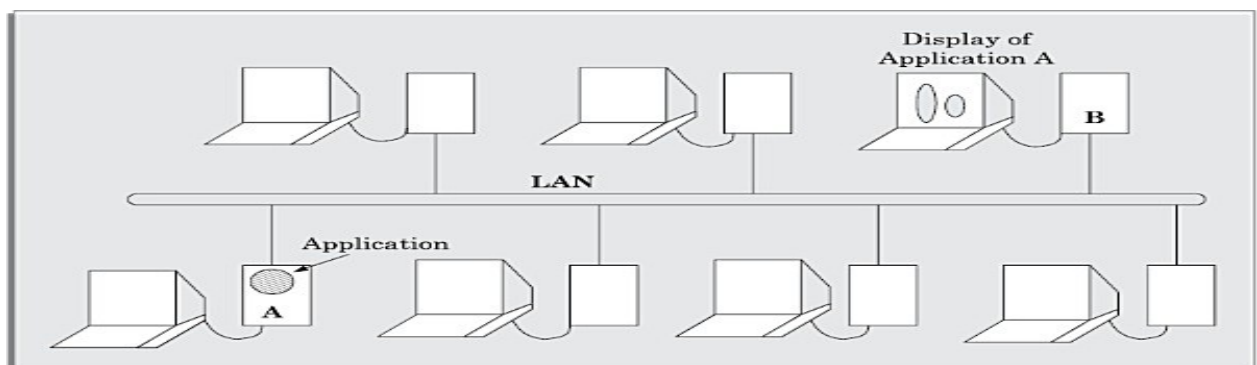


Figure: Network-independent GUI.

Architecture of an X-System The X-architecture is pictorially depicted in fig. The different terms used in this diagram are explained below.

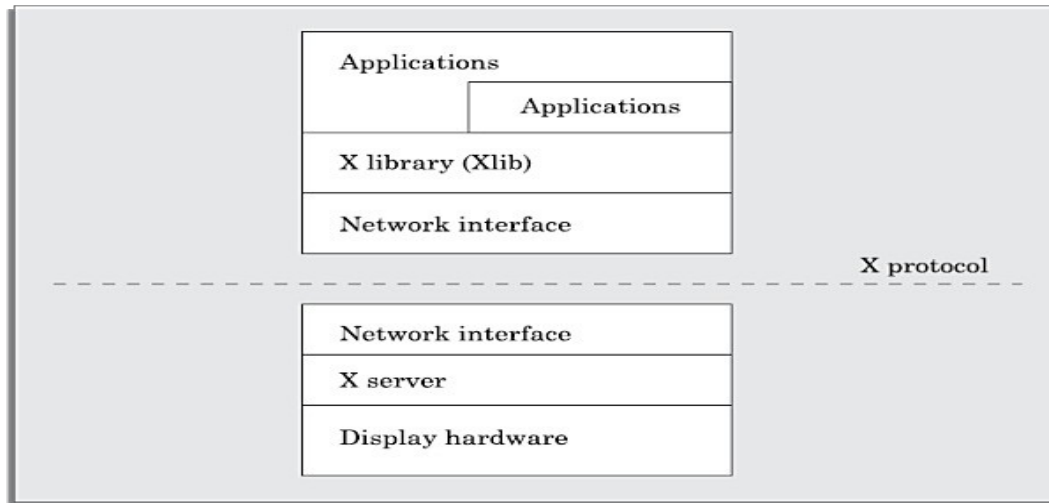


Figure: Architecture of the X System.

X-server. The X server runs on the hardware to which the display and keyboard are attached. The X server performs low-level graphics, manages windows, and user input functions. The X server controls accesses to a bit-mapped graphics display resource and manages it.

X-protocol. The X protocol defines the format of the requests between client applications and display servers over the network. The X protocol is designed to be independent of hardware, operating systems, underlying network protocol, and the programming language used.

X-library (Xlib). The Xlib provides a set of about 300 utility routines for applications to call. These routines convert procedure calls into requests that are transmitted to the server. Xlib provides low-level primitives for developing a user interface, such as displaying a window, drawing characteristics and graphics on the window, waiting for specific events, etc.

Xtoolkit (Xt). The Xtoolkit consists of two parts: the intrinsics and the widgets. We have already seen that widgets are predefined user interface components such as scroll bars, push buttons, etc. for designing GUIs. Intrinsics are a set of about a dozen library routines that allow a programmer to combine a set of widgets into a user interface. In order to develop a user interface, the designer has to put together the set of components (widgets) he needs, and then he needs to define the characteristics (called resources) and behavior of these widgets by using the intrinsic routines to complete the development of the interface. Therefore, developing an interface using Xtoolkit is much easier than developing the same interface using only X library.

USER INTERFACE DESIGN METHODOLOGY

At present, no step-by-step methodology is available which can be followed by rote to come up with a good user interface. What we present in this section is a set of recommendations which you can use to complement your ingenuity. Even though almost all popular GUI design methodologies are user-centered, this concept has to be clearly distinguished from a user interface design by users. Before we start discussing about the user interface design methodology, let us distinguish between a user-centered design and a design by users.

- User-centered design is the theme of almost all modern user interface design techniques. However, user-centered design does not mean design by users. One should not get the users to design the interface, nor should one assume that the user's opinion of which design alternative is superior is always right. Though users may have good knowledge of the tasks they have to perform using a GUI, but they may not know the GUI design issues.
- Users have good knowledge of the tasks they have to perform, they also know whether they find an interface easy to learn and use but they have less understanding and experience in GUI design than the GUI developers.

Implications of Human Cognition Capabilities on User Interface Design

An area of human-computer interaction where extensive research has been conducted is how human cognitive capabilities and limitations influence the way an interface should be designed. In the following subsections, we discuss some of the prominent issues that have been extensively reported in the literature.

Limited memory: Humans can remember at most seven unrelated items of information for short periods of time. Therefore, the GUI designer should not require the user to remember too many items of information at a time. It is the GUI designer's responsibility to anticipate what information the user will need at what point of each task and to ensure that the relevant information is displayed for the user to see. Showing the user some information at some point, and then asking him to recollect that information in a different screen where they no longer see the information, places a memory burden on the user and should be avoided wherever possible.

Frequent task closure: Doing a task (except for very trivial tasks) requires doing several subtasks. When the system gives a clear feedback to the user that a task has been successfully completed, the user gets a sense of achievement and relief. The user can clear out information regarding the completed task from memory. This is known as task closure. When the overall task is fairly big and complex, it should be divided into subtasks, each of which has a clear subgoal which can be a closure point.

Recognition rather than recall. Information recall incurs a larger memory burden on the users and is to be avoided as far as possible. On the other hand, recognition of information from the alternatives shown to him is more acceptable.

Procedural versus object-oriented: Procedural designs focus on tasks, prompting the user in each step of the task, giving them very few options for anything else. This approach is best applied in situations where the tasks are narrow and well-defined or where the users are inexperienced, such as a bank ATM. An object-oriented interface on the other hand focuses on objects. This allows the users a wide range of options.

GUI design methodology

GUI design methodology consists of the following important steps:

- Examine the use case model of the software. Interview, discuss, and review the GUI issues with the end-users.
- Task and object modeling

- Metaphor selection
- Interaction design and rough layout
- Detailed presentation and graphics design
- GUI construction
- Usability evaluation

The starting point for GUI design is the use case model. This captures the important tasks the users need to perform using the software. As far as possible, a user interface should be developed using one or more metaphors. Metaphors help in interface development at lower effort and reduced costs for training the users. Over time, people have developed efficient methods of dealing with some commonly occurring situations. These solutions are the themes of the metaphors. Metaphors can also be based on physical objects such as a visitor's book, a catalog, a pen, a brush, a scissor, etc. A solution based on metaphors is easily understood by the users, reducing learning time and training costs. Some commonly used metaphors are the following:

- Shopping cart
- whiteboard
- Desktop
- Editor's work bench
- White page
- Yellow page
- Office cabinet
- Post box
- Bulletin board
- Visitor's book

Task and Object Modeling

A task is a human activity intended to achieve some goals. Example of task goals can be:

- reserve an airline seat
- buy an item
- transfer money from one account to another
- book a cargo for transmission to an address

A task model is an abstract model of the structure of a task. A task model should show the structure of the subtasks that the user needs to perform to achieve the overall task goal. Each task can be modeled as a hierarchy of subtasks. A task model can be drawn using a graphical notation similar to the activity network model. Tasks can be drawn as boxes with lines showing how a task is broken down into subtasks. An underlined task box would mean that no further decomposition of the task is required. An example decomposition of a task into subtasks is shown in fig.

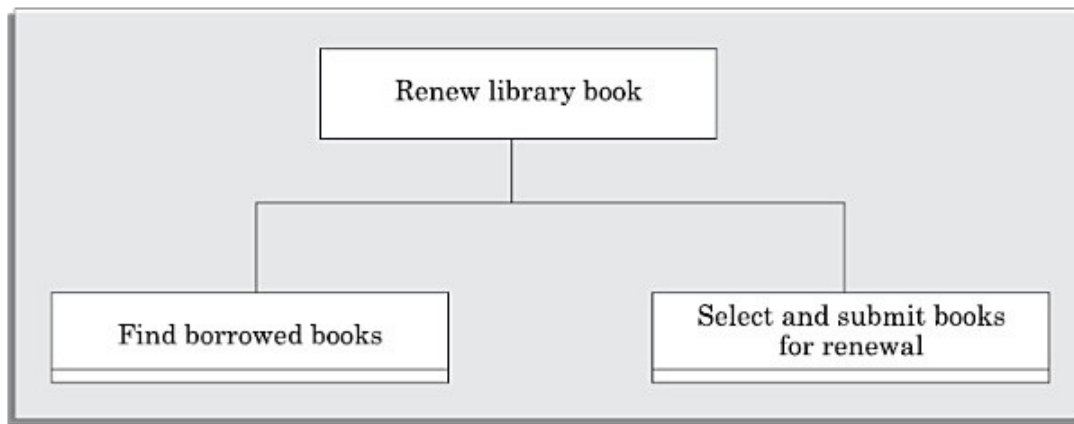


Fig: Decomposition of a task into subtasks

Selecting a metaphor

The first place one should look for while trying to identify the candidate metaphors is the set of parallels to objects, tasks, and terminologies of the use cases. If no obvious metaphors can be found, then the designer can fall back on the metaphors of the physical world of concrete objects. The appropriateness of each candidate metaphor should be tested by restating the objects and tasks of the user interface model in terms of the metaphor. Another criterion that can be used to judge metaphors is that the metaphor should be as simple as possible, the operations using the metaphor should be clear and coherent and it should fit with the users' 'common sense' knowledge. For example, it would indeed be very awkward and a nuisance for the users if the scissor metaphor is used to glue different items.

Example: We need to develop the interface for the automation shop, where the users can examine the contents of the shop through a web interface and can order them.

Several metaphors are possible for different parts of this problem.

- Different items can be picked up from **racks** and examined. The user can request for the **catalog** associated with the items by clicking on the item.
- Related items can be picked from the **drawers** of an item cabinet.
- The items can be organized in the form of a book, similar to the way information about electronic components are organized in a semiconductor hand book.

Once the users make up their mind about an item they wish to buy, they can put them into a **shopping cart**.

Visibility of the system status. The system should as far as possible keep the user informed about the status of the system and what is going on.

Match between the system and the real world. The system should speak the user's language words, phrases, and concepts familiar to that used by the user, rather than using system-oriented terms.

Undoing mistakes. The user should feel that he is in control rather than feeling helpless or to be at the control of the system. An important step toward this is that the users should be able to undo and redo operations.

Consistency. The user should not have to wonder whether different words, concepts, and operations mean the same thing in different situations.

Recognition rather than recall. The user should not have to recall information which was presented in another screen. All data and instructions should be visible on the screen for selection by the user.

Support for multiple skill levels. Provision of accelerations for experienced users allows them to efficiently carry out the actions they frequently require to perform.

Aesthetic and minimalist design. Dialogs should not contain information which are irrelevant and are rarely needed. Every extra unit of information in a dialog competes with the relevant units and diminishes their visibility.

Help and error messages. These should be expressed in plain language (no codes), precisely indicating the problem, and constructively suggesting a solution.

Error prevention. Error possibilities should be minimized. A key principle in this regard is to prevent the user from entering wrong values. In situations where a choice has to be made from among a discrete set of values, the control should present only the valid values using a drop-down list, a set of option buttons or a similar multichoice control. When a specific format is required for attribute data, the entered data should be validated when the user attempts to submit the data.

Coding- The objective of the coding phase is to transform the design of a system into code in a high level language and then to unit test this code. The programmers adhere to standard and well defined style of coding which they call their coding standard. The main advantages of adhering to a standard style of coding are as follows:

- A coding standard gives uniform appearances to the code written by different engineers
- It facilitates code of understanding.
- Promotes good programming practices.

For implementing our design into a code, we require a good high level language. A programming language should have the following features:

Characteristics of a Programming Language

- **Readability:** A good high-level language will allow programs to be written in some ways that resemble a quite-English description of the underlying algorithms. If care is taken, the coding may be done in a way that is essentially self-documenting.
- **Portability:** High-level languages, being essentially machine independent, should be able to develop portable software.
- **Generality:** Most high-level languages allow the writing of a wide variety of programs, thus relieving the programmer of the need to become expert in many diverse languages.
- **Brevity:** Language should have the ability to implement the algorithm with less amount of code. Programs expressed in high-level languages are often considerably shorter than their low-level equivalents.
- **Error checking:** Being human, a programmer is likely to make many mistakes in the development of a computer program. Many high-level languages enforce a great deal of error checking both at compile-time and at run-time.
- **Cost:** The ultimate cost of a programming language is a function of many of its characteristics.
- **Familiar notation:** A language should have familiar notation, so it can be understood by most of the programmers.
- **Quick translation:** It should admit quick translation.
- **Efficiency:** It should permit the generation of efficient object code.
- **Modularity:** It is desirable that programs can be developed in the language as a collection of separately compiled modules, with appropriate mechanisms for ensuring self-consistency between these modules.
- **Widely available:** Language should be widely available and it should be possible to provide translators for all the major machines and for all the major operating systems.

A coding standard lists several rules to be followed during coding, such as the way variables are to be named, the way the code is to be laid out, error return conventions, etc.

Coding standards and guidelines

Good software development organizations usually develop their own coding standards and guidelines depending on what best suits their organization and the type of products they develop.

The following are some representative coding standards.

1. **Rules for limiting the use of global:** These rules list what types of data can be declared global and what cannot.
2. **Contents of the headers preceding codes for different modules:** The information contained in the headers of different modules should be standard for an organization. The exact format in which the header information is organized in the header can also be specified. The following are some standard header data:
 1. Name of the module.
 2. Date on which the module was created.
 3. Author's name.
 4. Modification history.
 5. Synopsis of the module.
 6. Different functions supported, along with their input/output parameters.
 7. Global variables accessed/modified by the module.
3. **Naming conventions for global variables, local variables, and constant identifiers:** A possible naming convention can be that global variable names always start with a capital letter, local variable names are made of small letters, and constant names are always capital letters.
4. **Error return conventions and exception handling mechanisms:** The way error conditions are reported by different functions in a program are handled should be standard within an organization. For example, different functions while encountering an error condition should either return a 0 or 1 consistently.

The following are some representative coding guidelines recommended by many software development organizations.

1. **Do not use a coding style that is too clever or too difficult to understand:** Code should be easy to understand. Many inexperienced engineers actually take pride in writing cryptic and incomprehensible code. Clever coding can obscure meaning of the code and hamper understanding. It also makes maintenance difficult.
2. **Avoid obscure side effects:** The side effects of a function call include modification of parameters passed by reference, modification of global variables, and I/O operations. An obscure side effect is one that is not obvious from a casual examination of the code. Obscure side effects make it difficult to understand a piece of code. For example, if a global variable is changed obscurely in a called module or some file I/O is performed which is difficult to infer from the function's name and header information, it becomes difficult for anybody trying to understand the code.
3. **Do not use an identifier for multiple purposes:** Programmers often use the same identifier to denote several temporary entities. For example, some programmers use a temporary loop variable for computing and a storing the final result. The rationale that is usually given by these programmers for such multiple uses of variables is memory efficiency, e.g. three variables use up three memory locations, whereas the same variable used in three different ways uses just one memory location. However, there are several things wrong with this approach and hence should be avoided. Some of the problems caused by use of variables for multiple purposes as follows:

- Each variable should be given a descriptive name indicating its purpose. This is not possible if an identifier is used for multiple purposes. Use of a variable for multiple purposes can lead to confusion and make it difficult for somebody trying to read and understand the code.
- Use of variables for multiple purposes usually makes future enhancements more difficult.

4.The code should be well-documented: As a rule of thumb, there must be at least one comment line on the average for every three-source line.

5.The length of any function should not exceed 10 source lines: A function that is very lengthy is usually very difficult to understand as it probably carries out many different functions. For the same reason, lengthy functions are likely to have disproportionately larger number of bugs.

6.Do not use goto statements: Use of goto statements makes a program unstructured and very difficult to understand.

Code Review

Code review for a model is carried out after the module is successfully compiled and the all the syntax errors have been eliminated. Code reviews are extremely cost-effective strategies for reduction in coding errors and to produce high quality code. Normally, two types of reviews are carried out on the code of a module. These two types code review techniques are code inspection and code walk through.

Code Walk Throughs

Code walk through is an informal code analysis technique. In this technique, after a module has been coded, successfully compiled and all syntax errors eliminated. A few members of the development team are given the code few days before the walk through meeting to read and understand code. Each member selects some test cases and simulates execution of the code by hand (i.e. trace execution through each statement and function execution). The main objectives of the walk through are to discover the algorithmic and logical errors in the code. The members note down their findings to discuss these in a walk through meeting where the coder of the module is present. Even though a code walk through is an informal analysis technique, several guidelines have evolved over the years for making this naïve but useful analysis technique more effective. Of course, these guidelines are based on personal experience, common sense, and several subjective factors. Therefore, these guidelines should be considered as examples rather than accepted as rules to be applied dogmatically. Some of these guidelines are the following:

- The team performing code walk through should not be either too big or too small. Ideally, it should consist of between three to seven members.
- Discussion should focus on discovery of errors and not on how to fix the discovered errors.
- In order to foster cooperation and to avoid the feeling among engineers that they are being evaluated in the code walk through meeting, managers should not attend the walk through meetings.

Code Inspection

In contrast to code walk through, the aim of code inspection is to discover some common types of errors caused due to oversight and improper programming. In other words, during code inspection the code is examined for the presence of certain kinds of errors, in contrast to the hand simulation of code execution done in code walk throughs. For instance, consider the classical error of writing a procedure that modifies a formal parameter while the calling routine calls that procedure with a constant actual parameter. It is more likely that such an error will be discovered by looking for these kinds of mistakes in the code, rather than by simply hand simulating execution of the procedure. In addition to the commonly made errors, adherence to coding standards is also checked during code inspection. Good software development companies collect statistics regarding different types of errors commonly committed by their engineers and identify the type of errors most frequently committed. Such a list of commonly committed errors can be used during code inspection to look out for possible errors. Following is a list of some classical programming errors which can be checked during code inspection:

- Use of uninitialized variables.
- Jumps into loops.
- Nonterminating loops.
- Incompatible assignments.
- Array indices out of bounds.
- Improper storage allocation and deallocation.
- Mismatches between actual and formal parameter in procedure calls.
- Use of incorrect logical operators or incorrect precedence among operators.
- Improper modification of loop variables.
- Comparison of equality of floating point variables, etc.

