UNIT I: - Packet Switching Networks

What is a Computer Network?

- **Communication Networks**: “Sets of nodes that are interconnected to allow the exchange of information such as voice, sound, graphics, pictures, video, text, data, etc…”
- **Telephone Networks**: “The first well established and most widely used communication networks which are used for voice transmission”
  - Telephone networks originally used analog transmission as a transmission technology for the information. However, digital transmission started to evolve replacing a lot of the analog transmission techniques used in telephone networks.
- **Computer Networks**: “Collection of autonomous computers interconnected by a technology to allow exchange of information”

A network is a series of connected devices. Whenever we have many devices, the interconnection between them becomes more difficult as the number of devices increases. Some of the conventional ways of interconnecting devices are

- a. Point to point connection between devices as in mesh topology.
- b. Connection between central device and every other device – as in star topology
- c. Bus topology—not practical if the devices are at greater distances.

The solution to this interconnectivity problem is switching. A switched network consists of a series of interlinked nodes called switches. A switch is a device that creates temporary connections between two or more systems. Some of the switches are connected to end systems (computers and telephones) and others are used only for routing.

**Taxonomy of switched networks**

![Taxonomy of switched networks](image-url)

**Circuit switching**

- Traditional telephone networks operate on the basis of circuit switching
- In conventional telephone networks, a circuit between two users must be established for a communication to occur
- Circuit switched networks requires resources to be reserved for each pair of end users
- The resources allocated to a call cannot be used by others for the duration of the call
The reservation of the network resources for each user results in an inefficient use of bandwidth for applications in which information transfer is bursty or if the information is small.

Packet Switching
- Packet switched networks are the building blocks of computer communication systems in which data units known as packets flow across the networks.
- It provides flexible communication in handling all kinds of connections for a wide range of applications e.g. telephone calls, video conferencing, distributed data processing etc...
- Packet switched networks with a unified, integrated data infrastructure known as the Internet can provide a variety of communication services requiring different bandwidths.
- To make efficient use of available resources, packet switched networks dynamically allocate resources only when required.
- The form of information in packet switched networks is always digital bits.

Differences between Circuit Switching and Packet Switching

<table>
<thead>
<tr>
<th>Circuit switching</th>
<th>Packet switching</th>
</tr>
</thead>
<tbody>
<tr>
<td>1. Call set up is required.</td>
<td>1. Call setup is not required.</td>
</tr>
<tr>
<td>2. Dedicated connection between two Hosts.</td>
<td>2. No dedicated connection between two Hosts.</td>
</tr>
<tr>
<td>3. Connection/Communication is lost, if any link in the path between the Hosts is broken.</td>
<td>3. Connection/Communication could continue between the Hosts since data have many routes between the Hosts.</td>
</tr>
<tr>
<td>4. Information take the same route between the connected Hosts</td>
<td>4. Information could take different routes to reach the destination Host.</td>
</tr>
<tr>
<td>5. Information always arrives in order.</td>
<td>5. Information could arrive out of order to the destination</td>
</tr>
<tr>
<td>6. Bandwidth available is fixed.</td>
<td>6. Bandwidth available is variable.</td>
</tr>
<tr>
<td>7. Congestion is call based.</td>
<td>7. Congestion is packet based.</td>
</tr>
<tr>
<td>10. It is Transparent.</td>
<td>10. Not transparent.</td>
</tr>
<tr>
<td>11. Charging is time based.</td>
<td>11. Charging is packet based.</td>
</tr>
</tbody>
</table>

Packet networks can be viewed from two perspectives:
- External view of network :- It is Concerned with the services that the network provides to the transport layer
- Internal operation of the network.

Network services and internal network operation

Essential function of network:
- The essential function of network is to transfer information among the users that are attached to the network.
- Transfer of information may be single block of information or sequence of blocks as shown in below figure.
• In case of single block of information, we are interested in having the block delivered correctly to destination and also interested in delay experienced in traversing the network.
• In case of sequence of blocks, we are interested not only in receiving the blocks correctly and in right sequence.