Clean Room Testing

Clean room testing was pioneered by IBM. This type of testing relies heavily on walk throughs, inspection, and formal verification. The programmers are not allowed to test any of their code by executing the code other than doing some syntax testing using a compiler. The software development philosophy is based on avoiding software defects by using a rigorous inspection process. The objective of this software is zero-defect software. The name 'clean room' was derived from the analogy with semi-conductor fabrication units. In these units (clean rooms), defects are avoided by manufacturing in ultra-clean atmosphere. In this kind of development, inspections to check the consistency of the components with their specifications has replaced unit-testing. This technique reportedly produces documentation and code that is more reliable and maintainable than other development methods relying heavily on code execution-based testing.

The clean room approach to software development is based on five characteristics:

- **Formal specification:** The software to be developed is formally specified. A state-transition model which shows system responses to stimuli is used to express the specification.
- **Incremental development:** The software is partitioned into increments which are developed and validated separately using the clean room process. These increments are specified, with customer input, at an early stage in the process.

- **Structured programming:** Only a limited number of control and data abstraction constructs are used. The program development process is process of stepwise refinement of the specification.
- **Static verification:** The developed software is statically verified using rigorous software inspections. There is no unit or module testing process for code components
- **Statistical testing of the system:** The integrated software increment is tested statistically to determine its reliability. These statistical tests are based on the operational profile which is developed in parallel with the system specification. The main problem with this approach is that testing effort is increased as walk throughs, inspection, and verification are time-consuming.

Software Documentation

When various kinds of software products are developed then not only the executable files and the source code are developed but also various kinds of documents such as users' manual, software requirements specification (SRS) documents, design documents, test documents, installation manual, etc are also developed as part of any software engineering process. All these documents are a vital part of good software development practice. Good documents are very useful and server the following purposes:

1. Good documents enhance understandability and maintainability of a software product. They reduce the effort and time required for maintenance.
2. Use documents help the users in effectively using the system.
3. Good documents help in effectively handling the manpower turnover problem. Even when an engineer leaves the organization, and a new engineer comes in, he can build up the required knowledge easily.
4. Production of good documents helps the manager in effectively tracking the progress of the project. The project manager knows that measurable progress is achieved if a piece of work is done and the required documents have been produced and reviewed.

Different types of software documents can broadly be classified into the following:

- Internal documentation
- External documentation

Internal documentation is the code comprehension features provided as part of the source code itself. Internal documentation is provided through appropriate module headers and comments.

The important types of internal documentation are the following:

- Comments embedded in the source code.
- Use of meaningful variable names.
- Module and function headers.
- Code indentation, Code structuring (i.e., code decomposed into modules and functions).
- Use of enumerated types.
- Use of constant identifiers.

- Use of user-defined data types.

embedded in the source code. Internal documentation is also provided through the useful variable names, module and function headers, code indentation, code structuring, use of enumerated types and constant identifiers, use of user-defined data types, etc. Careful experiments suggest that out of all types of internal documentation meaningful variable names is most useful in understanding the code. This is of course in contrast to the common expectation that code commenting would be the most useful. The research finding is obviously true when comments are written without thought. For example, the following style of code commenting does not in any way help in understanding the code.

a = 10; /* a made 10 */

- But even when code is carefully commented, meaningful variable names still are more helpful in understanding a piece of code. Good software development organizations usually ensure good internal documentation by appropriately formulating their coding standards and coding guidelines.

External documentation is provided through various types of supporting documents such as users' manual, software requirements specification document, design document, test documents, etc. A systematic software development style ensures that all these documents are produced in an orderly fashion.

Program Testing

Testing a program consists of providing the program with a set of test inputs (or test cases) and observing if the program behaves as expected. If the program fails to behave as expected, then the conditions under which failure occurs are noted for later debugging and correction.

Some commonly used terms associated with testing are:

- **Failure:** This is a manifestation of an error (or defect or bug). But, the mere presence of an error may not necessarily lead to a failure.
- **Test case:** This is the triplet [I,S,O], where I is the data input to the system, S is the state of the system at which the data is input, and O is the expected output of the system.
- **Test suite:** This is the set of all test cases with which a given software product is to be tested.

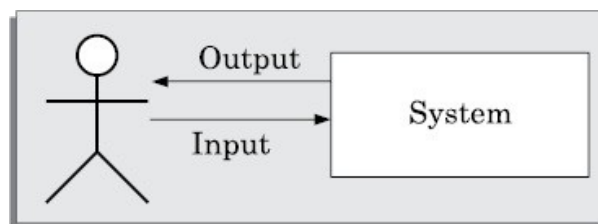


Figure: A simplified view of program testing.

Aim of testing

The aim of the testing process is to identify all defects existing in a software product. However for most practical systems, even after satisfactorily carrying out the testing phase, it is not possible to guarantee that the software is error free. This is because of the fact that the input data domain of most software products is very large. It is not practical to test the software exhaustively with respect to each value that the input data may assume. Even with this practical limitation of the testing process, the importance of testing should not be underestimated. It must be remembered that testing does expose many defects existing in a software product. Thus testing provides a practical way of reducing defects in a system and increasing the users' confidence in a developed system.

Differentiate between verification and validation.

- ➔ Verification is the process of determining whether the output of one phase of software development conforms to that of its previous phase, whereas validation is the process of determining whether a fully developed system conforms to its requirements specification. Thus while verification is concerned with phase containment of errors, the aim of validation is that the final product be error free.
- ➔ The primary techniques used for verification include review, simulation, formal verification, and testing. Review, simulation, and testing are usually considered as informal verification techniques.
- ➔ Verification does not require execution of the software, whereas validation requires execution of the software.
- ➔ Verification is carried out during the development process to check if the development activities are proceeding alright, whereas validation is carried out to check if the right as required by the customer has been developed.

Testing Activities

Testing involves performing the following main activities:

Test suite design: The set of test cases using which a program is to be tested is designed possibly using several test case design techniques. We discuss a few important test case design techniques later in this Chapter.

Running test cases and checking the results to detect failures: Each test case is run and the results are compared with the expected results. A mismatch between the actual result and expected results indicates a failure. The test cases for which the system fails are noted down for later debugging.

Locate error: In this activity, the failure symptoms are analysed to locate the errors. For each failure observed during the previous activity, the statements that are in error are identified.

Error correction: After the error is located during debugging, the code is appropriately changed to correct the error.

The testing activities have been shown schematically. As can be seen, the test cases are first designed, the test cases are run to detect failures. The bugs causing the failure are identified

through debugging, and the identified error is corrected. Of all the above mentioned testing activities, debugging often turns out to be the most time-consuming activity.

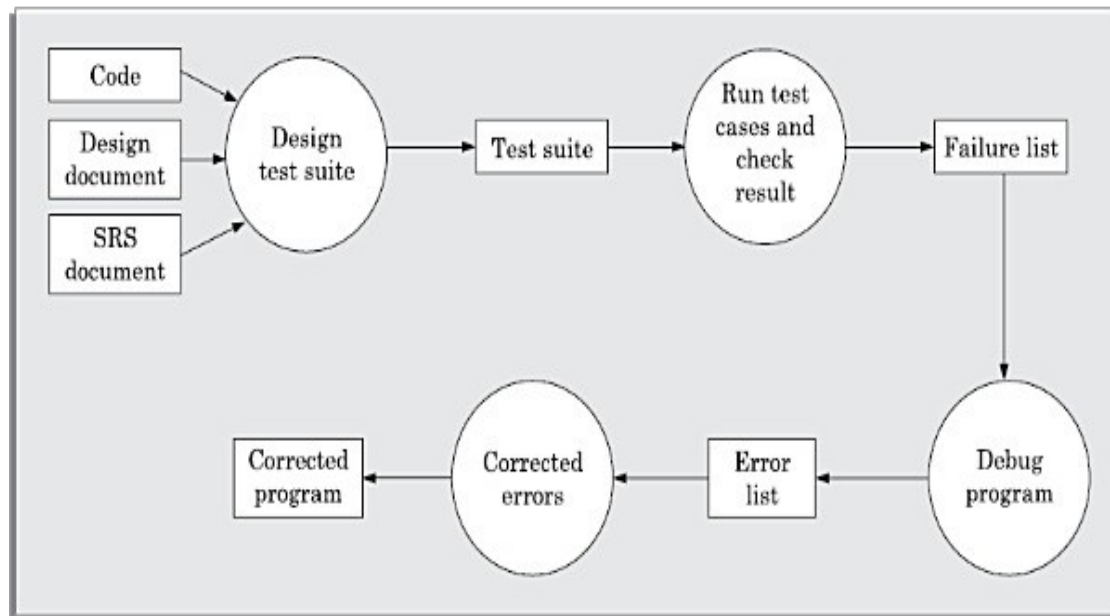


Figure: Testing process.

Design of test cases

Exhaustive testing of almost any non-trivial system is impractical due to the fact that the domain of input data values to most practical software systems is either extremely large or infinite. Therefore, we must design an optional test suite that is of reasonable size and can uncover as many errors existing in the system as possible. Actually, if test cases are selected randomly, many of these randomly selected test cases do not contribute to the significance of the test suite, i.e. they do not detect any additional defects not already being detected by other test cases in the suite. Thus, the number of random test cases in a test suite is, in general, not an indication of the effectiveness of the testing. In other words, testing a system using a large collection of test cases that are selected at random does not guarantee that all (or even most) of the errors in the system will be uncovered. Consider the following example code segment which finds the greater of two integer values x and y . This code segment has a simple programming error.

```

If (x>y)      max = x;
else         max = x;

```

For the above code segment, the test suite, $\{(x=3,y=2);(x=2,y=3)\}$ can detect the error, whereas a larger test suite $\{(x=3,y=2);(x=4,y=3);(x=5,y=1)\}$ does not detect the error. So, it would be incorrect to say that a larger test suite would always detect more errors than a smaller one, unless of course the larger test suite has also been carefully designed. This implies that the test suite should be carefully designed than picked randomly. Therefore, systematic approaches should be followed to design an optimal test suite. In an optimal test suite, each test case is designed to detect different errors.

Functional testing vs. Structural testing

In the black-box testing approach, test cases are designed using only the functional specification of the software, i.e. without any knowledge of the internal structure of the software. For this reason, black-box testing is known as functional testing.

On the other hand, in the white-box testing approach, designing test cases requires thorough knowledge about the internal structure of software, and therefore the white-box testing is called structural testing..

Testing in the large vs. testing in the small

Software products are normally tested first at the individual component (or unit) level. This is referred to as testing in the small. After testing all the components individually, the components are slowly integrated and tested at each level of integration (integration testing). Finally, the fully integrated system is tested (called system testing). Integration and system testing are known as testing in the large.

Unit testing

Unit testing is undertaken after a module has been coded and successfully reviewed. Unit testing (or module testing) is the testing of different units (or modules) of a system in isolation.

Driver and stub modules

In order to test a single module, a complete environment is needed to provide all that is necessary for execution of the module. That is, besides the module under test itself, the following steps are needed in order to be able to test the module:

- The procedures belonging to other modules that the module under test calls.
- Nonlocal data structures that the module accesses.
- A procedure to call the functions of the module under test with appropriate parameters.

Modules required to provide the necessary environment (which either call or are called by the module under test) is usually not available until they too have been unit tested, stubs and drivers are designed to provide the complete environment for a module.

Stub: The role of stub and driver modules is pictorially shown in fig. 10.1. A stub procedure is a dummy procedure that has the same I/O parameters as the given procedure but has a highly simplified behavior. For example, a stub procedure may produce the expected behavior using a simple table lookup mechanism.

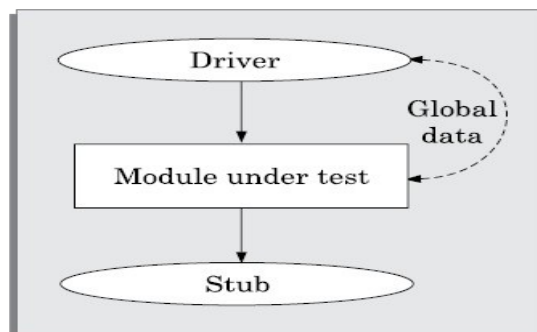


Figure: Unit testing with the help of driver and stub modules.

Driver: A driver module contains the nonlocal data structures accessed by the module under test, and would also have the code to call the different functions of the module with appropriate parameter values.

Black box testing

In the black-box testing, test cases are designed from an examination of the input/output values only and no knowledge of design, or code is required. The following are the two main approaches to designing black box test cases.

- Equivalence class partitioning
- Boundary value analysis

Equivalence Class Partitioning

In this approach, the domain of input values to a program is partitioned into a set of equivalence classes. This partitioning is done such that the behavior of the program is similar for every input data belonging to the same equivalence class. The main idea behind defining the equivalence classes is that testing the code with any one value belonging to an equivalence class is as good as testing the software with any other value belonging to that equivalence class. Equivalence classes for a software can be designed by examining the input data and output data.

The following are some general guidelines for designing the equivalence classes:

- 1.1 If the input data values to a system can be specified by a range of values, then one valid and two invalid equivalence classes should be defined.
- 1.2 If the input data assumes values from a set of discrete members of some domain, then one equivalence class for valid input values and another equivalence class for invalid input values should be defined.

Example#1: For a software that computes the square root of an input integer which can assume values in the range of 0 to 5000, there are three equivalence classes: The set of negative integers, the set of integers in the range of 0 and 5000, and the integers larger than 5000. Therefore, the test cases must include representatives for each of the three equivalence classes and a possible test set can be: {-5, 500, 6000}.

Example#2: Design the black-box test suite for the following program. The program computes the intersection point of two straight lines and displays the result. It reads two integer pairs (m_1 , c_1) and (m_2 , c_2) defining the two straight lines of the form $y = mx + c$.

The equivalence classes are the following:

- Parallel lines ($m_1 = m_2$, $c_1 \neq c_2$)
- Intersecting lines ($m_1 \neq m_2$)
- Coincident lines ($m_1 = m_2$, $c_1 = c_2$)

Now, selecting one representative value from each equivalence class, the test suit (2, 2) (2, 5), (5, 5) (7, 7), (10, 10) (10, 10) are obtained.

Example#3: Design equivalence class partitioning test suite for a function that reads a character string of size less than five characters and displays whether it is a palindrome.

Answer: The equivalence classes are the leaf level classes shown in Figure. The equivalence classes are palindromes, non-palindromes, and invalid inputs. Now, selecting one representative value from each equivalence class, we have the required test suite: {abc, aba, abcdef}.

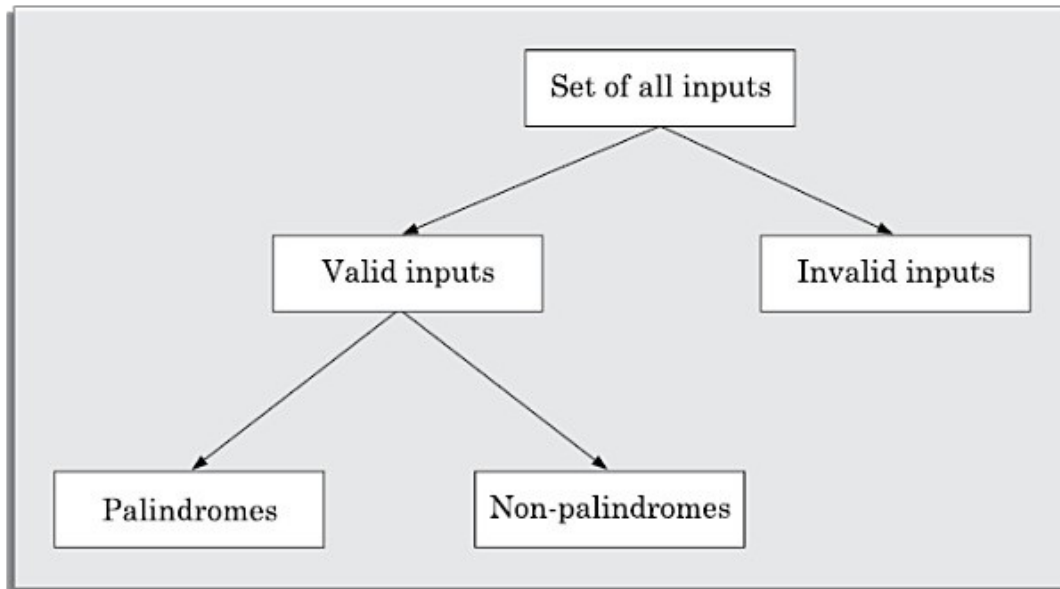


Figure: Equivalence classes for Example

Boundary Value Analysis

A type of programming error frequently occurs at the boundaries of different equivalence classes of inputs. The reason behind such errors might purely be due to psychological factors. Programmers often fail to see the special processing required by the input values that lie at the boundary of the different equivalence classes. For example, programmers may improperly use $<$ instead of \leq , or conversely \leq for $<$. Boundary value analysis leads to selection of test cases at the boundaries of the different equivalence classes.

Example: For a function that computes the square root of the integer values in the range of 0 and 5000, determine the boundary value test suite.

Answer: There are three equivalence classes—The set of negative integers, the set of integers in the range of 0 and 5000, and the set of integers larger than 5000. The boundary value-based test suite is: {0, -1, 5000, 5001}.

White box testing

One white-box testing strategy is said to be *stronger than* another strategy, if all types of errors detected by the first testing strategy is also detected by the second testing strategy, and the second testing strategy additionally detects some more types of errors. When two testing strategies detect errors that are different at least with respect to some types of errors, then they are called *complementary*. The concepts of stronger and complementary testing are schematically illustrated.

A white-box testing strategy can either be coverage-based or fault-based.

Fault-based testing

A fault-based testing strategy targets to detect certain types of faults. These faults that a test strategy focuses on constitutes the **fault model** of the strategy. An example of a fault-based strategy is mutation testing, which is discussed later in this section.

Coverage-based testing

A coverage-based testing strategy attempts to execute (or cover) certain elements of a program. Popular examples of coverage-based testing strategies are statement coverage, branch coverage, multiple condition coverage, and path coverage-based testing.

Testing criterion for coverage-based testing

A coverage-based testing strategy typically targets to execute (i.e., cover) certain program elements for discovering failures

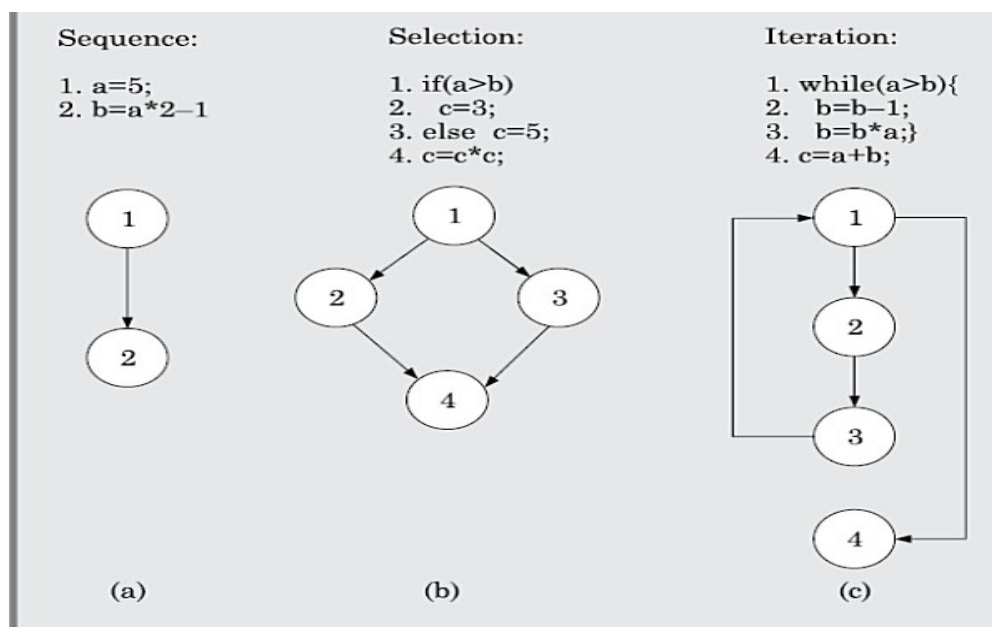


Figure: Illustration of stronger, weaker, and complementary testing strategies.

Statement coverage

The statement coverage strategy aims to design test cases so that every statement in a program is executed at least once. The principal idea governing the statement coverage strategy is that unless a statement is executed, it is very hard to determine if an error exists in that statement. Unless a statement is executed, it is very difficult to observe whether it causes failure due to some illegal memory access, wrong result computation, etc. However, executing some statement once and observing that it behaves properly for that input value is no guarantee that it will behave correctly for all input values. In the following, designing of test cases using the statement coverage strategy have been shown.

Example: Consider the Euclid's GCD computation algorithm: int

```
compute_gcd(x, y)
  int x, y;
  {
    1  while (x != y){
    2    if (x > y) then
    3      x = x - y;
    4    else y = y - x;
    5  }
    6  return x;
  }
```

By choosing the test set $\{(x=3, y=3), (x=4, y=3), (x=3, y=4)\}$, we can exercise the program such that all statements are executed at least once.

Branch coverage

In the branch coverage-based testing strategy, test cases are designed to make each branch condition to assume true and false values in turn. Branch testing is also known as edge testing as in this testing scheme, each edge of a program's control flow graph is traversed at least once.

It is obvious that branch testing guarantees statement coverage and thus is a stronger testing strategy compared to the statement coverage-based testing. For Euclid's GCD computation algorithm, the test cases for branch coverage can be $\{(x=3, y=3), (x=3, y=2), (x=4, y=3), (x=3, y=4)\}$.

Condition coverage

In this structural testing, test cases are designed to make each component of a composite conditional expression to assume both true and false values. For example, in the conditional expression $((c1.and.c2).or.c3)$, the components $c1$, $c2$ and $c3$ are each made to assume both true and false values. Branch testing is probably the simplest condition testing strategy where only the compound conditions appearing in the different branch statements are made to assume the true and false values. Thus, condition testing is a stronger testing strategy than branch testing and branch testing is stronger testing strategy than the statement coverage-based testing. For a composite conditional expression of n components, for condition coverage, 2^n test cases are required. Thus, for condition coverage, the number of test cases increases exponentially with the number of component conditions. Therefore, a condition coverage-based testing technique is practical only if n (the number of conditions) is small.

Path coverage

The path coverage-based testing strategy requires us to design test cases such that all linearly independent paths in the program are executed at least once. A linearly independent path can be defined in terms of the control flow graph (CFG) of a program.

Control Flow Graph (CFG)

A control flow graph describes the sequence in which the different instructions of a program get executed. In other words, a control flow graph describes how the control flows through the program. In order to draw the control flow graph of a program, all the statements of a program must be numbered first. The different numbered statements serve as nodes of the control flow graph (as shown in fig. 10.3). An edge from one node to another node exists if the execution of the statement representing the first node can result in the transfer of control to the other node.

The CFG for any program can be easily drawn by knowing how to represent the sequence, selection, and iteration type of statements in the CFG. After all, a program is made up from these types of statements. Fig. 10.3 summarizes how the CFG for these three types of statements can be drawn. It is important to note that for the iteration type of constructs such as the while construct, the loop condition is tested only at the beginning of the loop and therefore the control flow from the last statement of the loop is always to the top of the loop. Using these basic ideas, the CFG of Euclid's GCD computation algorithm.

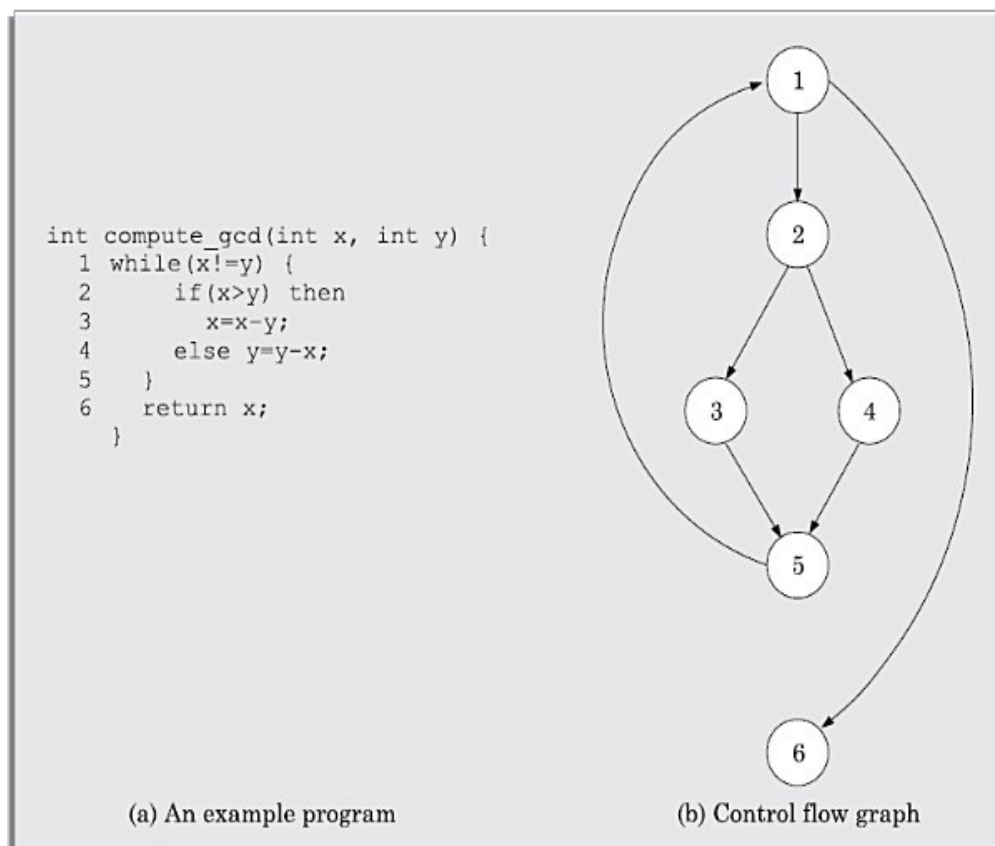


Figure: Control flow diagram of an example program.

Path

A path through a program is a node and edge sequence from the starting node to a terminal node of the control flow graph of a program. There can be more than one terminal node in a program. Writing test cases to cover all the paths of a typical program is impractical. For this reason, the path-coverage testing does not require coverage of all paths but only coverage of linearly independent paths.

Linearly independent path

A linearly independent path is any path through the program that introduces at least one new edge that is not included in any other linearly independent paths. If a path has one new node compared to all other linearly independent paths, then the path is also linearly independent. This is because, any path having a new node automatically implies that it has a new edge. Thus, a path that is subpath of another path is not considered to be a linearly independent path.

In order to understand the path coverage-based testing strategy, it is very much necessary to understand the control flow graph (CFG) of a program. Control flow graph (CFG) of a program has been discussed earlier.

Cyclomatic complexity

For more complicated programs it is not easy to determine the number of independent paths of the program. McCabe's cyclomatic complexity defines an upper bound for the number of linearly independent paths through a program. Also, the McCabe's cyclomatic complexity is very simple to compute. Thus, the McCabe's cyclomatic complexity metric provides a practical way of determining the maximum number of linearly independent paths in a program. Though the McCabe's metric does not directly identify the linearly independent paths, but it informs approximately how many paths to look for.

There are three different ways to compute the cyclomatic complexity. The answers computed by the three methods are guaranteed to agree.

Method 1:

Given a control flow graph G of a program, the cyclomatic complexity $V(G)$ can be computed as:

$$V(G) = E - N + 2$$

where N is the number of nodes of the control flow graph and E is the number of edges in the control flow graph.

For the CFG of example shown in fig. 10.4, $E=7$ and $N=6$. Therefore, the cyclomatic complexity = $7-6+2 = 3$.

Method 2:

An alternative way of computing the cyclomatic complexity of a program from an inspection of its control flow graph is as follows:

$$V(G) = \text{Total number of bounded areas} + 1$$

In the program's control flow graph G , any region enclosed by nodes and edges can be called as a bounded area. This is an easy way to determine the McCabe's cyclomatic complexity. But, what if the graph G is not planar, i.e. however you draw the graph, two or more edges intersect? Actually, it can be shown that structured programs always yield planar graphs. But, presence of GOTO's can easily add intersecting edges. Therefore, for non-structured programs, this way of computing the McCabe's cyclomatic complexity cannot be used.

The number of bounded areas increases with the number of decision paths and loops. Therefore, the McCabe's metric provides a quantitative measure of testing difficulty and the ultimate reliability. For the CFG example shown in fig. 10.4, from a visual examination of the CFG the number of bounded areas is 2. Therefore the cyclomatic complexity, computing with this method is also $2+1 = 3$. This method provides a very easy way of computing the cyclomatic complexity of CFGs, just from a visual examination of the CFG. On the other hand, the other method of computing CFGs is more amenable to automation, i.e. it can be easily coded into a program which can be used to determine the cyclomatic complexities of arbitrary CFGs.

Method 3:

The cyclomatic complexity of a program can also be easily computed by computing the number of decision statements of the program. If N is the number of decision statement of a program, then the McCabe's metric is equal to $N+1$.

Data flow-based testing

Data flow-based testing method selects test paths of a program according to the locations of the definitions and uses of different variables in a program.

For a statement numbered S , let

DEF(S) = {X/statement S contains a definition of X}, and
USES(S) = {X/statement S contains a use of X}

For the statement $S: a=b+c; ,$ DEF(S) = {a}. USES(S) = {b, c}. The definition of variable X at statement S is said to be live at statement $S1$, if there exists a path from statement S to statement $S1$ which does not contain any definition of X .

The *definition-use chain* (or DU chain) of a variable X is of form $[X, S, S1]$, where S and $S1$ are statement numbers, such that $X \in \text{DEF}(S)$ and $X \in \text{USES}(S1)$, and the definition of X in the statement S is live at statement $S1$. One simple data flow testing strategy is to require that every DU chain be covered at least once. Data flow testing strategies are useful for selecting test paths of a program containing nested if and loop statements.

Mutation testing

In mutation testing, the software is first tested by using an initial test suite built up from the different white box testing strategies. After the initial testing is complete, mutation testing is taken up. The idea behind mutation testing is to make few arbitrary changes to a program at a time. Each time the program is changed, it is called as a mutated program and the change effected is called as a mutant. A mutated program is tested against the full test suite of the program. If there exists at least one test case in the test suite for which a mutant gives an incorrect result, then the mutant is said to be dead. If a mutant remains alive even after all the test cases have been exhausted, the test data is enhanced to kill the mutant.

The process of generation and killing of mutants can be automated by predefining a set of primitive changes that can be applied to the program. These primitive changes can be alterations such as changing an arithmetic operator, changing the value of a constant, changing a data type, etc. A major disadvantage of the mutation-based testing approach is that it is computationally very expensive, since a large number of possible mutants can be generated.

Since mutation testing generates a large number of mutants and requires us to check each mutant with the full test suite, it is not suitable for manual testing. Mutation testing should be used in conjunction of some testing tool which would run all the test cases automatically.

Need for debugging

Once errors are identified in a program code, it is necessary to first identify the precise program statements responsible for the errors and then to fix them. Identifying errors in a program code and then fix them up are known as debugging.

Debugging approaches

The following are some of the approaches popularly adopted by programmers for debugging.

Brute Force Method:

This is the most common method of debugging but is the least efficient method. In this approach, the program is loaded with print statements to print the intermediate values with the hope that some of the printed values will help to identify the statement in error. This approach becomes more systematic with the use of a symbolic debugger (also called a source code debugger), because values of different variables can be easily checked and break points and watch points can be easily set to test the values of variables effortlessly.

Backtracking:

This is also a fairly common approach. In this approach, beginning from the statement at which an error symptom has been observed, the source code is traced backwards until the error is discovered. Unfortunately, as the number of source lines to be traced back increases, the number of potential backward paths increases and may become unmanageably large thus limiting the use of this approach.

Cause Elimination Method:

In this approach, a list of causes which could possibly have contributed to the error symptom is developed and tests are conducted to eliminate each. A related technique of identification of the error from the error symptom is the software fault tree analysis.

Program Slicing:

This technique is similar to back tracking. Here the search space is reduced by defining slices. A slice of a program for a particular variable at a particular statement is the set of source lines preceding this statement that can influence the value of that variable [Mund2002].

Debugging guidelines

Debugging is often carried out by programmers based on their ingenuity. The following are some general guidelines for effective debugging:

- Many times debugging requires a thorough understanding of the program design. Trying to debug based on a partial understanding of the system design and implementation may require an inordinate amount of effort to be put into debugging even simple problems.
- Debugging may sometimes even require full redesign of the system. In such cases, a common mistake that novice programmers often make is attempting not to fix the error but its symptoms.
- One must be beware of the possibility that an error correction may introduce new errors. Therefore after every round of error-fixing, regression testing must be carried out.

Program analysis tools

A program analysis tool means an automated tool that takes the source code or the executable code of a program as input and produces reports regarding several important characteristics of the program, such as its size, complexity, adequacy of commenting, adherence to programming standards, etc. We can classify these into two broad categories of program analysis tools:

- Static Analysis tools
- Dynamic Analysis tools

Static program analysis tools

Static analysis tool is also a program analysis tool. It assesses and computes various characteristics of a software product without executing it. Typically, static analysis tools analyze some structural representation of a program to arrive at certain analytical conclusions, e.g. that some structural properties hold. The structural properties that are usually analyzed are:

- Whether the coding standards have been adhered to?
- Certain programming errors such as uninitialized variables and mismatch between actual and formal parameters, variables that are declared but never used are also checked.

Code walk throughs and code inspections might be considered as static analysis methods. But, the term static program analysis is used to denote automated analysis tools. So, a compiler can be considered to be a static program analysis tool.

Dynamic program analysis tools

Dynamic program analysis techniques require the program to be executed and its actual behavior recorded. A dynamic analyzer usually instruments the code (i.e. adds additional statements in the source code to collect program execution traces). The instrumented code when executed allows us to record the behavior of the software for different test cases. After the software has been tested with its full test suite and its behavior recorded, the dynamic analysis tool carries out a post execution analysis and produces reports which describe the structural coverage that has been achieved by the complete test suite for the program. For example, the post execution dynamic analysis report might provide data on extent statement, branch and path coverage achieved.

Normally the dynamic analysis results are reported in the form of a histogram or a pie chart to describe the structural coverage achieved for different modules of the program. The output of a dynamic analysis tool can be stored and printed easily and provides evidence that thorough testing has been done. The dynamic analysis results the extent of testing performed in white-box mode. If the testing coverage is not satisfactory more test cases can be designed and added to the test suite. Further, dynamic analysis results can help to eliminate redundant test cases from the test suite.

Integration testing

The primary objective of integration testing is to test the module interfaces, i.e. there are no errors in the parameter passing, when one module invokes another module. During integration testing, different modules of a system are integrated in a planned manner using an integration plan. The integration plan specifies the steps and the order in which modules are combined to realize the full system. After each integration step, the partially integrated system is tested. An important factor that guides the integration plan is the module dependency graph. The structure chart (or module dependency graph) denotes the order in which different modules call each other. By examining the structure chart the integration plan can be developed.

Integration test approaches

There are four types of integration testing approaches. Any one (or a mixture) of the following approaches can be used to develop the integration test plan. Those approaches are the following:

- Big bang approach
- Top-down approach
- Bottom-up approach
- Mixed-approach

Big-Bang Integration Testing

It is the simplest integration testing approach, where all the modules making up a system are integrated in a single step. In simple words, all the modules of the system are simply put together and tested. However, this technique is practicable only for very small systems. The main problem with this approach is that once an error is found during the integration testing, it is very difficult to localize the error as the error may potentially belong to any of the modules being integrated. Therefore, debugging errors reported during big bang integration testing are very expensive to fix.

Bottom-Up Integration Testing

In bottom-up testing, each subsystem is tested separately and then the full system is tested. A subsystem might consist of many modules which communicate among each other through well-defined interfaces. The primary purpose of testing each subsystem is to test the interfaces among various modules making up the subsystem. Both control and data interfaces are tested. The test cases must be carefully chosen to exercise the interfaces in all possible manners.

Large software systems normally require several levels of subsystem testing; lower-level subsystems are successively combined to form higher-level subsystems. A principal advantage of bottom-up integration testing is that several disjoint subsystems can be tested simultaneously. In a pure bottom-up testing no stubs are required, only test-drivers are required. A disadvantage of bottom-up testing is the complexity that occurs when the system is made up of a large number of small subsystems. The extreme case corresponds to the big-bang approach.

Top-Down Integration Testing

Top-down integration testing starts with the main routine and one or two subordinate routines in the system. After the top-level 'skeleton' has been tested, the immediately subroutines of the 'skeleton' are combined with it and tested. Top-down integration testing approach requires the use of program stubs to simulate the effect of lower-level routines that are called by the routines under test. A pure top-down integration does not require any driver routines. A disadvantage of the top-down integration testing approach is that in the absence of lower-level routines, many times it may become difficult to exercise the top-level routines in the desired manner since the lower-level routines perform several low-level functions such as I/O.

Mixed Integration Testing

A mixed (also called sandwiched) integration testing follows a combination of top-down and bottom-up testing approaches. In top-down approach, testing can start only after the top-level modules have been coded and unit tested. Similarly, bottom-up testing can start only after the bottom level modules are ready. The mixed approach overcomes this shortcoming of the top-down and bottom-up approaches. In the mixed testing approaches, testing can start as and when modules become available. Therefore, this is one of the most commonly used integration testing approaches.

Phased vs. incremental testing

The different integration testing strategies are either phased or incremental. A comparison of these two strategies is as follows:

- In incremental integration testing, only one new module is added to the partial system each time.
- In phased integration, a group of related modules are added to the partial system each time.