Network service can be **Connection-oriented service** or **connectionless service**

**Connectionless service:**
• Connectionless service is simple with two basic interactions (1) a request to network layer that it send a packet (2) an indication from the network layer that a packet has arrived
• It puts total responsibility of error control, sequencing and flow control on the end system transport layer

**Connection-oriented service**
• The Transport layer can not request transmission of information until a connection is established between end systems
• Network layer must be informed about the new flow
• Network layer maintains state information about the flows it is handling
• During connection set up, parameters related to usage and quality of services may be negotiated and network resources may be allocated
• Connection release procedure may be required to terminate the connection

*It is also possible for a network layer to provide a choice of services to the user of network like:*
• best-effort connectionless services
• Low delay connectionless services
• Connection oriented reliable stream services
• Connection oriented transfer of packets with guaranteed delay and bandwidth

<table>
<thead>
<tr>
<th>End To End argument for system design</th>
</tr>
</thead>
<tbody>
<tr>
<td>1. The end to end argument in system design state that an end-to-end function is best implemented at higher level than at a lower level.</td>
</tr>
<tr>
<td>2. The reason is that the correct end-to-end implementation requires all intermediate low-level components to operate correctly.</td>
</tr>
<tr>
<td>3. The higher-level components at the ends are in better position to determine that a function has been carried out correctly and in better position to take corrective action if they have not.</td>
</tr>
</tbody>
</table>

➢ Keeping the core of the network simple and adding the necessary complexity at the edge enhances the scalability of the network to larger size and scope

**Internal network operation**
The fig above shows the relation between the service offered by the network and the internal network operation

- The internal operation of the network is **connectionless** if packets are transferred within the network as datagrams
- Each packets are routed independently
- Packets follow different paths from end to end and arrive out of order

- The internal operation of the network is **connection-oriented** if packets follow a virtual circuit along a path that has been established from source to destination.
- Virtual circuit setup is done once, then packets are simply forwarded
- If resources are reserved then bandwidth, delay and loss guarantees are provided.

**Network layer essentials**

The Functions that need to be carried out at every node in the Network Layer are:-

- **Routing**: mechanisms for determining the set of best paths for routing packets requires the collaboration of network elements
- **Forwarding**: transfer of packets from NE inputs to outputs
- **Priority & Scheduling**: determining order of packet transmission in each NE

**Optional**: congestion control, segmentation & reassembly, security

**Packet Network Topology**

How users access packet networks?
Example -1 – Access Multiplexing

- The diagram above shows an access network with a point to point topology where computer in homes are connected to an access multiplexer located in service provider network.
- The main purpose of the access multiplexer is to combine the bursty traffic flows from individual computers into aggregated flows.
- Eg1. DSL traffic multiplexed at DSL Access Mux
- Eg2. Cable modem traffic multiplexed at Cable Modem Termination System
- Private IP addresses in Home is done using Network Address Translation (NAT).

Example -2- Campus Network

- LANs are interconnected through use of LAN switches identified by letter ‘S’ in the figure.
- Resources such as servers and databases that are primarily use are kept within the subnet. This reduces delay in accessing resources.
- Subnet has access to rest of organization through router R that access campus backbone network.
- Subnet uses campus backbone to reach outside world such as Internet through a border router.
Servers containing critical resources that are required by the entire organization are located in a data center where it can be easily maintained and security can be enforced.

Critical servers maybe provided with redundant paths to the campus backbone network.

The routers in the campus network are interconnected to form the campus backbone network.

**Example - 3 Connecting to Internet Service Provider**

- **Domain**: the routers running the same routing protocol
- **Autonomous System**: one or more domains under the single administration.
- The campus network maybe connected to the internet service provider (ISP) through one or more border routers.
- To communicate with other networks, the autonomous system must provide information about its network routes in border routers.
- The border router communicates on an interdomain level, whereas other routers operate at an intradomain level.

**Example - 4: Internet Backbone**
National ISP provides points of presence (POPs) where customer can connect to their network
- The ISP has its own national backbone network for interconnecting its POPs
- The ISPs exchange traffic at public peering points called network access points (NAPs)
- NAP is a collocated set of high-speed routers through which the routers from different ISPs exchange traffic.
- Private peering points can be used to connect ISPs to exchange traffic directly with agreement routing polices.

### Datagram and Virtual Circuits

#### Message Switching

- In message switching, a message is relayed from one switch to another until the message arrives at the destination
- A message switch operates in *store and forward* fashion (a message has to be completely received by the switch before it can be forwarded to next switch)
- At the source each message has header attached to it to provide source and destination address.
- CRC check bits are attached to detect errors
• Each switch performs an error check, and if no errors are detected, the switch examines the header to determine the next hop in the path to the destination.
• Loss of messages may occur when a switch has insufficient buffering to store the arriving message.