Phased integration requires less number of integration steps compared to the incremental integration approach. However, when failures are detected, it is easier to debug the system in the incremental testing approach since it is known that the error is caused by addition of a single module. In fact, [big bang testing](#) is a degenerate case of the phased integration testing approach.

Grey-Box Testing of Object-oriented Programs

As we have already mentioned, model-based testing is important for object-oriented programs, as these test cases help detect bugs that are specific to the object-orientation constructs.

The following are some important types of grey-box testing that can be carried on based on UML models:

State-model-based testing

State coverage: Each method of an object are tested at each state of the object.

State transition coverage: It is tested whether all transitions depicted in the state model work satisfactorily.

State transition path coverage: All transition paths in the state model are tested.

Use case-based testing

Scenario coverage: Each use case typically consists of a mainline scenario and several alternate scenarios. For each use case, the mainline and all alternate sequences are tested to check if any errors show up.

Class diagram-based testing

Testing derived classes: All derived classes of the base class have to be instantiated and tested. In addition to testing the new methods defined in the derived class, the inherited methods must be retested.

Association testing: All association relations are tested.

Aggregation testing: Various aggregate objects are created and tested.

Sequence diagram-based testing Method coverage: All methods depicted in the sequence diagrams are covered.

Message path coverage: All message paths that can be constructed from the sequence diagrams are covered.

Integration Testing of Object-oriented Programs

There are two main approaches to integration testing of object-oriented programs:

- Thread-based
- Use based

Thread-based approach: In this approach, all classes that need to collaborate to realise the behaviour of a single use case are integrated and tested. After all the required classes for a use case are integrated and tested, another use case is taken up and other classes (if any) necessary for execution of the second use case to run are integrated and tested. This is continued till all use cases have been considered.

Use-based approach: Use-based integration begins by testing classes that either need no service from other classes or need services from at most a few other classes. After these classes have been integrated and tested, classes that use the services from the already integrated classes are integrated and tested. This is continued till all the classes have been integrated and tested.

System testing

System tests are designed to validate a fully developed system to assure that it meets its requirements. There are essentially three main kinds of system testing:

- **Alpha Testing.** Alpha testing refers to the system testing carried out by the test team within the developing organization.
- **Beta testing.** Beta testing is the system testing performed by a select group of friendly customers.

- **Acceptance Testing.** Acceptance testing is the system testing performed by the customer to determine whether he should accept the delivery of the system.

In each of the above types of tests, various kinds of test cases are designed by referring to the SRS document. Broadly, these tests can be classified into functionality and performance tests. The functionality tests test the functionality of the software to check whether it satisfies the functional requirements as documented in the SRS document. The performance tests test the conformance of the system with the nonfunctional requirements of the system.

Smoke Testing

Smoke testing is carried out before initiating system testing to ensure that system testing would be meaningful, or whether many parts of the software would fail. The idea behind smoke testing is that if the integrated program cannot pass even the basic tests, it is not ready for a vigorous testing. For smoke testing, a few test cases are designed to check whether the basic functionalities are working. For example, for a library automation system, the smoke tests may check whether books can be created and deleted, whether member records can be created and deleted, and whether books can be loaned and returned.

Performance testing

Performance testing is carried out to check whether the system needs the non-functional requirements identified in the SRS document. There are several types of performance testing. Among of them nine types are discussed below. The types of performance testing to be carried out on a system depend on the different non-functional requirements of the system documented in the SRS document. All performance tests can be considered as black-box tests.

- Stress testing
- Volume testing
- Configuration testing
- Compatibility testing
- Regression testing
- Recovery testing
- Maintenance testing
- Documentation testing
- Usability testing

Stress Testing

Stress testing is also known as endurance testing. Stress testing evaluates system performance when it is stressed for short periods of time. Stress tests are black box tests which are designed to impose a range of abnormal and even illegal input conditions so as to stress the capabilities of the software. Input data volume, input data rate, processing time, utilization of memory, etc. are tested beyond the designed capacity. For example, suppose an operating system is supposed to support 15 multiprogrammed jobs, the system is stressed by attempting to run 15 or more jobs simultaneously. A real-time system might be tested to determine the effect of simultaneous arrival of several high-priority interrupts.

Stress testing is especially important for systems that usually operate below the maximum capacity but are severely stressed at some peak demand hours. For example, if the non-functional requirement specification states that the response time should not be more than

20 secs per transaction when 60 concurrent users are working, then during the stress testing the response time is checked with 60 users working simultaneously.

Volume Testing

It is especially important to check whether the data structures (arrays, queues, stacks, etc.) have been designed to successfully extraordinary situations. For example, a compiler might be tested to check whether the symbol table overflows when a very large program is compiled.

Configuration Testing

This is used to analyze system behavior in various hardware and software configurations specified in the requirements. Sometimes systems are built in variable configurations for different users. For instance, we might define a minimal system to serve a single user, and other extension configurations to serve additional users. The system is configured in each of the required configurations and it is checked if the system behaves correctly in all required configurations.

Compatibility Testing

This type of testing is required when the system interfaces with other types of systems. Compatibility aims to check whether the interface functions perform as required. For instance, if the system needs to communicate with a large database system to retrieve information, compatibility testing is required to test the speed and accuracy of data retrieval.

Regression Testing

This type of testing is required when the system being tested is an upgradation of an already existing system to fix some bugs or enhance functionality, performance, etc. Regression testing is the practice of running an old test suite after each change to the system or after each bug fix to ensure that no new bug has been introduced due to the change or the bug fix. However, if only a few statements are changed, then the entire test suite need not be run - only those test cases that test the functions that are likely to be affected by the change need to be run.

Recovery Testing

Recovery testing tests the response of the system to the presence of faults, or loss of power, devices, services, data, etc. The system is subjected to the loss of the mentioned resources (as applicable and discussed in the SRS document) and it is checked if the system recovers satisfactorily. For example, the printer can be disconnected to check if the system hangs. Or, the power may be shut down to check the extent of data loss and corruption.

Maintenance Testing

This testing addresses the diagnostic programs, and other procedures that are required to be developed to help maintenance of the system. It is verified that the artifacts exist and they perform properly.

Documentation Testing

It is checked that the required user manual, maintenance manuals, and technical manuals exist and are consistent. If the requirements specify the types of audience for which a specific manual should be designed, then the manual is checked for compliance.

Usability Testing

Usability testing concerns checking the user interface to see if it meets all user requirements concerning the user interface. During usability testing, the display screens, report formats, and other aspects relating to the user interface requirements are tested.

Error seeding

Sometimes the customer might specify the maximum number of allowable errors that may be present in the delivered system. These are often expressed in terms of maximum number of allowable errors per line of source code. Error seed can be used to estimate the number of residual errors in a system.

Error seeding, as the name implies, seeds the code with some known errors. In other words, some artificial errors are introduced into the program artificially. The number of these seeded errors detected in the course of the standard testing procedure is determined. These values in conjunction with the number of unseeded errors detected can be used to predict:

- The number of errors remaining in the product.
- The effectiveness of the testing strategy.

Let N be the total number of defects in the system and let n of these defects be found by testing.

Let S be the total number of seeded defects, and let s of these defects be found during testing.

$$n/N = s/S$$

or

$$N = S \times n/s$$

Defects still remaining after testing = $N - n = n \times (S - s) / s$

Error seeding works satisfactorily only if the kind of seeded errors matches closely with the kind of defects that actually exist. However, it is difficult to predict the types of errors that exist in a software. To some extent, the different categories of errors that remain can be estimated to a first approximation by analyzing historical data of similar projects. Due to the shortcoming that the types of seeded errors should match closely with the types of errors actually existing in the code, error seeding is useful only to a moderate extent.

Test documentation

A piece of documentation that is produced towards the end of testing is the test summary report. This report normally covers each subsystem and represents a summary of tests which have been applied to the subsystem and their outcome. It normally specifies the following:

- What is the total number of tests that were applied to a subsystem.
- Out of the total number of tests how many tests were successful.
- How many were unsuccessful, and the degree to which they were unsuccessful, e.g., whether a test was an outright failure or whether some of the expected results of the test were actually observed.

Software Reliability

Reliability of a software product essentially denotes its trustworthiness or dependability. Alternatively, reliability of a software product can also be defined as the probability of the product working “correctly” over a given period of time. It is obvious that a software product having a large number of defects is unreliable. It is also clear that the reliability of a system improves, if the number of defects in it is reduced. However, there is no simple relationship between the observed system reliability and the number of latent defects in the system. For example, removing errors from parts of a software which are rarely executed makes little difference to the perceived reliability of the product. It has been experimentally observed by analyzing the behavior of a large number of programs that 90% of the execution time of a typical program is spent in executing only 10% of the instructions in the program. These most used 10% instructions are often called the core of the program. The rest 90% of the program statements are called non-core and are executed only for 10% of the total execution time. It therefore may not be very surprising to note that removing 60% product defects from the least used parts of a system would typically lead to only 3% improvement to the product reliability. It is clear that the quantity by which the overall reliability of a program improves due to the correction of a single error depends on how frequently the corresponding instruction is executed.

Thus, reliability of a product depends not only on the number of latent errors but also on the exact location of the errors. Apart from this, reliability also depends upon how the product is used, i.e. on its execution profile. If it is selected input data to the system such that only the “correctly” implemented functions are executed, none of the errors will be exposed and the perceived reliability of the product will be high. On the other hand, if the input data is selected such that only those functions which contain errors are invoked, the perceived reliability of the system will be very low.

Reasons for software reliability being difficult to measure

The reasons why software reliability is difficult to measure can be summarized as follows:

- The reliability improvement due to fixing a single bug depends on where the bug is located in the code.
- The perceived reliability of a software product is highly observer-dependent.
- The reliability of a product keeps changing as errors are detected and fixed.

Hardware reliability vs. software reliability differs.

Reliability behavior for hardware and software are very different. For example, hardware failures are inherently different from software failures. Most hardware failures are due to component wear and tear. A logic gate may be stuck at 1 or 0, or a resistor might short circuit. To fix hardware faults, one has to either replace or repair the failed part. On the other hand, a software product would continue to fail until the error is tracked down and either the design or the code is changed. For this reason, when a hardware is repaired its reliability is maintained at the level that existed before the failure occurred; whereas when a software failure is repaired, the reliability may either increase or decrease (reliability may decrease if a bug introduces new errors). To put this fact in a different perspective, hardware reliability study is concerned with stability (for example, inter-failure times remain constant). On the other hand, software reliability study aims at reliability growth (i.e. inter-failure times increase). The change of failure rate over the product lifetime for a typical hardware and a software product are sketched in fig. 26.1. For hardware products, it can be observed that failure rate is high initially but decreases as the faulty components are identified and removed. The system then enters its useful life. After some time (called product life time) the components wear out, and the

failure rate increases. This gives the plot of hardware reliability over time its characteristics “bath tub” shape. On the other hand, for software the failure rate is at it’s highest during integration and test. As the system is tested, more and more errors are identified and removed resulting in reduced failure rate. This error removal continues at a slower pace during the useful life of the product. As the software becomes obsolete no error corrections occurs and the failure rate remains unchanged.

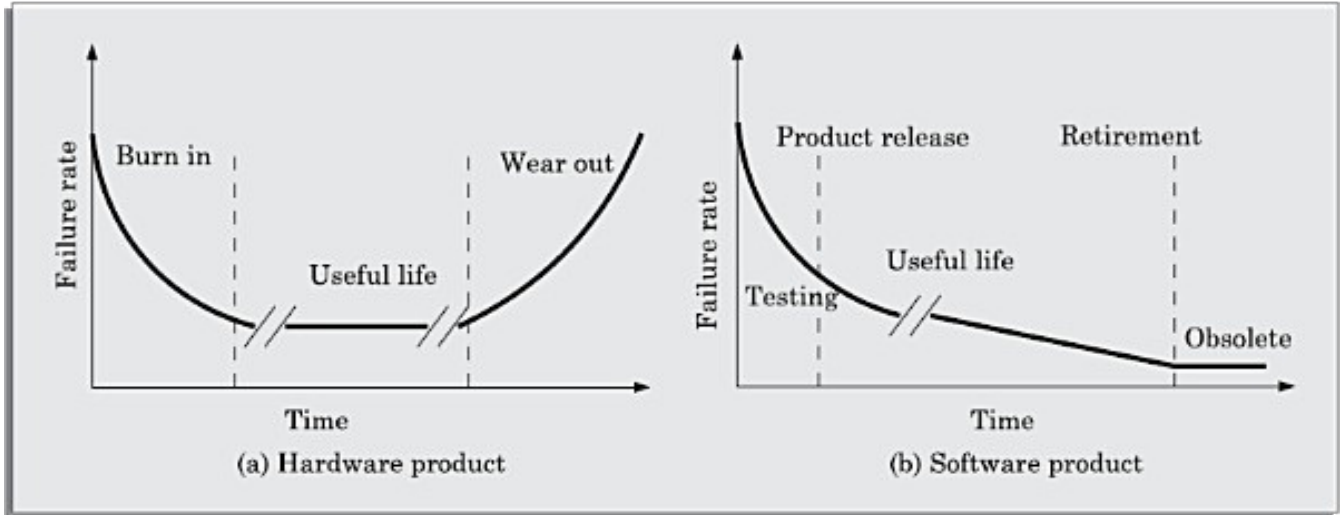


Fig: Change in failure rate of a product

Reliability Metrics

The reliability requirements for different categories of software products may be different. For this reason, it is necessary that the level of reliability required for a software product should be specified in the SRS (software requirements specification) document. In order to be able to do this, some metrics are needed to quantitatively express the reliability of a software product. A good reliability measure should be observer-dependent, so that different people can agree on the degree of reliability a system has. For example, there are precise techniques for measuring performance, which would result in obtaining the same performance value irrespective of who is carrying out the performance measurement. However, in practice, it is very difficult to formulate a precise reliability measurement technique. The next base case is to have measures that correlate with reliability. There are six reliability metrics which can be used to quantify the reliability of software products.

- **Rate of occurrence of failure (ROCOF)**- ROCOF measures the frequency of occurrence of unexpected behavior (i.e. failures). ROCOF measure of a software product can be obtained by observing the behavior of a software product in operation over a specified time interval and then recording the total number of failures occurring during the interval.
- **Mean Time To Failure (MTTF)** - MTTF is the average time between two successive failures, observed over a large number of failures. To measure MTTF, we can record the failure data for n failures. Let the failures occur at the time instants t_1, t_2, \dots, t_n . Then, MTTF can be calculated as

$$\sum_{i=1}^n \frac{t_{i+1} - t_i}{(n - 1)}$$

It is important to note that only run time is considered in the time measurements, i.e. the time for which the system is down to fix the error, the boot time, etc are not taken into account in the time measurements and the clock is stopped at these times.

- **Mean Time To Repair (MTTR)** - Once failure occurs, sometime is required to fix the error. MTTR measures the average time it takes to track the errors causing the failure and to fix them.
- **Mean Time Between Failure (MTBF)** - MTTF and MTTR can be combined to get the MTBF metric: $MTBF = MTTF + MTTR$. Thus, MTBF of 300 hours indicates that once a failure occurs, the next failure is expected after 300 hours. In this case, time measurements are real time and not the execution time as in MTTF.
- **Probability of Failure on Demand (POFOD)** - Unlike the other metrics discussed, this metric does not explicitly involve time measurements. POFOD measures the likelihood of the system failing when a service request is made. For example, a POFOD of 0.001 would mean that 1 out of every 1000 service requests would result in a failure.
- **Availability**- Availability of a system is a measure of how likely shall the system be available for use over a given period of time. This metric not only considers the number of failures occurring during a time interval, but also takes into account the repair time (down time) of a system when a failure occurs. This metric is important for systems such as telecommunication systems, and operating systems, which are supposed to be never down and where repair and restart time are significant and loss of service during that time is important.

Classification of software failures

A possible classification of failures of software products into five different types is as follows:

- **Transient**- Transient failures occur only for certain input values while invoking a function of the system.
- **Permanent**- Permanent failures occur for all input values while invoking a function of the system.
- **Recoverable**- When recoverable failures occur, the system recovers with or without operator intervention.
- **Unrecoverable**- In unrecoverable failures, the system may need to be restarted.
- **Cosmetic**- These classes of failures cause only minor irritations, and do not lead to incorrect results. An example of a cosmetic failure is the case where the mouse button has to be clicked twice instead of once to invoke a given function through the graphical user interface.

RELIABILITY GROWTH MODELS

A reliability growth model is a mathematical model of how software reliability improves as errors are detected and repaired. A reliability growth model can be used to predict when (or if at all) a particular level of reliability is likely to be attained. Thus, reliability growth modeling can be used to determine when to stop testing to attain a given reliability level. Although several different reliability growth models have been proposed, in this text we will discuss only two very simple reliability growth models.

Jelinski and Moranda Model -The simplest reliability growth model is a step function model where it is assumed that the reliability increases by a constant increment each time an error is detected and repaired. However, this simple model of reliability which implicitly assumes that all errors contribute equally to

reliability growth, is highly unrealistic since it is already known that correction of different types of errors contribute differently to reliability growth.

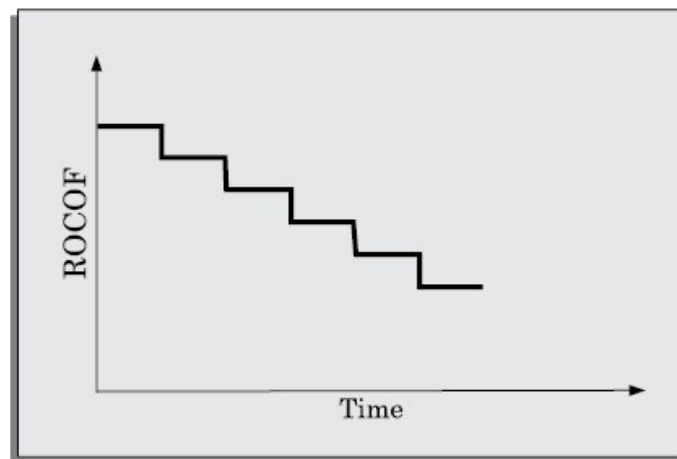


Fig: Step function model of reliability growth

Littlewood and Verall's Model -This model allows for negative reliability growth to reflect the fact that when a repair is carried out, it may introduce additional errors. It also models the fact that as errors are repaired, the average improvement in reliability per repair decreases. It treats an error's contribution to reliability improvement to be an independent random variable having Gamma distribution. This distribution models the fact that error corrections with large contributions to reliability growth are removed first. This represents diminishing return as test continues.

Statistical testing

Statistical testing is a testing process whose objective is to determine the reliability of software products rather than discovering errors. Test cases are designed for statistical testing with an entirely different objective than those of conventional testing.

Operation profile

Different categories of users may use a software for different purposes. For example, a Librarian might use the library automation software to create member records, add books to the library, etc. whereas a library member might use the software to query about the availability of the book, or to issue and return books. Formally, the operation profile of a software can be defined as the probability distribution of the input of an average user. If the input to a number of classes $\{C_i\}$ is divided, the probability value of a class represents the probability of an average user selecting his next input from this class. Thus, the operation profile assigns a probability value P_i to each input class C_i .

Steps in statistical testing

Statistical testing allows one to concentrate on testing those parts of the system that are most likely to be used. The first step of statistical testing is to determine the operation profile of the software. The next step is to generate a set of test data corresponding to the determined operation profile. The third step is to apply the test cases to the software and record the time between each failure. After a statistically significant number of failures have been observed, the reliability can be computed.

Advantages and disadvantages of statistical testing

Statistical testing allows one to concentrate on testing parts of the system that are most likely to be used. Therefore, it results in a system that the users to be more reliable (than actually it is!). Reliability estimation using statistical testing is more accurate compared to those of other methods such as ROCOF, POFOD etc. But it is not easy to perform statistical testing properly. There is no simple and repeatable way of defining operation profiles. Also it is very much cumbersome to generate test cases for statistical testing cause the number of test cases with which the system is to be tested should be statistically significant.

Software Quality

Traditionally, a quality product is defined in terms of its fitness of purpose. That is, a quality product does exactly what the users want it to do. For software products, fitness of purpose is usually interpreted in terms of satisfaction of the requirements laid down in the SRS document. Although “fitness of purpose” is a satisfactory definition of quality for many products such as a car, a table fan, a grinding machine, etc. – for software products, “fitness of purpose” is not a wholly satisfactory definition of quality. To give an example, consider a software product that is functionally correct. That is, it performs all functions as specified in the SRS document. But, has an almost unusable user interface. Even though it may be functionally correct, we cannot consider it to be a quality product. Another example may be that of a product which does everything that the users want but has an almost incomprehensible and unmaintainable code. Therefore, the traditional concept of quality as “fitness of purpose” for software products is not wholly satisfactory.

The modern view of a quality associates with a software product several quality factors such as the following:

- ◆ **Portability:** A software product is said to be portable, if it can be easily made to work in different operating system environments, in different machines, with other software products, etc.
- ◆ **Usability:** A software product has good usability, if different categories of users (i.e. both expert and novice users) can easily invoke the functions of the product.
- ◆ **Reusability:** A software product has good reusability, if different modules of the product can easily be reused to develop new products.
- ◆ **Correctness:** A software product is correct, if different requirements as specified in the SRS document have been correctly implemented.
- ◆ **Maintainability:** A software product is maintainable, if errors can be easily corrected as and when they show up, new functions can be easily added to the product, and the functionalities of the product can be easily modified, etc.

Software quality management system

A quality management system (often referred to as quality system) is the principal methodology used by organizations to ensure that the products they develop have the desired quality. A quality system consists of the following:

- ◆ **Managerial Structure and Individual Responsibilities.** A quality system is actually the responsibility of the organization as a whole. However, every organization has a separate quality department to perform several quality system activities. The quality system of an organization should have support of the top management. Without support

for the quality system at a high level in a company, few members of staff will take the quality system seriously.

◆ **Quality System Activities.** The quality system activities encompass the following:

- auditing of projects
- review of the quality system
- development of standards, procedures, and guidelines, etc.
- production of reports for the top management summarizing the effectiveness of the quality system in the organization.

Evolution of quality management system

Quality systems have rapidly evolved over the last five decades. Prior to World War II, the usual method to produce quality products was to inspect the finished products to eliminate defective products. Since that time, quality systems of organizations have undergone through four stages of evolution as shown in the figure. The initial product inspection method gave way to quality control (QC). Quality control focuses not only on detecting the defective products and eliminating them but also on determining the causes behind the defects. Thus, quality control aims at correcting the causes of errors and not just rejecting the products. The next breakthrough in quality systems was the development of quality assurance principles.

◆ The basic premise of modern quality assurance is that if an organization's processes are good and are followed rigorously, then the products are bound to be of good quality. The modern quality paradigm includes guidance for recognizing, defining, analyzing, and improving the production process. Total quality management (TQM) advocates that the process followed by an organization must be continuously improved through process measurements.

◆ TQM goes a step further than quality assurance and aims at continuous process improvement. TQM goes beyond documenting processes to optimizing them through redesign. A term related to TQM is Business Process Reengineering (BPR). BPR aims at reengineering the way business is carried out in an organization. From the above discussion it can be stated that over the years the quality paradigm has shifted from product assurance to process assurance.

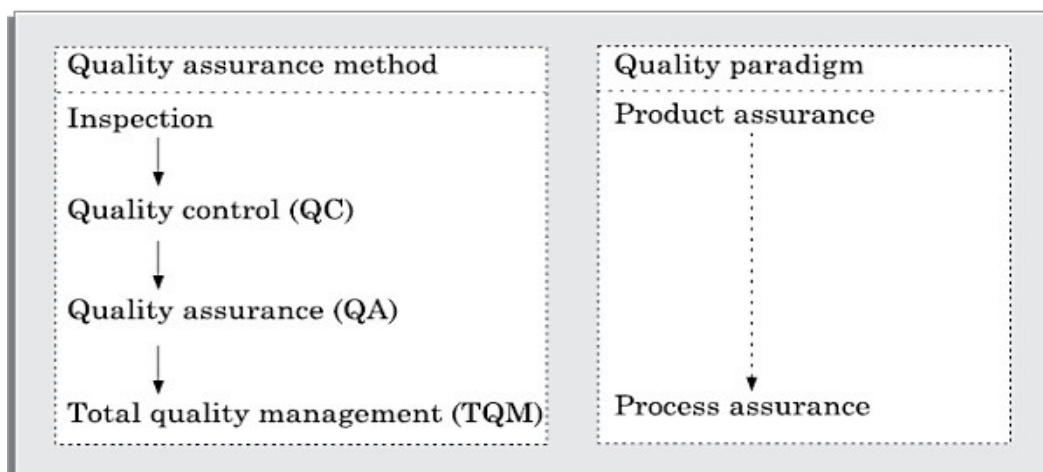


Fig: Evolution of quality system and corresponding shift in the quality paradigm

ISO 9000 certification

ISO (International Standards Organization) is a consortium of 63 countries established to formulate and foster standardization. ISO published its 9000 series of standards in 1987. ISO certification serves as a reference for contract between independent parties. The ISO 9000 standard

specifies the guidelines for maintaining a quality system. We have already seen that the quality system of an organization applies to all activities related to its product or service. The ISO standard mainly addresses operational aspects and organizational aspects such as responsibilities, reporting, etc. In a nutshell, ISO 9000 specifies a set of guidelines for repeatable and high quality product development. It is important to realize that ISO 9000 standard is a set of guidelines for the production process and is not directly concerned about the product itself.

Types of ISO 9000 quality standards

- ◆ ISO 9000 is a series of three standards: ISO 9001, ISO 9002, and ISO 9003. The ISO 9000 series of standards is based on the premise that if a proper process is followed for production, then good quality products are bound to follow automatically. The types of industries to which the different ISO standards apply are as follows.
- ◆ ISO 9001 applies to the organizations engaged in design, development, production, and servicing of goods. This is the standard that is applicable to most software development organizations.
- ◆ ISO 9002 applies to those organizations which do not design products but are only involved in production. Examples of these category industries include steel and car manufacturing industries that buy the product and plant designs from external sources and are involved in only manufacturing those products. Therefore, ISO 9002 is not applicable to software development organizations.
- ◆ ISO 9003 applies to organizations that are involved only in installation and testing of the products.

Software products vs. other products

There are mainly two differences between software products and any other type of products.

- Software is intangible in nature and therefore difficult to control. It is very difficult to control and manage anything that is not seen. In contrast, any other industries such as car manufacturing industries where one can see a product being developed through various stages such as fitting engine, fitting doors, etc. Therefore, it is easy to accurately determine how much work has been completed and to estimate how much more time will it take.
- During software development, the only raw material consumed is data. In contrast, large quantities of raw materials are consumed during the development of any other product.

Need for obtaining ISO 9000 certification

There is a mad scramble among software development organizations for obtaining ISO certification due to the benefits it offers. Some benefits that can be acquired to organizations by obtaining ISO certification are as follows:

Confidence of customers in an organization increases when organization qualifies for ISO certification. This is especially true in the international market. In fact, many organizations awarding international software development contracts insist that the development organization have ISO 9000 certification. For this reason, it is vital for software organizations involved in software export to obtain ISO 9000 certification.

- ISO 9000 requires a well-documented software production process to be in place. A well-documented software production process contributes to repeatable and higher quality of the developed software.
- ISO 9000 makes the development process focused, efficient, and cost-effective.
- ISO 9000 certification points out the weak points of an organization and recommends remedial action.
- ISO 9000 sets the basic framework for the development of an optimal process and Total Quality Management (TQM)

Summary of ISO 9001 certification

A summary of the main requirements of ISO 9001 as they relate to software development is as follows. Section numbers in brackets correspond to those in the standard itself:

Management Responsibility (4.1)

- The management must have an effective quality policy.
- The responsibility and authority of all those whose work affects quality must be defined and documented.
- A management representative, independent of the development process, must be responsible for the quality system. This requirement probably has been put down so that the person responsible for the quality system can work in an unbiased manner.
- The effectiveness of the quality system must be periodically reviewed by audits.

Quality System (4.2)

A quality system must be maintained and documented.

Contract Reviews (4.3)

◆ Before entering into a contract, an organization must review the contract to ensure that it is understood, and that the organization has the necessary capability for carrying out its obligations.

Design Control (4.4)

- The design process must be properly controlled, this includes controlling coding also. This requirement means that a good configuration control system must be in place.
- Design inputs must be verified as adequate.
- Design must be verified.
- Design output must be of required quality.
- Design changes must be controlled.

Document Control (4.5)

- There must be proper procedures for document approval, issue and removal.
- Document changes must be controlled. Thus, use of some configuration management tools is necessary.

Purchasing (4.6)

◆ purchasing material, including bought-in software must be checked for conforming to requirements.

Purchaser Supplied Product (4.7)

◆ Material supplied by a purchaser, for example, client-provided software must be properly managed and checked.

Product Identification (4.8)

◆ The product must be identifiable at all stages of the process. In software terms this means configuration management.

Process Control (4.9)

- The development must be properly managed.
- Quality requirement must be identified in a quality plan.

Inspection and Testing (4.10)

- ◆ In software terms this requires effective testing i.e., unit testing, integration testing and system testing. Test records must be maintained.

Inspection, Measuring and Test Equipment (4.11)

- ◆ If integration, measuring, and test equipments are used, they must be properly maintained and calibrated.

Inspection and Test Status (4.12)

- ◆ The status of an item must be identified. In software terms this implies configuration management and release control.

Control of Nonconforming Product (4.13)

- ◆ In software terms, this means keeping untested or faulty software out of the released product, or other places where it might cause damage.

Corrective Action (4.14)

- ◆ This requirement is both about correcting errors when found, and also investigating why the errors occurred and improving the process to prevent occurrences. If an error occurs despite the quality system, the system needs improvement.

Handling, (4.15)

- ◆ This clause deals with the storage, packing, and delivery of the software product.

Quality records (4.16)

- ◆ Recording the steps taken to control the quality of the process is essential in order to be able to confirm that they have actually taken place.

Quality Audits (4.17)

- ◆ Audits of the quality system must be carried out to ensure that it is effective.

Training (4.18)

- ◆ Training needs must be identified and met.

Salient features of ISO 9001 certification

The salient features of ISO 9001 are as follows:

- ◆ All documents concerned with the development of a software product should be properly managed, authorized, and controlled. This requires a configuration management system to be in place.
- ◆ Proper plans should be prepared and then progress against these plans should be monitored.
- ◆ Important documents should be independently checked and reviewed for effectiveness and correctness.
- ◆ The product should be tested against specification.
- ◆ Several organizational aspects should be addressed e.g., management reporting of the quality team.

Shortcomings of ISO 9000 certification

- ◆ Even though ISO 9000 aims at setting up an effective quality system in an organization, it suffers from several shortcomings. Some of these shortcomings of the ISO 9000 certification process are the following:

- ◆ ISO 9000 requires a software production process to be adhered to but does not guarantee the process to be of high quality. It also does not give any guideline for defining an appropriate process.
- ◆ ISO 9000 certification process is not fool-proof and no international accreditation agency exists. Therefore it is likely that variations in the norms of awarding certificates can exist among the different accreditation agencies and also among the registrars.
- ◆ Organizations getting ISO 9000 certification often tend to downplay domain expertise. These organizations start to believe that since a good process is in place, any engineer is as effective as any other engineer in doing any particular activity relating to software development. However, many areas of software development are so specialized that special expertise and experience in these areas (domain expertise) is required. In manufacturing industry there is a clear link between process quality and product quality. Once a process is calibrated, it can be run again and again producing quality goods. In contrast, software development is a creative process and individual skills and experience are important.
- ◆ ISO 9000 does not automatically lead to continuous process improvement, i.e. does not automatically lead to TQM.

SEI Capability Maturity Model

SEI Capability Maturity Model (SEI CMM) helped organizations to improve the quality of the software they develop and therefore adoption of SEI CMM model has significant business benefits.

SEI CMM can be used two ways: capability evaluation and software process assessment. Capability evaluation and software process assessment differ in motivation, objective, and the final use of the result. Capability evaluation provides a way to assess the software process capability of an organization. The results of capability evaluation indicates the likely contractor performance if the contractor is awarded a work. Therefore, the results of software process capability assessment can be used to select a contractor. On the other hand, software process assessment is used by an organization with the objective to improve its process capability. Thus, this type of assessment is for purely internal use.

SEI CMM classifies software development industries into the following five maturity levels. The different levels of SEI CMM have been designed so that it is easy for an organization to slowly build its quality system starting from scratch.

Level 1: Initial. A software development organization at this level is characterized by ad hoc activities. Very few or no processes are defined and followed. Since software production processes are not defined, different engineers follow their own process and as a result development efforts become chaotic. Therefore, it is also called chaotic level. The success of projects depends on individual efforts and heroics. When engineers leave, the successors have great difficulty in understanding the process followed and the work completed. Since formal project management practices are not followed, under time pressure short cuts are tried out leading to low quality.

Level 2: Repeatable. At this level, the basic project management practices such as tracking cost and schedule are established. Size and cost estimation techniques like function point analysis, COCOMO, etc. are used. The necessary process discipline is in place to repeat earlier success on

projects with similar applications. Please remember that opportunity to repeat a process exists only when a company produces a family of products.

Level 3: Defined. At this level the processes for both management and development activities are defined and documented. There is a common organization-wide understanding of activities, roles, and responsibilities. The processes though defined, the process and product qualities are not measured. ISO 9000 aims at achieving this level.

Level 4: Managed. At this level, the focus is on software metrics. Two types of metrics are collected. Product metrics measure the characteristics of the product being developed, such as its size, reliability, time complexity, understandability, etc. Process metrics reflect the effectiveness of the process being used, such as average defect correction time, productivity, average number of defects found per hour inspection, average number of failures detected during testing per LOC, etc. Quantitative quality goals are set for the products. The software process and product quality are measured and quantitative quality requirements for the product are met. Various tools like Pareto charts, fishbone diagrams, etc. are used to measure the product and process quality. The process metrics are used to check if a project performed satisfactorily. Thus, the results of process measurements are used to evaluate project performance rather than improve the process.

Level 5: Optimizing. At this stage, process and product metrics are collected. Process and product measurement data are analyzed for continuous process improvement. For example, if from an analysis of the process measurement results, it was found that the code reviews were not very effective and a large number of errors were detected only during the unit testing, then the process may be fine tuned to make the review more effective. Also, the lessons learned from specific projects are incorporated in to the process. Continuous process improvement is achieved both by carefully analyzing the quantitative feedback from the process measurements and also from application of innovative ideas and technologies. Such an organization identifies the best software engineering practices and innovations which may be tools, methods, or processes.

Key process areas (KPA) of a software organization: Except for SEI CMM level 1, each maturity level is characterized by several Key Process Areas (KPAs) that includes the areas an organization should focus to improve its software process to the next level. The focus of each level and the corresponding key process areas are shown in the figure

CMM Level	Focus	Key Process Ares
1. Initial	Competent people	
2. Repeatable	Project management	Software project planning Software configuration management
3. Defined	Definition of processes	Process definition Training program Peer reviews
4. Managed	Product and process quality	Quantitative process metrics Software quality management
5. Optimizing	Continuous process improvement	Defect prevention Process change management Technology change management

Fig: The focus of each SEI CMM level and the corresponding key process areas

SEI CMM provides a list of key areas on which to focus to take an organization from one level of maturity to the next. Thus, it provides a way for gradual quality improvement over several stages. Each stage has been carefully designed such that one stage enhances the capability already built up. For example, it considers that trying to implement a defined process (SEI CMM level 3) before a repeatable process (SEI CMM level 2) would be counterproductive as it becomes difficult to follow the defined process due to schedule and budget pressures.

ISO 9000 certification vs. SEI/CMM

For quality appraisal of a software development organization, the characteristics of ISO 9000 certification and the SEI CMM differ in some respects. The differences are as follows:

- ◆ ISO 9000 is awarded by an international standards body. Therefore, ISO 9000 certification can be quoted by an organization in official documents, communication with external parties, and the tender quotations. However, SEI CMM assessment is purely for internal use.
- ◆ SEI CMM was developed specifically for software industry and therefore addresses many issues which are specific to software industry alone.
- ◆ SEI CMM goes beyond quality assurance and prepares an organization to ultimately achieve [Total Quality Management](#) (TQM). In fact, ISO 9001 aims at level 3 of SEI CMM model.
- ◆ SEI CMM model provides a list of key process areas (KPAs) on which an organization at any maturity level needs to concentrate to take it from one maturity level to the next. Thus, it provides a way for achieving gradual quality improvement.

Applicability of SEI CMM to organizations

Highly systematic and measured approach to software development suits large organizations dealing with negotiated software, safety-critical software, etc. For those large organizations, SEI CMM model is perfectly applicable. But small organizations typically handle applications such as Internet, e-commerce, and are without an established product range, revenue base, and experience on past projects, etc. For such organizations, a CMM-based appraisal is probably excessive. These organizations need to operate more efficiently at the lower levels of maturity. For example, they need to practice effective project management, reviews, configuration management, etc.

Personal software process

Personal Software Process (PSP) is a scaled down version of the industrial software process. PSP is suitable for individual use. It is important to note that SEI CMM does not tell software developers how to analyze, design, code, test, or document software products, but assumes that engineers use effective personal practices. PSP recognizes that the process for individual use is different from that necessary for a team.

The quality and productivity of an engineer is to a great extent dependent on his process. PSP is a framework that helps engineers to measure and improve the way they work. It helps in developing personal skills and methods by estimating and planning, by showing how to track performance against plans, and provides a defined process which can be tuned by individuals.

Time measurement. PSP advocates that engineers should track the way they spend time. Because, boring activities seem longer than actual and interesting activities seem short. Therefore,

the actual time spent on a task should be measured with the help of a stop-clock to get an objective picture of the time spent. For example, he may stop the clock when attending a telephone call, taking a coffee break etc. An engineer should measure the time he spends for designing, writing code, testing, etc.

The PSP is schematically shown in While carrying out the different phases, they must record the log data using time measurement. During post-mortem, they can compare the log data with their project plan to achieve better planning in the future projects, to improve their process, etc.