![Diagram showing minimum delay](image)

• The above figure shows the minimum delay that is incurred when a message is transmitted over a path that involves two intermediate switches.
• The message has to traverse the link to the first switch.
• We assume
  – the link has propagation delay in seconds,
  – the transmission time
• The message must traverse the link that connect two switches and from second switch to the destination.
• It follows that the minimum delay is \(3T + 3T\).
• In general, the delay incurred in message switching involving L hops is \(L + LT\).

**Disadvantages of message switching**
• The probability of error increases with the length of the block. Thus long messages are not desirable.
• Not suitable for interactive applications.

**Datagram or Connectionless Packet Switching**

- Messages broken into smaller units (packets)
- Source & destination addresses in packet header
- Connectionless, packets routed independently (datagram)
- When a message arrives at the packet switch, the destination address is examined to determine the next hop.
- Packet may arrive out of order
- Re-sequencing maybe required at destination.
- Pipelining of packets across network can reduce delay, increase throughput
- Lower delay than message switching, suitable for interactive traffic

**Packet Switching Delay**

Assume three packets corresponding to one message traverse same path

\[
\text{Minimum Delay} = 3\tau + 5(T/3) \quad (\text{single path assumed})
\]

- Additional queueing delays possible at each link
- Packet pipelining enables message to arrive sooner

In general the delay incurred using a datagram switch involving \( L \) hops and consisting of \( k \) packets is \( L + LP+(k-1)P \)

**Virtual – Circuit Packet Switching**
Before the transmission of packets, it involves establishment of **virtual circuit** between source and destination.

- All packets follow the **same path**
- **Abbreviated header** identifies connection on each link
- Packets queue for transmission
- **Variable bit rates possible**

### Connection Setup

- Signaling messages propagate as route is selected
- Signaling messages identify connection and setup tables in switches
- Typically a connection is identified by a local tag, **Virtual Circuit Identifier (VCI)**
- Each switch only needs to know how to relate an **incoming tag in one input to an outgoing tag** in the corresponding output
- Once tables are setup, packets can flow along path

### Connection Setup Delay
• Connection setup delay is incurred before any packet can be transferred
• Delay is acceptable for sustained transfer of large number of packets
• This delay may be unacceptably high if only a few packets are being transferred
• The minimum delay in virtual circuit packet switching is similar to that in datagram packet switching, except for an additional delay required to setup the virtual circuit.

**Cut-Through switching**

- It is the modified form of virtual circuit packet switching
- Some networks perform error checking on header only, so packet can be forwarded as soon as header is received & processed
- Delays reduced further with cut-through switching
A packet switch performs two functions 1) routing 2) forwarding.
Routing functions use algorithms to find a path to each destination and store the result in a routing table. Forwarding function processes each incoming packet from an input port and forwards the packet to an appropriate output port based on the information stored in the routing table.

- The above fig (a) shows a generic packet switch consisting of input ports, output ports, an interconnection fabric, and a switch controller.
- Input ports and output ports are normally paired.
- A line card contains several input and output ports so that the capacity of the link connecting the line card to the interconnection fabric, which is typically high-speed, is fully utilized.
- Line cards are concerned with symbol timing, line coding, framing, physical addressing, and error checking.
- The line card is made up of various chipsets as shown in fig (b).
- The programmable network processor performs packet-related tasks such as table lookup and packet scheduling.
- The controller in a packet switch contains a general-purpose processor to carry out a number of control and management functions.
- The controller also communicates with line cards and the interconnection fabric.
- The function of the Interconnection fabric is to transfer packets between the line cards.
- If there are high-speed line cards, the interconnection fabric is likely to be the bottleneck, since all traffic goes through it.
- A bus-type interconnection structure (whereby packets are transmitted serially) does not scale to large size, since the speed of the bus has to be about N times faster than the port speed.
- A crossbar interconnection fabric can transfer packets in parallel between input ports and output ports.
- The buffers need to be added to the crossbar to accommodate packet contention.
- The buffers can be located at the input ports or output ports as shown in figure below.