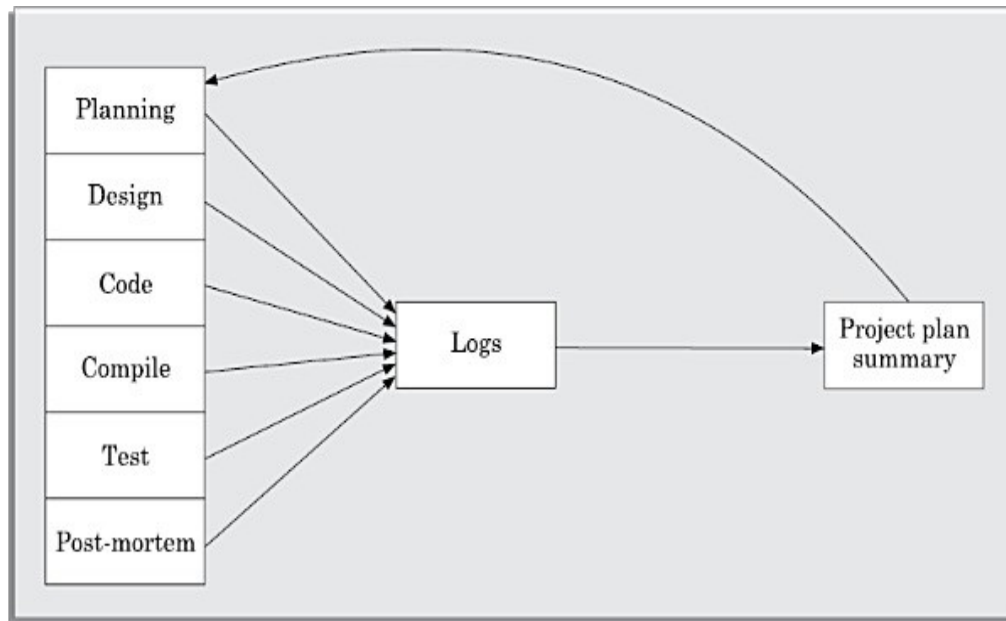


Fig: Schematic representation of PSP

The PSP levels are summarized in figure. PSP2 introduces defect management via the use of checklists for code and design reviews. The checklists are developed from gathering and analyzing defect data earlier projects.

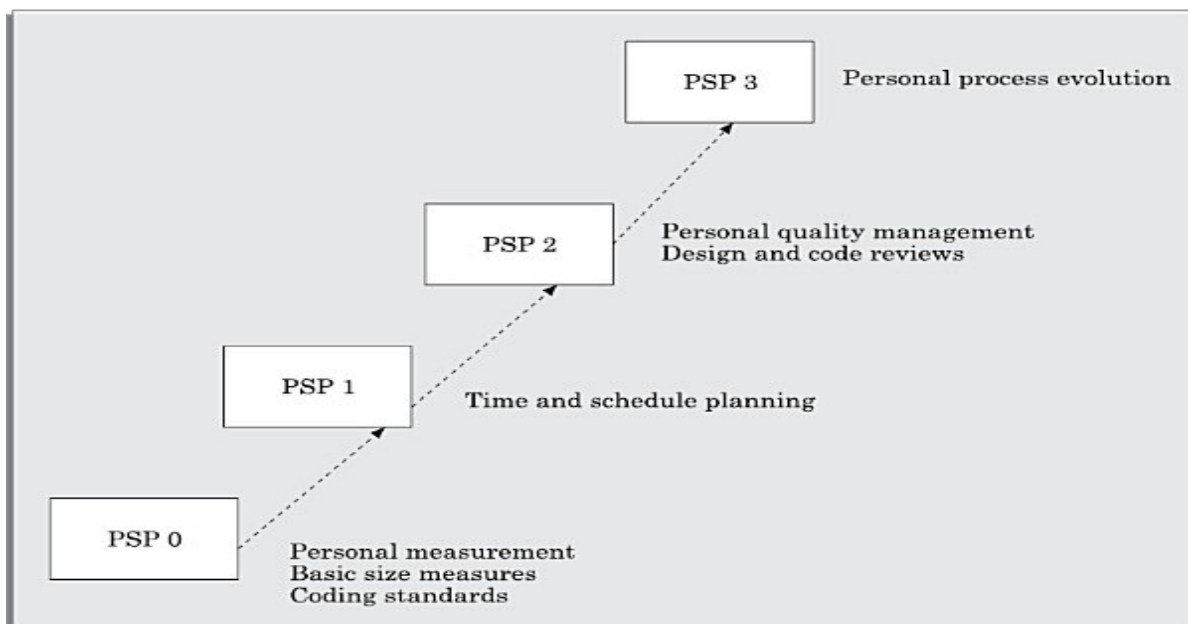


Fig: Levels of PSP

Six sigma

The purpose of Six Sigma is to improve processes to do things better, faster, and at lower cost. It can be used to improve every facet of business, from production, to human resources, to order entry, to technical support. Six Sigma can be used for any activity that is concerned with cost, timeliness, and quality of results. Therefore, it is applicable to virtually every industry.

Six Sigma at many organizations simply means striving for near perfection. Six Sigma is a disciplined, data-driven approach to eliminate defects in any process – from manufacturing to transactional and product to service.

The statistical representation of Six Sigma describes quantitatively how a process is performing. To achieve Six Sigma, a process must not produce more than 3.4 defects per million opportunities. A Six Sigma defect is defined as any system behavior that is not as per customer specifications. Total number of Six Sigma opportunities is then the total number of chances for a defect. Process sigma can easily be calculated using a Six Sigma calculator.

The fundamental objective of the Six Sigma methodology is the implementation of a measurement-based strategy that focuses on process improvement and variation reduction through the application of Six Sigma improvement projects. This is accomplished through the use of two Six Sigma sub-methodologies: DMAIC and DMADV. The Six Sigma DMAIC process (define, measure, analyze, improve, control) is an improvement system for existing processes failing below specification and looking for incremental improvement. The Six Sigma DMADV process (define, measure, analyze, design, verify) is an improvement system used to develop new processes or products at Six Sigma quality levels. It can also be employed if a current process requires more than just incremental improvement. Both Six Sigma processes are executed by Six Sigma Green Belts and Six Sigma Black Belts, and are overseen by Six Sigma Master Black Belts.

Many frameworks exist for implementing the Six Sigma methodology. Six Sigma Consultants all over the world have also developed proprietary methodologies for implementing Six Sigma quality, based on the similar change management philosophies and applications of tools.

CASE tool and its scope

A CASE (Computer Aided Software Engineering) tool is a generic term used to denote any form of automated support for software engineering. In a more restrictive sense, a CASE tool means any tool used to automate some activity associated with software development. Many CASE tools are available. Some of these CASE tools assist in phase related tasks such as specification, structured analysis, design, coding, testing, etc.; and others to non-phase activities such as project management and configuration management.

Reasons for using CASE tools

The primary reasons for using a CASE tool are:

- To increase productivity
- To help produce better quality software at lower cost

CASE Environment

Although individual CASE tools are useful, the true power of a tool set can be realized only when these set of tools are integrated into a common framework or environment. CASE tools are characterized by the stage or stages of software development life cycle on which they focus. Since different tools covering different stages share common information, it is required that they integrate through some central repository to have a consistent view of information associated with the software development artifacts. This central repository is usually a data dictionary containing the definition of all composite and elementary data items. Through the central repository all the CASE tools in a CASE environment share common information among themselves. Thus a CASE environment facilitates the automation of the step-by-step methodologies for software development. A schematic representation of a CASE environment is shown in figure.

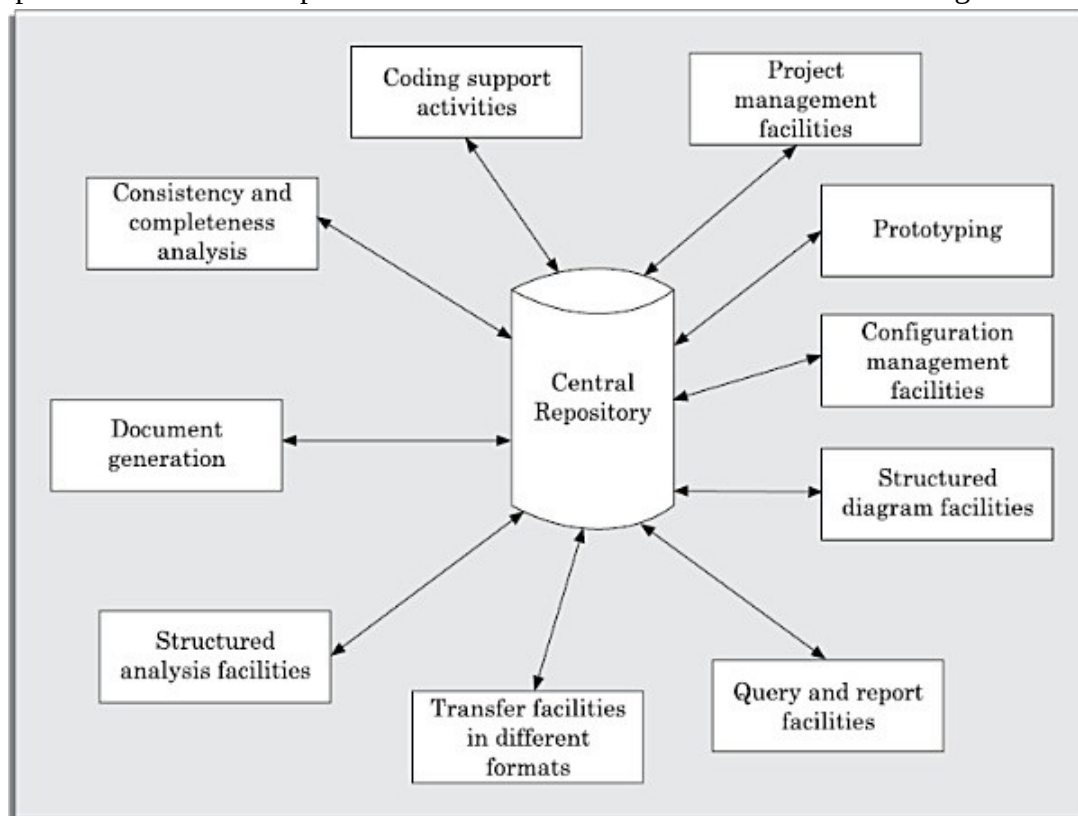


Fig: A CASE Environment

The standard programming environments such as Turbo C, Visual C++, etc. come equipped with a program editor, compiler, debugger, linker, etc., All these tools are integrated. If you click on an error reported by the compiler, not only does it take you into the editor, but also takes the cursor to the specific line or statement causing the error.

CASE environment vs programming environment

A CASE environment facilitates the automation of the step-by-step methodologies for software development. In contrast to a CASE environment, a programming environment is an integrated collection of tools to support only the coding phase of software development.

Benefits of CASE

Several benefits accrue from the use of a CASE environment or even isolated CASE tools. Some of those benefits are:

- Use of CASE tools leads to considerable improvements to quality. This is mainly due to the facts that one can effortlessly iterate through the different phases of software development and the chances of human error are considerably reduced.
- CASE tools help produce high quality and consistent documents. Since the important data relating to a software product are maintained in a central repository, redundancy in the stored data is reduced and therefore chances of inconsistent documentation is reduced to a great extent.
- CASE tools take out most of the drudgery in a software engineer's work. For example, they need not check meticulously the balancing of the DFDs but can do it effortlessly through the press of a button.
- CASE tools have led to revolutionary cost saving in software maintenance efforts. This arises not only due to the tremendous value of a CASE environment in traceability and consistency checks, but also due to the systematic information capture during the various phases of software development as a result of adhering to a CASE environment.
- Introduction of a CASE environment has an impact on the style of working of a company, and makes it oriented towards the structured and orderly approach.

CASE SUPPORT IN SOFTWARE LIFE CYCLE

Let us examine the various types of support that CASE provides during the different phases of a software life cycle. CASE tools should support a development methodology, help enforce the same, and provide certain amount of consistency checking between different phases. Some of the possible support that CASE tools usually provide in the software development life cycle are discussed below.

Requirements of a prototyping CASE tool

Prototyping is useful to understand the requirements of complex software products, to demonstrate a concept, to market new ideas, and so on. The important features of a prototyping CASE tool are as follows:

- ◆ Define user interaction
- ◆ Define the system control flow
- ◆ Store and retrieve data required by the system
- ◆ Incorporate some processing logic

Features of a good prototyping CASE tool

There are several stand-alone prototyping tools. But a tool that integrates with the data dictionary can make use of the entries in the data dictionary, help in populating the data dictionary and ensure the consistency between the design data and the prototype. A good prototyping tool should support the following features:

- Since one of the main uses of a prototyping CASE tool is graphical user interface (GUI) development, prototyping CASE tool should support the user to create a GUI using a graphics editor. The user should be allowed to define all data entry forms, menus and controls.

- It should integrate with the data dictionary of a CASE environment.
- If possible, it should be able to integrate with external user defined modules written in C or some popular high level programming languages.
- The user should be able to define the sequence of states through which a created prototype can run. The user should also be allowed to control the running of the prototype.
- The run time system of prototype should support mock runs of the actual system and management of the input and output data

Structured analysis and design with CASE tools

Several diagramming techniques are used for structured analysis and structured design. The following supports might be available from CASE tools.

- A CASE tool should support one or more of the structured analysis and design techniques.
- It should support effortlessly drawing analysis and design diagrams.
- It should support drawing for fairly complex diagrams, preferably through a hierarchy of levels.
- The CASE tool should provide easy navigation through the different levels and through the design and analysis.
- The tool must support completeness and consistency checking across the design and analysis and through all levels of analysis hierarchy. Whenever it is possible, the system should disallow any inconsistent operation, but it may be very difficult to implement such a feature. Whenever there arises heavy computational load while consistency checking, it should be possible to temporarily disable consistency checking.

Code generation and CASE tools

As far as code generation is concerned, the general expectation of a CASE tool is quite low. A reasonable requirement is traceability from source file to design data. More pragmatic supports expected from a CASE tool during code generation phase are the following:

- The CASE tool should support generation of module skeletons or templates in one or more popular languages. It should be possible to include copyright message, brief description of the module, author name and the date of creation in some selectable format.
- The tool should generate records, structures, class definition automatically from the contents of the data dictionary in one or more popular languages.
- It should generate database tables for relational database management systems.
- The tool should generate code for user interface from prototype definition for X window and MS window based applications.

Test case generation CASE tool

The CASE tool for test case generation should have the following features:

- 1.It should support both design and requirement testing.
- 2.It should generate test set reports in ASCII format which can be directly imported into the test plan document.

Hardware and Environmental Requirements

In most cases, it is the existing hardware that would place constraints upon the CASE tool selection. Thus, instead of defining hardware requirements for a CASE tool, the task at hand becomes to fit in an optimal configuration of CASE tool in the existing hardware capabilities. Therefore, we have to emphasise on selecting the most optimal CASE tool configuration for a given hardware configuration.

The heterogeneous network is one instance of distributed environment and we choose this for illustration as it is more popular due to its machine independent features. The CASE tool implementation in heterogeneous network makes use of client-server paradigm. The multiple clients which run different modules access data dictionary through this server. The data dictionary server may support one or more projects. Though it is possible to run many servers for different projects but distributed implementation of data dictionary is not common. The tool set is integrated through the data dictionary which supports multiple projects, multiple users working simultaneously and allows to share information between users and projects. The data dictionary provides consistent view of all project entities, e.g., a data record definition and its entity-relationship diagram be consistent. The server should depict the per-project logical view of the data dictionary. This means that it should allow back up/restore, copy, cleaning part of the data dictionary, etc. The tool should work satisfactorily for maximum possible number of users working simultaneously. The tool should support multi- windowing environment for the users. This is important to enable the users to see more than one diagram at a time. It also facilitates navigation and switching from one part to the other.

Documentation Support

The deliverable documents should be organized graphically and should be able to incorporate text and diagrams from the central repository. This helps in producing up-to-date documentation. The CASE tool should integrate with one or more of the commercially available desk- top publishing packages. It should be possible to export text, graphics, tables, data dictionary reports to the DTP package in standard forms such as PostScript.

Project Management

It should support collecting, storing, and analysing information on the software project's progress such as the estimated task duration, scheduled and actual task start, completion date, dates and results of the reviews, etc.

External Interface

The tool should allow exchange of information for reusability of design. The information which is to be exported by the tool should be preferably in ASCII format and support open architecture. Similarly, the data dictionary should provide a programming interface to access information. It is required for integration of custom utilities, building new techniques, or populating the data dictionary.

Reverse Engineering Support

The tool should support generation of structure charts and data dictionaries from the existing source codes. It should populate the data dictionary from the source code. If the tool is used for re-engineering information systems, it should contain conversion tool from indexed sequential file structure, hierarchical and network database to relational database systems.

Data Dictionary Interface

The data dictionary interface should provide view and update access to the entities and relations stored in it. It should have print facility to obtain hard copy of the viewed screens. It should provide analysis reports like cross-referencing, impact analysis, etc. Ideally, it should support a query language to view its contents.

Tutorial and Help

The application of CASE tool and thereby its success depends on the users' capability to effectively use all the features supported. Therefore, for the uninitiated users, a tutorial is very important. The tutorial should not be limited to teaching the user interface part only, but should comprehensively cover the following points:

- The tutorial should cover all techniques and facilities through logically classified sections.
- The tutorial should be supported by proper documentation.

Second-generation CASE tool

An important feature of the second-generation CASE tool is the direct support of any adapted methodology. This would necessitate the function of a CASE administrator organization who can tailor the CASE tool to a particular methodology. In addition, the second-generation CASE tools have following features:

- ◆ **Intelligent diagramming support.** The fact that diagramming techniques are useful for system analysis and design is well established. The future CASE tools would provide help to aesthetically and automatically lay out the diagrams.
- ◆ **Integration with implementation environment.** The CASE tools should provide integration between design and implementation.
- ◆ **Data dictionary standards.** The user should be allowed to integrate many development tools into one environment. It is highly unlikely that any one vendor will be able to deliver a total solution. Moreover, a preferred tool would require tuning up for a particular system. Thus the user would act as a system integrator. This is possibly only if some standard on data dictionary emerges.
- ◆ **Customization support.** The user should be allowed to define new types of objects and connections. This facility may be used to build some special methodologies. Ideally it should be possible to specify the rules of a methodology to a rule engine for carrying out the necessary consistency checks.

Architecture of a CASE environment

The architecture of a typical modern CASE environment is shown diagrammatically in figure. The important components of a modern CASE environment are user interface, tool set, object management system (OMS), and a repository. Characteristics of a tool set have been discussed earlier.

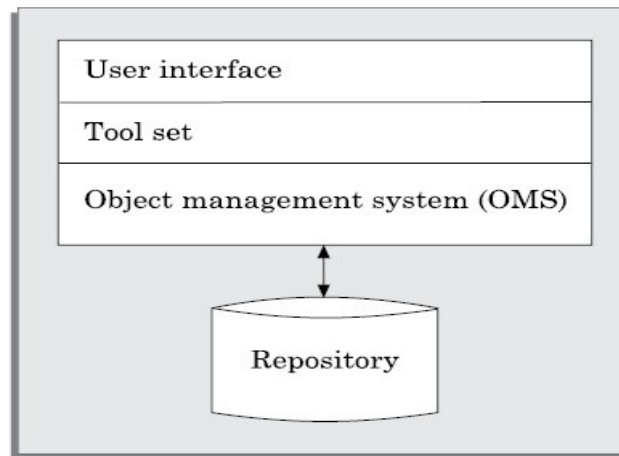


Fig: Architecture of a Modern CASE Environment

Object Management System (OMS) and Repository

Different case tools represent the software product as a set of entities such as specification, design, text data, project plan, etc. The object management system maps these logical entities such into the underlying storage management system (repository). The commercial relational database management systems are geared towards supporting large volumes of information structured as simple relatively short records. There are a few types of entities but large number of instances. By contrast, CASE tools create a large number of entity and relation types with perhaps a few instances of each. Thus the object management system takes care of appropriately mapping into the underlying storage management system.

CHARACTERISTICS OF SOFTWARE MAINTENANCE

Software maintenance is becoming an important activity of a large number of organisations. This is no surprise, given the rate of hardware obsolescence, the immortality of a software product per se, and the demand of the user community to see the existing software products run on newer platforms, run in newer environments, and/or with enhanced features. When the hardware platform changes, and a software product performs some low-level functions, maintenance is necessary. Also, whenever the support environment of a software product changes, the software product requires rework to cope up with the newer interface. For instance, a software product may need to be maintained when the operating system changes. Thus, every software product continues to evolve after its development through maintenance efforts.

Types of software maintenance

There are basically three types of software maintenance. These are:

Corrective: Corrective maintenance of a software product is necessary to rectify the bugs observed while the system is in use.

Perfective: A software product needs maintenance to support the new features that users want it to support, to change different functionalities of the system according to customer demands or to enhance the performance of the system.

Problems associated with software maintenance

- Software maintenance work typically is much more expensive than what it should be and takes more time than required. In software organizations, maintenance work is mostly carried out using ad hoc techniques. The primary reason being that software maintenance is one of the most neglected areas of software engineering. Even though software maintenance is fast becoming an important area of work for many companies as the software products of yester years age, still software maintenance is mostly being carried out as fire-fighting operations, rather than through systematic and planned activities.
- Software maintenance has a very poor image in industry. Therefore, an organization often cannot employ bright engineers to carry out maintenance work. Even though maintenance suffers from a poor image, the work involved is often more challenging than development work. During maintenance it is necessary to thoroughly understand someone else's work and then carry out the required modifications and extensions.
- Another problem associated with maintenance work is that the majority of software products needing maintenance are legacy products.

Software reverse engineering

- Software reverse engineering is the process of recovering the design and the requirements specification of a product from an analysis of its code. The purpose of reverse engineering is to facilitate maintenance work by improving the understandability of a system and to produce the necessary documents for a legacy system. Reverse engineering is becoming important, since legacy software products lack proper documentation, and are highly unstructured. Even well-designed products become legacy software as their structure degrades through a series of maintenance efforts.
- The first stage of reverse engineering usually focuses on carrying out cosmetic changes to the code to improve its readability, structure, and understandability, without changing of its functionalities. A process model for reverse engineering has been shown in figure. A program can be reformatted using any of the several available prettyprinter programs which layout the program neatly. Many legacy software products with complex control structure and unthoughtful variable names are difficult to comprehend. Assigning meaningful variable names is important because meaningful variable names are the most helpful thing in code documentation. All variables, data structures, and functions should be assigned meaningful names wherever possible. Complex nested conditionals in the program can be replaced by simpler conditional statements or whenever appropriate by case statements.

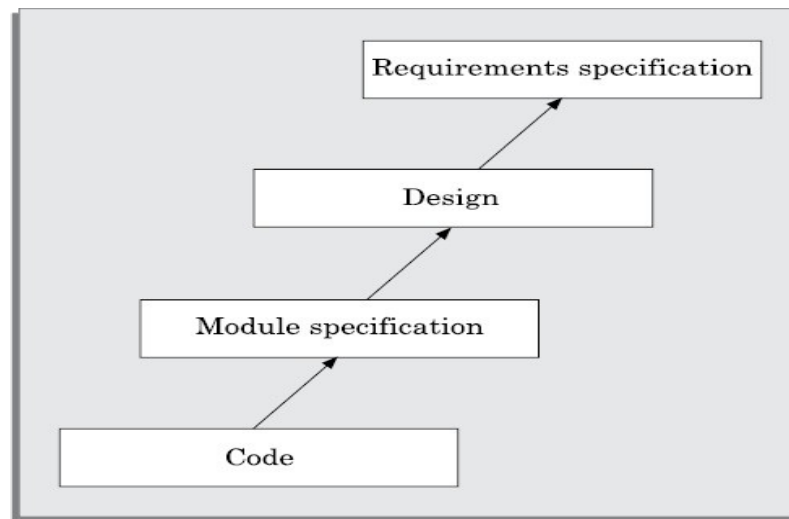


Fig: A process model for reverse engineering

After the cosmetic changes have been carried out on a legacy software, the process of extracting the code, design, and the requirements specification can begin. These activities are schematically shown in figure. In order to extract the design, a full understanding of the code is needed. Some automatic tools can be used to derive the data flow and control flow diagram from the code. The structure chart (module invocation sequence and data interchange among modules) should also be extracted. The SRS document can be written once the full code has been thoroughly understood and the design extracted.

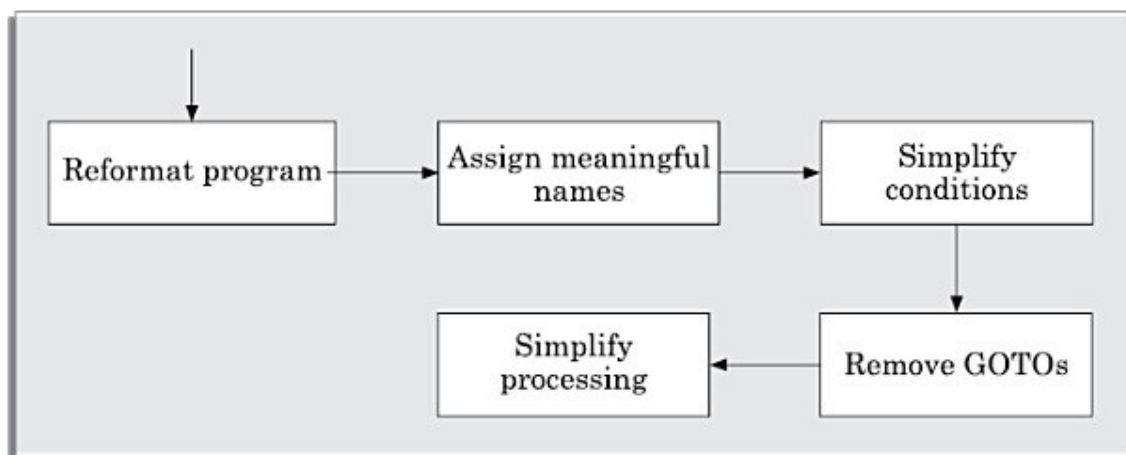


Fig: Cosmetic changes carried out before reverse engineering

Legacy software products

- It is prudent to define a legacy system as any software system that is hard to maintain. The typical problems associated with legacy systems are poor documentation, unstructured (spaghetti code with ugly control structure), and lack of personnel knowledgeable in the product. Many of the legacy systems were developed long time back. But, it is possible that a recently developed system having poor design and documentation can be considered to be a legacy system.

Factors on which software maintenance activities depend

The activities involved in a software maintenance project are not unique and depend on several factors such as:

- the extent of modification to the product required
- the resources available to the maintenance team
- the conditions of the existing product (e.g., how structured it is, how well documented it is, etc.)
- the expected project risks, etc.

When the changes needed to a software product are minor and straightforward, the code can be directly modified and the changes appropriately reflected in all the documents. But more elaborate activities are required when the required changes are not so trivial. Usually, for complex maintenance projects for legacy systems, the software process can be represented by a reverse engineering cycle followed by a forward engineering cycle with an emphasis on as much reuse as possible from the existing code and other documents.

Software maintenance process models

Two broad categories of process models for software maintenance can be proposed.

First model: The first model is preferred for projects involving small reworks where the code is changed directly and the changes are reflected in the relevant documents later. This maintenance process is graphically presented in figure. In this approach, the project starts by gathering the requirements for changes. The requirements are next analyzed to formulate the strategies to be adopted for code change. At this stage, the association of at least a few members of the original development team goes a long way in reducing the cycle time, especially for projects involving unstructured and inadequately documented code. The availability of a working old system to the maintenance engineers at the maintenance site greatly facilitates the task of the maintenance team as they get a good insight into the working of the old system and also can compare the working of their modified system with the old system. Also, debugging of the reengineered system becomes easier as the program traces of both the systems can be compared to localize the bugs.

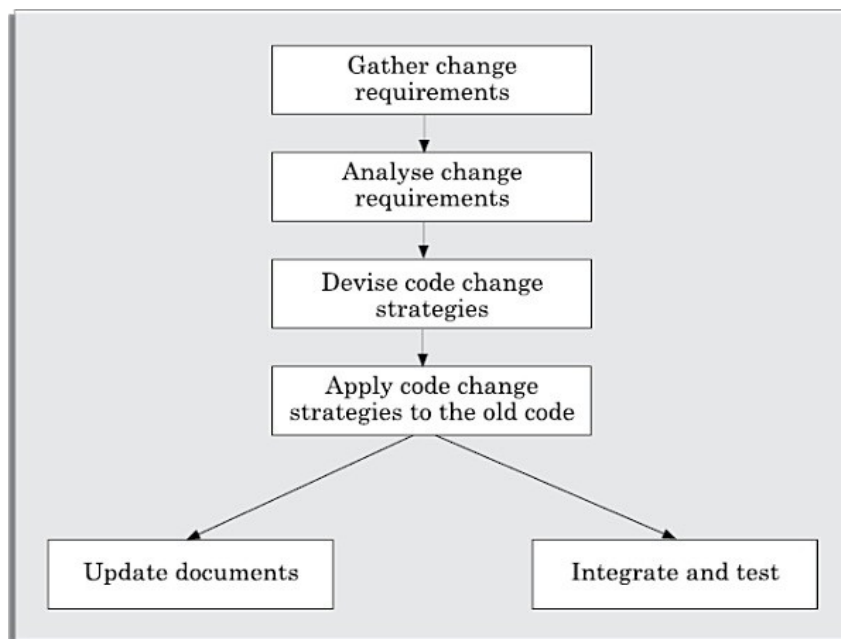


Fig: Maintenance process model 1

Second Model: The second process model for software maintenance is preferred for projects where the amount of rework required is significant. This approach can be represented by a reverse

engineering cycle followed by a forward engineering cycle. Such an approach is also known as software reengineering. This process model is depicted in figure.

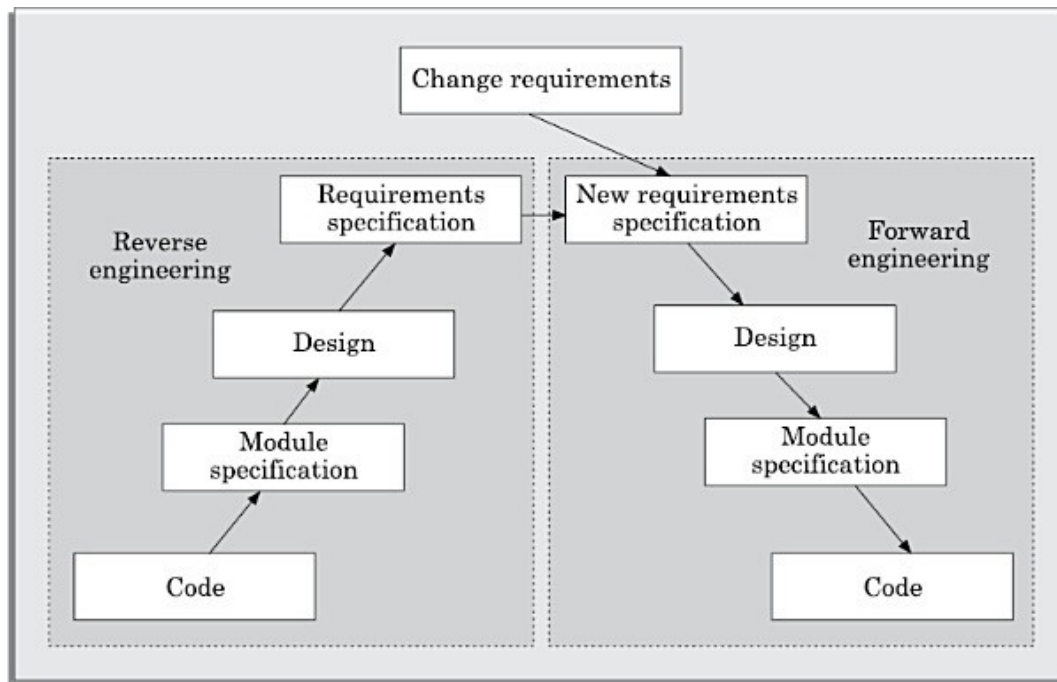


Fig: Maintenance process model 2

The reverse engineering cycle is required for legacy products. During the reverse engineering, the old code is analyzed (abstracted) to extract the module specifications. The module specifications are then analyzed to produce the design. The design is analyzed (abstracted) to produce the original requirements specification. The change requests are then applied to this requirements specification to arrive at the new requirements specification. At the design, module specification, and coding a substantial reuse is made from the reverse engineered products. An important advantage of this approach is that it produces a more structured design compared to what the original product had, produces good documentation, and very often results in increased efficiency. The efficiency improvements are brought about by a more efficient design. However, this approach is more costly than the first approach. An empirical study indicates that process 1 is preferable when the amount of rework is no more than 15%. Besides the amount of rework, several other factors might affect the decision regarding using process model 1 over process model 2:

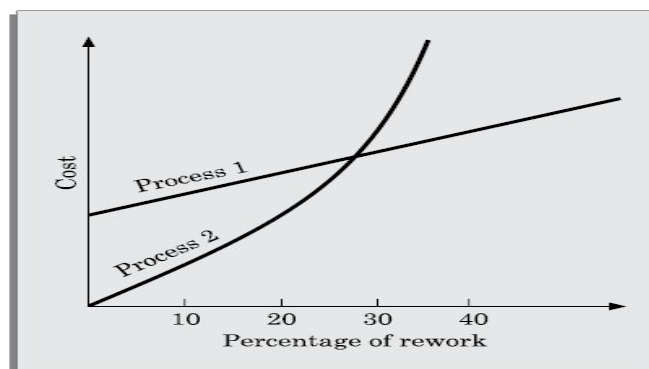


Fig: Empirical estimation of maintenance cost versus percentage rework

- Reengineering might be preferable for products which exhibit a high failure rate.
- Reengineering might also be preferable for legacy products having poor design and code structure.

Software reengineering

Software reengineering is a combination of two consecutive processes i.e. software reverse engineering and software forward engineering as shown in the figure.

Estimation of approximate maintenance cost

- It is well known that maintenance efforts require about 60% of the total life cycle cost for a typical software product. However, maintenance costs vary widely from one application domain to another. For embedded systems, the maintenance cost can be as much as 2 to 4 times the development cost.
- Boehm [1981] proposed a formula for estimating maintenance costs as part of his COCOMO cost estimation model. Boehm's maintenance cost estimation is made in terms of a quantity called the Annual Change Traffic (ACT). Boehm defined ACT as the fraction of a software product's source instructions which undergo change during a typical year either through addition or deletion.

$$ACT = \frac{KLOC_{added} + KLOC_{deleted}}{KLOC_{total}}$$

where, $KLOC_{added}$ is the total kilo lines of source code added during maintenance. $KLOC_{deleted}$ is the total KLOC deleted during maintenance.

- Thus, the code that is changed, should be counted in both the code added and the code deleted. The annual change traffic (ACT) is multiplied with the total development cost to arrive at the maintenance cost:

$$\text{maintenance cost} = ACT \times \text{development cost}$$

- Most maintenance cost estimation models, however, yield only approximate results because they do not take into account several factors such as experience level of the engineers, and familiarity of the engineers with the product, hardware requirements, software complexity, etc.

Advantages of software reuse

Software products are expensive. Software project managers are worried about the high cost of software development and are desperately look for ways to cut development cost. A possible way to reduce development cost is to reuse parts from previously developed software. In addition to reduced development cost and time, reuse also leads to higher quality of the developed products since the reusable components are ensured to have high quality.

Artifacts that can be reused

It is important to know about the kinds of the artifacts associated with software development that can be reused. Almost all artifacts associated with software development, including project plan and test plan can be reused. However, the prominent items that can be effectively reused are:

- Requirements specification
- Design
- Code
- Test cases
- Knowledge

Pros and cons of knowledge reuse

Knowledge is the most abstract development artifact that can be reused. Out of all the reuse artifacts i.e. requirements specification, design, code, test cases, the details of the potentially reusable development knowledge. A planned reuse of knowledge can increase the effectiveness of reuse. For this, the reusable knowledge should be systematically extracted and documented. But, it is usually very difficult to extract and document reusable knowledge.

Easiness of reuse of mathematical functions

The routines of mathematical libraries are being reused very successfully by almost every programmer. No one in his right mind would think of writing a routine to compute sine or cosine. Reuse of commonly used mathematical functions is easy. Several interesting aspects emerge. Cosine means the same to all. Everyone has clear ideas about what kind of argument should cosine take, the type of processing to be carried out and the results returned. Secondly, mathematical libraries have a small interface. For example, cosine requires only one parameter. Also, the data formats of the parameters are standardized.

Basic issues in any reuse program

The following are some of the basic issues that must be clearly understood for starting any reuse program.

- Component creation
- Component indexing and storing
- Component search
- Component understanding
- Component adaptation
- Repository maintenance

Component creation. For component creation, the reusable components have to be first identified. Selection of the right kind of components having potential for reuse is important. Domain analysis is a promising technique which can be used to create reusable components.

Component indexing and storing. Indexing requires classification of the reusable components so that they can be easily searched when looking for a component for reuse. The components need to be stored in a Relational Database Management System (RDBMS) or an Object-Oriented Database System (ODBMS) for efficient access when the number of components becomes large.

Component searching. The programmers need to search for right components matching their requirements in a database of components. To be able to search components efficiently, the programmers require a proper method to describe the components that they are looking for.

Component understanding. The programmers need a precise and sufficiently complete understanding of what the component does to be able to decide whether they can reuse the component. To facilitate understanding, the components should be well documented and should do something simple.

Component adaptation. Often, the components may need adaptation before they can be reused, since a selected component may not exactly fit the problem at hand. However, tinkering with the code is also not a satisfactory solution because this is very likely to be a source of bugs.

Repository maintenance. A component repository once is created requires continuous maintenance. New components, as and when created have to be entered into the repository. The faulty components have to be tracked. Further, when new applications emerge, the older applications become obsolete. In this case, the obsolete components might have to be removed from the repository.