Only one packet is allowed to proceed to a particular output in case of input buffering. Input buffering causes a problem head-of-line (HOL) blocking.
- Consider a situation where there are two packets at the input buffer, as shown in fig. above.
- The first packet would like to go to output 3 and the second packet to output 8.
Suppose that the packet from input buffer 1 would like to go output 3 at the same time.
Suppose that the fabric arbiter decides to transmit the packet from input buffer 1. Then the first packet from input buffer 2 needs to wait until output 3 has transferred the packet from input buffer1. Meanwhile, the second packet has to wait behind the first packet even though output 8 is idle.
This results in performance degradation of crossbar with input buffering.
The problem of the first packet holding back other subsequent packets behind it is called head-of-line (HOL) blocking.
One way to eliminate HOL blocking is to provide N separate input buffers at each input port. Such an input buffer is called virtual output buffer.

Routing in Packet Networks

- **Routing**: it is concerned with determining feasible path for packets to follow from each source to destination.
- Which path is “best”? The term best depends on the objective function that network operator tries to optimize which maybe Min delay, Min hop, Max bandwidth, Min cost, Max reliability

**Goals of routing algorithm**
1. Rapid and accurate delivery of packets
2. Adaptability to changes in network topology resulting from node or link failures.
3. Adaptability to varying source-destination traffic loads
4. Ability to route packets away from congested links
5. Ability to determine the connectivity of the network
6. Ability to avoid routing loops.
7. Low overhead

Classification of routing algorithms

**Static vs Dynamic Routing**

<table>
<thead>
<tr>
<th>Static</th>
<th>Dynamic</th>
</tr>
</thead>
<tbody>
<tr>
<td>1. The paths are pre-computed by a host and are loaded into routing table.</td>
<td>1. Each node computes the best path by communicating with its neighbors.</td>
</tr>
<tr>
<td>2. The paths are fixed for long duration of time.</td>
<td>2. Each node continuously learns the state of the network by communicating with its neighbors.</td>
</tr>
<tr>
<td>3. Static routing is good when</td>
<td>3. Adapt to changes in network conditions</td>
</tr>
<tr>
<td>- network size is small,</td>
<td></td>
</tr>
<tr>
<td>- traffic load does not change variably</td>
<td>4. Scales well</td>
</tr>
<tr>
<td>- network topology is fixed</td>
<td>5. Disadvantage is added complexity in the node.</td>
</tr>
<tr>
<td>4. Does not scale to large network</td>
<td></td>
</tr>
<tr>
<td>5. Disadvantage is its inability to react rapidly to network failures.</td>
<td></td>
</tr>
</tbody>
</table>

Centralized vs Distributed Routing
Centralized Routing | Distributed Routing
--- | ---
1. Network control center computes the path and uploads the information | 1. Nodes cooperate by means of message exchanges and perform their own computations.
2. does not scale well | 2. scale well
3. consistent results | 3. Inconsistent results due to loops.
4. Problems adapting to frequent topology changes | 4. Adapts to topology and other changes

Hierarchical Routing

- The hierarchical approach reduces the size of the routing tables at the routers in assigning the addresses.
- Hosts that are near each other (i.e. a group) should have addresses that have common prefixes. The routers examine only part of the address (i.e., the prefix) to decide how a packet should be routed.
- Figure below gives an example of hierarchical address assignment and a flat address assignment.

![Hierarchical Routing and Flat Routing](image)

- In figure (a) the hosts at each of the four sites have the same prefix. Thus the two routers need only maintain tables with four entries as shown.
- On the other hand, if the addresses are not hierarchical (Figure 7.27b), then the routers need to maintain 16 entries in their routing tables.

ROUTING TABLES
Routing Table for Virtual circuit networks

- virtual circuit identifier determines the destination
Specialized Routing

Flooding

- **Principle of flooding**: a node (or a packet switch) forwards an incoming packet to all ports except to the one it arrived on.
- Each node (a switch) performs the flooding process such that the packet will reach the destination as long as at least one path exists between the source and the destination.
- Flooding is a useful
  - when the information in the routing tables is not available, such as during system startup,
  - when survivability is required, such as in military networks.
  - when the source needs to send a packet to all hosts connected to the network (i.e., broadcast delivery).
Flooding generates vast numbers of duplicate packets.

In figure, initially one packet arriving at node 1 triggers three packets to nodes 2, 3, and 4. In the second phase nodes 2, 3, and 4 send two, two, and three packets respectively. These packets arrive at nodes 2 through 6. In the third phase 15 more packets are generated giving a total of 25 packets after three phases.

The flooding needs to be controlled so that packets are not generated excessively.

How to control this?

There are three methods to reduce the resource consumption in the network:

1) **Use a time-to-live (TTL) field in each packet.**
   - When the source sends a packet, the time-to-live field is initially set to some small number.
   - Each node decrements the field by one before flooding the packet. If the value reaches zero, the switch discards the packet.
   - To avoid unnecessary waste of bandwidth, the time-to-live should ideally be set to the minimum hop number between two furthest nodes (called the diameter of the network).

2) **Add an identifier before flooding**
   - Every node adds an identifier before flooding.
   - When a node identifies a packet that contains the identifier of the switch, it discards the packet.
   - This method effectively prevents a packet from going around a loop.

3) **Have a unique sequence number**
   - Each packet from the given source is uniquely identified with a sequence number.
• When a node receives a packet, it records the source address and the sequence number.
• If node discovers that packet has already visited the node, it will discard the packet.

**Deflection Routing**

• Deflection routing was first called as Hot-potato routing.
• It requires the network to provide multiple paths for each source-destination pair.
• Each switch first tries to forward a packet to the preferred port. If the preferred port is busy or congested, the packet is deflected to another port.
• Deflection routing works well in a regular topology.

**Example:** Manhattan Street network:

• Each column represents an avenue, and each row represents a Street.
• Each switch is labeled (i,j) where i denotes the row number and j denotes the column number.
• The links have directions that alternate for each column or row.
• If switch (0,2) would like to send a packet to switch (1,0), the packet could go two left and one down. However, if the left port of switch (0,1) is busy (see Figure), the packet will be deflected to switch (3,1). Then it can go through switches (2,1), (1,1), (1,2), (1,3) and eventually reach the destination switch (1,0).
• One advantage of deflection routing is that the switch can be bufferless, since packets do not have to wait for a specific port to become available. Since packets can take alternative paths, deflection routing cannot guarantee in-sequence delivery of packets.
• Deflection routing is used to implement many high-speed packet switches where the topology is very regular and high-speed buffers are relatively expensive compared to deflection routing logic.

**Figure:** Manhattan street network

**Shortest Path Routing**
Shortest path algorithms are used to determine the shortest path based on following metrics.

The shortest path routing are Bellman-Ford algorithm and Dijkstra’s algorithm

- Possible metrics
  - Hop count
  - Reliability:
  - Delay: sum of delays along path
  - Bandwidth: “available capacity” in a path
  - Load: Link & router utilization along path
  - Cost: $$$

**Bellman-Ford Algorithm or Ford-Fulkerson algorithm**

**Principle:** If each neighbor of node A knows the shortest path to node Z, then node A can determine its shortest path to node Z by calculating the cost/distance to node Z through each of its neighbors and picking the minimum.

**Bellman-Ford algorithm**

- Consider computations for one destination $d$
- Initialization
  - Each node table has 1 row for destination $d$
  - Distance of node $d$ to itself is zero: $D_d=0$
  - Distance of other node $j$ to $d$ is infinite: $D_j=\mu$, for $j \neq d$
  - Next hop node $n_j = -1$ to indicate not yet defined for $j \neq d$
- Send Step
  - Send new distance vector to immediate neighbors across local link
- Receive Step
  - At node $j$, find the next hop that gives the minimum distance to $d$,
    - Minj \{ Cij + Dj \}
  - Replace old $(n_j, D_j(d))$ by new $(n_j^*, D_j^*(d))$ if new next node or distance
  - Go to send step

For the above figure **Apply Bellman-ford algorithm** to find both minimum cost from each node to the destination node 6 and the next node along the shortest path.

Each node I maintains an entry $(n,D_i)$, where $n$ is the next node along the current shortest path and $D_i$ is the current minimum cost from node $i$ to destination.

Initially all nodes, other than the destination node 6, are at infinite cost (distance) to node 6. Node 6 informs its neighbors it is distance 0 from itself.

**Iteration 1:** Node 3 finds that it is connected to node 6 with cost of 1. Node 5 finds it is connected to node 6 at a cost of 2. nodes 3 and 5 update their entries and inform their neighbors.

**Iteration 2:** Node 1 finds it can reach node 6, via node 3 with cost 3. Node 2 finds it can reach node 6, via node 5 with cost 6. Node 4 finds it has paths via nodes 3 and 5,
with costs 3 and 7 respectively. Node 4 selects the path via node 3. Nodes 1, 2, and 4 update their entries and inform their neighbors.