Domain analysis

The aim of domain analysis is to identify the reusable components for a problem domain.

Reuse domain. A reuse domain is a technically related set of application areas. A body of information is considered to be a problem domain for reuse, if a deep and comprehensive relationship exists among the information items as categorized by patterns of similarity among the development components of the software product. A reuse domain is shared understanding of some community, characterized by concepts, techniques, and terminologies that show some coherence. Examples of domains are accounting software domain, banking software domain, business software domain, manufacturing automation software domain, telecommunication software domain, etc.

Just to become familiar with the vocabulary of a domain requires months of interaction with the experts. Often, one needs to be familiar with a network of related domains for successfully carrying out domain analysis. Domain analysis identifies the objects, operations, and the relationships among them. For example, consider the airline reservation system, the reusable objects can be seats, flights, airports, crew, meal orders, etc. The reusable operations can be scheduling a flight, reserving a seat, assigning crew to flights, etc. The domain analysis generalizes the application domain. A domain model transcends specific applications. The common characteristics or the similarities between systems are generalized.

During domain analysis, a specific community of software developers gets together to discuss community-wide-solutions. Analysis of the application domain is required to identify the reusable components. The actual construction of reusable components for a domain is called domain engineering.

Evolution of a reuse domain. The ultimate result of domain analysis is development of problem-oriented languages. The problem-oriented languages are also known as application generators. These application generators, once developed form application development standards. The domains slowly develop. As a domain develops, it is distinguishable the various stages it undergoes:

Stage 1: There is no clear and consistent set of notations. Obviously, no reusable components are available. All software is written from scratch.

Stage 2: Here, only experience from similar projects is used in a development effort. This means that there is only knowledge reuse.

Stage 3: At this stage, the domain is ripe for reuse. The set of concepts are stabilized and the notations standardized. Standard solutions to standard problems are available. There is both knowledge and component reuse.

Stage 4: The domain has been fully explored. The software development for the domain can be largely automated. Programs are not written in the traditional sense any more. Programs are written using a domain specific language, which is also known as an application generator.

Re-use at organization level

Achieving organization-level reuse requires adoption of the following steps:

1. Assessing a product's potential for reuse
2. Refining products for greater reusability
3. Entering the product in the reuse repository

1. Assessing a product's potential for reuse. Assessment of components reuse potential can be obtained from an analysis of a questionnaire circulated among the developers. The questionnaire can be devised to assess a component's reusability. The programmers working in similar application domain can be used to answer the questionnaire about the product's reusability. Depending on the answers given by the programmers, either the component be taken up for reuse as it is, it is modified and refined before it is entered into the reuse repository, or it is ignored. A sample questionnaire to assess a component's reusability is the following.

- ◆ Is the component's functionality required for implementation of systems in the future?
- ◆ How common is the component's function within its domain?
- ◆ Would there be a duplication of functions within the domain if the component is taken up?
- ◆ Is the component hardware dependent?
- ◆ Is the design of the component optimized enough?
- ◆ If the component is non-reusable, then can it be decomposed to yield some reusable components?
- ◆ Can we parameterize a non-reusable component so that it becomes reusable?

2. Refining products for greater reusability. For a product to be reusable, it must be relatively easy to adapt it to different contexts. Machine dependency must be abstracted out or localized using data encapsulation techniques. The following refinements may be carried out:

- ◆ **Name generalization:** The names should be general, rather than being directly related to a specific application.
- ◆ **Operation generalization:** Operations should be added to make the component more general. Also, operations that are too specific to an application can be removed.
- ◆ **Exception generalization:** This involves checking each component to see which exceptions it might generate. For a general component, several types of exceptions might have to be handled.
- ◆ **Handling portability problems:** Programs typically make some assumption regarding the representation of information in the underlying machine. These

assumptions are in general not true for all machines. The programs also often need to call some operating system functionality and these calls may not be same on all machines. Also, programs use some function libraries, which may not be available on all host machines. A portability solution to overcome these problems is shown in figure. The portability solution suggests that rather than call the operating system and I/O procedures directly, abstract versions of these should be called by the application program. Also, all platform-related calls should be routed through the portability interface. One problem with this solution is the significant overhead incurred, which makes it inapplicable to many real-time systems and applications requiring very fast response.

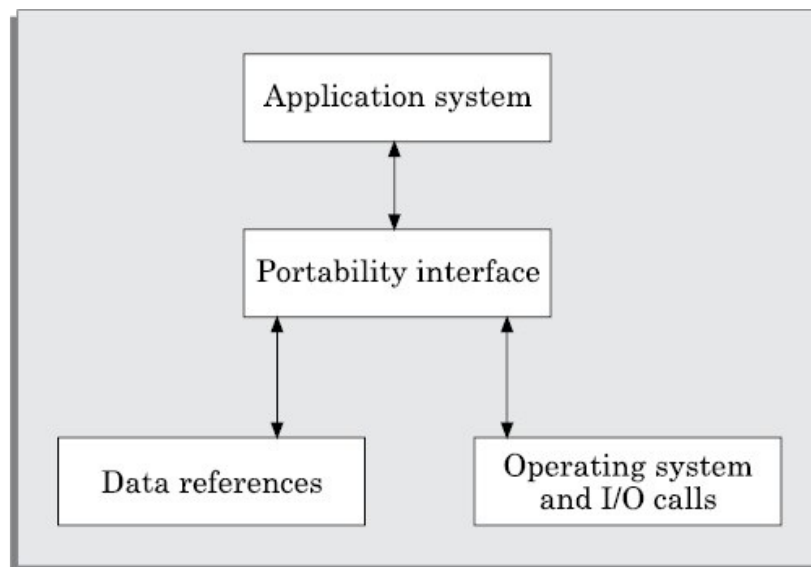


Fig: Improving reusability of a component by using a portability interface

Factors that inhibit an effective reuse program

In spite of all the shortcomings of the state-of-the-art reuse techniques, it is the experience of several organizations that most of the factors inhibiting an effective reuse program are non-technical. Some of these factors are the following.

- ◆ Need for commitment from the top management.
- ◆ Adequate documentation to support reuse.
- ◆ Adequate incentive to reward those who reuse. Both the people contributing new reusable components and those reusing the existing components should be rewarded to start a reuse program and keep it going.

Providing access to and information about reusable components. Organizations are often hesitant to provide an open access to the reuse repository for the fear of the reuse components finding a way to their competitors.

Challenges faced by software developers

Following are some of the challenges that are being faced by software developers:

- To cope up with fierce competitions, business houses are rapidly changing their business processes. This requires rapid changes to also occur to the software that support the business process activities. Therefore, there is a pressing demand to shorten the software delivery time. However, software is still taking unacceptably long time to develop and is turning out to be a bottleneck in implementing rapid business process changes. To reduce the software delivery times, software is being developed by teams working from globally distributed locations. How

software can be effectively developed using globally distributed development teams is not yet clear and poses many challenges. On the other hand, radical changes to the software development principles are being put forward to shorten the development time.

- Business houses are getting tired of astronomical software costs, late deliveries, and poor quality products. On the other hand, hardware costs are dropping and at the same time hardware is becoming more powerful, sophisticated, and reliable. Hardware and software cost differentials are becoming more and more glaring. The wisdom of developing every software from scratch is being questioned. Also, alternate software delivery models are being proposed to reduce the software cost.
- Software sizes are further increasing.
- After Internet has become vastly popular, many software products are now required to interface with the Internet. Many products are even expected to work across the Internet. Also, with the availability of fast networks, distributed applications are becoming common place. However, it is not clear that how software is to be effectively developed in the context of distributed platforms and Internet.

In response to the challenges faced, the following software engineering trends are becoming noticeable:

- Client-server software
- Service-oriented architecture (SOA)
- Software as a service (SaaS)

Client-server software

A client is basically a consumer of services and server is a provider of services as shown in figure. A client requests some services from the server and the server provides the required services to the client. Client and server are usually software components running on independent machines. Even a single machine can sometimes act as a client and at other times a server depending on the situations. Thus, client and server are mere roles.

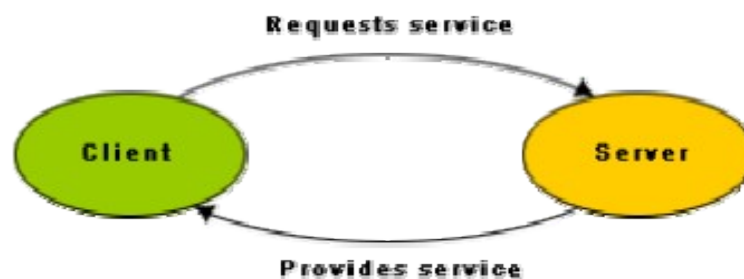


Fig: Client-server model

Example:

A man was visiting his friend's town in his car. The man had a handheld computer (client). He knew his friend's name but he didn't know his friend's address. So he sent a wireless message (request) to the nearest "address server" by his handheld computer to enquire his friend's address. The message first came to the base station. The base station forwarded that message through landline to local area network where the server is located. After some processing, LAN sent back that friend's address (service) to the man.

Advantages of client-server software

- The client-server software architecture is a versatile, message-based and modular infrastructure that is intended to improve usability, flexibility, interoperability and scalability as compared to centralized, mainframe, time sharing computing.
- There are many advantages of client-server software products as compared to monolithic ones. These advantages are:

- **Simplicity and modularity** – Client and server components are loosely coupled and therefore modular. These are easy to understand and develop.
- **Flexibility** – Both client and server software can be easily migrated across different machines in case some machine becomes unavailable or crashes. The client can access the service anywhere. Also, clients and servers can be added incrementally.
- **Loose coupling:** Client and server components are inherently loosely- coupled, making these easy to understand and develop.
- **Extensibility** – More servers and clients can be effortlessly added.
- **Concurrency** – The processing is naturally divided across several machines. Clients and servers reside in different machines which can operate in parallel and thus processing becomes faster.
- **Cost Effectiveness** – Clients can be cheap desktop computers whereas servers can be sophisticated and expensive computers. To use a sophisticated software, one needs to own only a cheap client and invoke the server.
- **Specialization** – One can have different types of computers to run different types of servers. Thus, servers can be specialized to solve some specific problems.
- **Heterogeneous hardware:** In a client-server solution, it is easy to have specialised servers that can efficiently solve specific problems. It is possible to efficiently integrate heterogeneous computing platforms to support the requirements of different types of server software.
- **Mobile computing** – Mobile computing implicitly uses client-server technique. Cell phones (handheld computers) are being provided with small processing power, keyboard, small memory, and LCD display. Cell phones cannot really compute much as they have very limited processing power and storage capacity but they can act as clients. The handheld computers only support the interface to place requests on some remote servers.
- **Application Service Providers (ASPs)** – There are many application software products which are very expensive. Thus it makes prohibitively costly to own those applications. The cost of those applications often runs into millions of dollars. For example, a Chemical Simulation Software named “*Aspen*” is very expensive but very powerful. For small industries it would not be practical to own that software. Application Service Providers can own ASPEN and let the small industries use it as client and charge them based on usage time. A client can simply log in and ASP charges according to the time that the software is used.
- **Component-based development** – It is the enabler of the client-server technology. Component-based development is radically different from traditional software development. In component-based development, a developer essentially integrates pre-built components purchased off-the- shelf. This is akin to the way hardware developers integrate ICs on a Printed Circuit Board (PCB). Components might reside on different computers which act as servers and clients.
- **Fault-tolerance** – Client-server based systems are usually fault-tolerant. There can be many servers. If one server crashes then client requests can be switched to a redundant server.

There are many other advantages of client-server software. For example, we can locate a server near to the client. There might be several servers and the client requests can be routed to the nearest server. This would reduce the communication overhead.

Disadvantages of client-server software

There are several disadvantages of client-server software development. Those disadvantages are:

- **Security** – In a monolithic application, implementation of security is very easy. But in a client-server based development a lot of flexibility is provided and a client can connect from anywhere. This makes it easy for hackers to break into the system. Therefore, ensuring security in client- server system is very challenging.
- **Servers can be bottlenecks** – Servers can turn out to be bottlenecks because many clients might try to connect to a server at the same time. This problem arises due to the flexibility given that any client can connect anytime required.
- **Compatibility** – Clients and servers may not be compatible to each other. Since the client and server components may be manufactured by different vendors, they may not be compatible with respect to data types, language, etc.
- **Inconsistency** – Replication of servers is a problem as it can make data inconsistent.

CLIENT-SERVER ARCHITECTURES

The simplest way to connect clients and servers is by using a two-tier architecture shown in Figure. In a two-tier architecture, any client can get service from any server by sending a request over the network.

Two-tier client-server architecture

The simplest way to connect clients and servers is a two-tier architecture as shown in figure. In a two-tier architecture, any client can get service from any server by initiating a request over the network. With two tier client-server architectures, the user interface is usually located in the user's desktop and the services are usually supported by a server that is a powerful machine that can service many clients. Processing is split between the user interface and the database management server. There are a number of software vendors who provide tools to simplify development of applications for the two-tier client-server architecture.

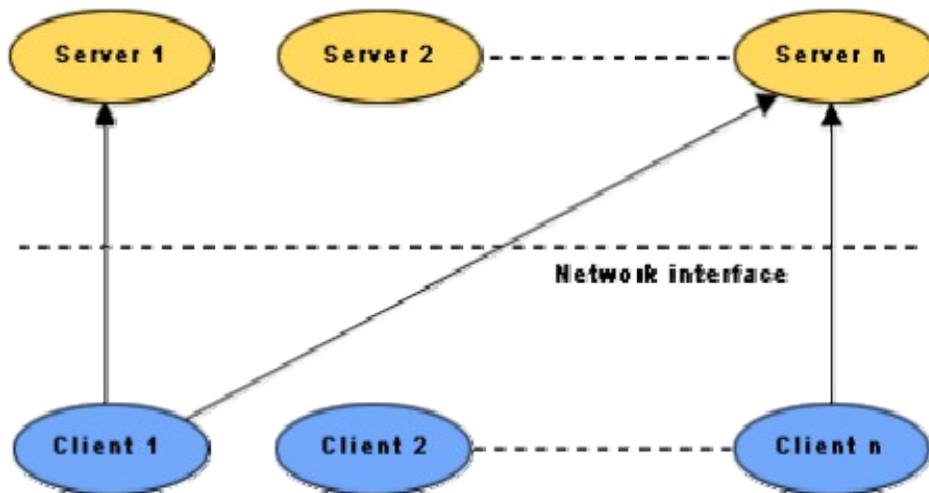


Fig: Two-tier client-server architecture

Three-tier client-server architecture

- The three-tier architecture overcomes the important limitations of the two-tier architecture. In the three-tier architecture, a middleware was added between the user system interface client environment and the server environment as shown in figure. The middleware keeps track of all server locations. It also translates client's requests into server understandable form. For example,

if the middleware provides queuing, the client can deliver its request to the middleware and disengage because the middleware will access the data and return the answer to the client.

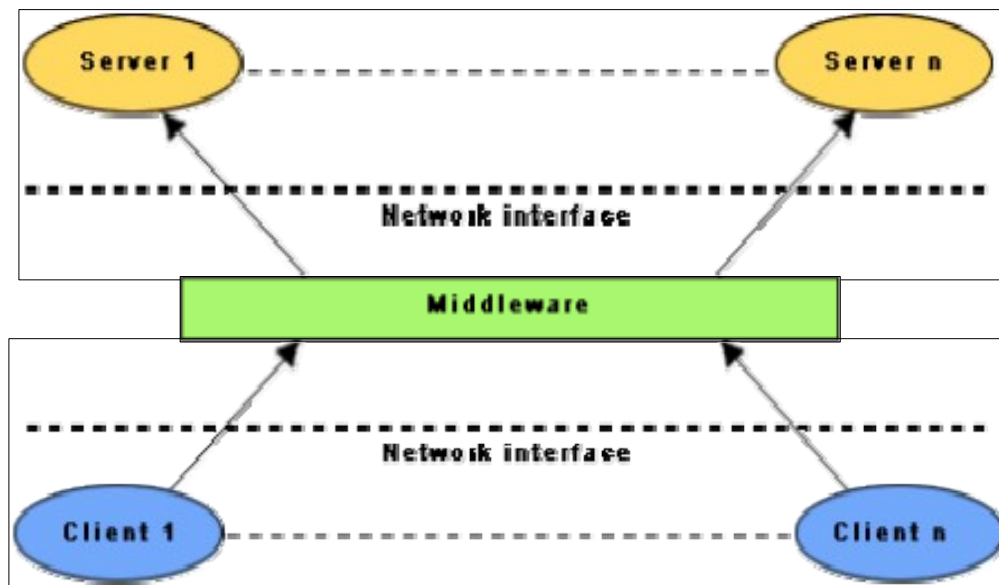


Fig: Three-tier client-server architecture

SERVICE-ORIENTED ARCHITECTURE (SOA)

Service-orientation principles have their roots in the object-oriented designing. Many claim that service-orientation will replace object-orientation; others think that the two are complementary paradigms.

SOA views software as providing a set of services. Each service composed of smaller services. Let us first understand what are software services. Services are implemented and provided by a component for use by an application developer. A service is a contractually defined behaviour. That is, a component providing a service guarantees that its behaviour is as per the specifications. A few examples of services are the following—Filling out an on-line application, viewing an on-line bank-statement, and placing an online booking. Different services in an application communicate with each other.

SOA principally leverages the Internet and emerging the standardisations on it for interoperability among various services. An application is built using the services available on the Internet, and writing only the missing ones.

There are several similarities between services and components, which are as follows:

- **Reuse:** Both a component and a service are reused across multiple applications.
- **Generic:** The components and services are usually generic enough to be useful to a wide range of applications.
- **Composable:** Both services and components are integrated together to develop an application.
- **Encapsulated:** Both components and services are non-investigable through their interfaces.

- **Independent development and versioning:** Both components and services are developed independently by different vendors and also continue to evolve independently.
- **Loose coupling:** Both applications developed using the component paradigm and the SOA paradigm have loose coupling inherent to them.

SOFTWARE AS A SERVICE (SAAS)

Owning software is very expensive. For example, a Rs. 50 Lakh software running on an Rs. 1 Lakh computer is common place. As with hardware, owning software is the current tradition across individuals and business houses. Most of IT budget now goes in supporting the software assets. The support cost includes annual maintenance charge (AMC), keeping the software secure and virus free, and taking regular back-ups, etc. But, often the usage of a specific software package does not exceed a couple of hours of usage per week. In this situation, it would be economically worthwhile to pay per hour of usage. This would also free the user from the botherance of maintenance, upgradation, backup, etc. This is exactly what is advocated by SaaS.

SaaS shifts “ownership” of the software from the customer to a service provider. Software owner provides maintenance, daily technical operation, and support for the software. Services are provided to the clients on amount of usage basis. The service provider is a vendor who hosts the software and lets the users execute on-demand charges per usage units. It also shifts the responsibility for hardware and software management from the customer to the provider. The cost of providing software services reduces as more and more customers subscribe to the service. Elements of outsourcing and application service provisioning are implicit in the SaaS model. Also, it makes the software accessible to a large number of customers who cannot afford to purchase the software outright. Target the “long tail” of small customers.