**Iteration 3:** Node 2 finds that it can reach node 6 via node 4 with distance 4. Node 2 changes its entry to (4,4) and informs its neighbors.

**Iteration 4:** nodes 1, 4, and 5 process the new entry from node 2 but do not find any new shortest paths. The algorithm has converged.

<table>
<thead>
<tr>
<th>Iteration</th>
<th>Node 1</th>
<th>Node 2</th>
<th>Node 3</th>
<th>Node 4</th>
<th>Node 5</th>
</tr>
</thead>
<tbody>
<tr>
<td>Initial</td>
<td>(-1, ¥)</td>
<td>(-1, ¥)</td>
<td>(-1, ¥)</td>
<td>(-1, ¥)</td>
<td>(-1, ¥)</td>
</tr>
<tr>
<td>1</td>
<td>(-1, ¥)</td>
<td>(-1, ¥)</td>
<td>(6, 1)</td>
<td>(-1, ¥)</td>
<td>(6, 2)</td>
</tr>
<tr>
<td>2</td>
<td>(3,3)</td>
<td>(5,6)</td>
<td>(6, 1)</td>
<td>(3,3)</td>
<td>(6, 2)</td>
</tr>
<tr>
<td>3</td>
<td>(3,3)</td>
<td>(4,4)</td>
<td>(6, 1)</td>
<td>(3,3)</td>
<td>(6, 2)</td>
</tr>
</tbody>
</table>

**What happens when a link fails?**

<table>
<thead>
<tr>
<th>Iteration</th>
<th>Node 1</th>
<th>Node 2</th>
<th>Node 3</th>
<th>Node 4</th>
<th>Node 5</th>
</tr>
</thead>
<tbody>
<tr>
<td>1</td>
<td>(3,3)</td>
<td>(4,4)</td>
<td>(4, 5)</td>
<td>(3,3)</td>
<td>(6,2)</td>
</tr>
<tr>
<td>2</td>
<td>(3,7)</td>
<td>(4,4)</td>
<td>(4, 5)</td>
<td>(2,5)</td>
<td>(6,2)</td>
</tr>
<tr>
<td>3</td>
<td>(3,7)</td>
<td>(4,6)</td>
<td>(4, 7)</td>
<td>(5,5)</td>
<td>(6,2)</td>
</tr>
<tr>
<td>4</td>
<td>(2,9)</td>
<td>(4,6)</td>
<td>(4, 7)</td>
<td>(5,5)</td>
<td>(6,2)</td>
</tr>
</tbody>
</table>

**Counting to Infinity Problem**
Nodes believe best path is through each other (Destination is node 4)

<table>
<thead>
<tr>
<th>Update</th>
<th>Node 1</th>
<th>Node 2</th>
<th>Node 3</th>
</tr>
</thead>
<tbody>
<tr>
<td>Before break</td>
<td>(2,3)</td>
<td>(3,2)</td>
<td>(4, 1)</td>
</tr>
<tr>
<td>After break</td>
<td>(2,3)</td>
<td>(3,2)</td>
<td>(2,3)</td>
</tr>
<tr>
<td>1</td>
<td>(2,3)</td>
<td>(3,4)</td>
<td>(2,3)</td>
</tr>
<tr>
<td>2</td>
<td>(2,5)</td>
<td>(3,4)</td>
<td>(2,5)</td>
</tr>
<tr>
<td>3</td>
<td>(2,5)</td>
<td>(3,6)</td>
<td>(2,5)</td>
</tr>
<tr>
<td>4</td>
<td>(2,7)</td>
<td>(3,6)</td>
<td>(2,7)</td>
</tr>
<tr>
<td>5</td>
<td>(2,7)</td>
<td>(3,8)</td>
<td>(2,7)</td>
</tr>
<tr>
<td>...</td>
<td>...</td>
<td>...</td>
<td>...</td>
</tr>
</tbody>
</table>

**Problem:** Bad News Travels Slowly  
Remedies  
- Split Horizon  
  - Do not report route to a destination to the neighbor from which route was learned  
- Poisoned Reverse  
  - Report route to a destination to the neighbor from which route was learned, but with infinite distance  
  - Breaks erroneous direct loops immediately  
  - Does not work on some indirect loops  

**Split Horizon with Poison Reverse**
Nodes believe best path is through each other

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<td>(2, 3)</td>
<td>(3, 2)</td>
<td>(4, 1)</td>
</tr>
<tr>
<td>After break</td>
<td>(2, 3)</td>
<td>(3, 2)</td>
<td>(-1, ¥)</td>
</tr>
</tbody>
</table>

Node 2 advertizes its route to 4 to node 3 as having distance infinity; node 3 finds there is no route to 4.

Node 1 advertizes its route to 4 to node 2 as having distance infinity; node 2 finds there is no route to 4.

Node 1 finds there is no route to 4.

**Source Routing**

- In source routing, path to the destination is determined by the source.
- Source routing works in either datagram or virtual-circuit packet switching.
- Before sending packet, the source has to know the path to the destination in order to include the path information in the packet header.
- The path information contains the sequence of nodes to traverse and should give sufficient information to intermediate node to forward the packet.
- The below fig. shows how source routing works in datagram network.
- Intermediate switch reads header and removes the address identifying the node and forwards the packets to next node.
- Source host selects path that is to be followed by a packet
  - Strict: Here, the source specifies the address of each node along the path to the destination.
  - Loose: When source knows partial information of network topology, the source can use loose source routing.

**Packet Switching Networks-1**
1. What are the different network services provided by network layer of the packet switching networks? (Aug 06, 10M)

2. Why is packet switching more suitable than message switching for interactive applications? Compare the delays in datagram packet switching and message switching. (Jan 10, 6M) (Dec 12 10 M)

3. Compare the Bellman –Ford algorithm and Dijkstra’s algorithm for finding the shortest paths from a source node to all other nodes in a network. (Jan 10, 6M)

4. Differentiate between virtual circuits and datagram subnets. (July 09, 10M) (AUG 02, Feb 06 6M) (July 07, 5M)

5. Define routing and forwarding. What are the goals of routing algorithm? (AUG 05, 6M)

6. Using Dijkstra’s algorithm find the shortest path between A and D

   \[ \begin{array}{cccc}
   A & 6 & 1 & 4 & 2 \\
   6 & 2 & 5 & 2 & 2 \\
   G & E & F & D & H \\
   
   \end{array} \]

   (Feb 06, 6M) (July 07, 5M)

7. Explain hierarchical routing. (Feb 05, 6M) (July 07, 5M)

8. What are the drawbacks of flooding? (Aug 06, 4M)
   Define routing. Explain Bellman-Ford routing algorithm with necessary illustration. What are the drawbacks of this algorithm? (Aug 06, 10M) (June 12 10 M)

9. Good news spreads fast; bad news propagates slowly in bellman ford algorithm. Explain with an example. (July 07, 5M)

10. Explain Count to infinity problem. (July 06, 6M)
11. Consider the network in fig

(i) Use the Dijkstra’s algorithm to find the set of shortest path from node 4 to other node.
(ii) Find the set of associated routing table entries
(iii) Find the shortest path from node 5 to other destination node and find the shortest path tree from 5 to other nodes. (Dec 12 10 M)

(July 09, 10M)(June/July 11 5M)

12. Suppose that 64kbps PCM coded speech is packetized into a constant bit rate ATM cell stream.

i) What is the interval between production of full cells?
ii) How long does it take to transmit the cell at 155Mbps?
iii) How many cells could be transmitted in this system between consecutive voice cells? (Jan 10, 6M)

13. Differentiate between virtual circuit and datagram’s. Explain routing table for both. (Dec 10, 10 M)

14. Differentiate between connectionless packet switching and virtual packet switching. (June/July 11 8 M)

15. Explain briefly the structure of a generic packet switch, with the help of the diagram (June/July 11 7 M)

16. With examples, Differentiate between datagram, virtual circuits and packet switching. (June 12 6 M)

17. Write a short note on ATM networks (June 12 4 M)

18. Explain Dijkstra’s algorithm. Solve the following problem (Dec 10 10 M)
19. Consider the network given below use Dijkstra’s algorithm to find shortest path from all nodes to destination node 2.

**UNIT II- Packet Switching Networks-2**

**Link-State Algorithm**

- Basic idea: two stage procedure
- Each source node gets a map of all nodes and link metrics (link state) of the entire network
  - Learning who the neighbors are and what are delay to them
  - Construct a link state packet, and deliver it to other
- Find the shortest path on the map from the source node to all destination nodes;
  - Dijkstra’s algorithm Broadcast of link-state information
  - Every node i in the network broadcast to every other node in the network:
    - It requires each link cost to be positive
    - The main idea is to progressively identify the closest nodes from source node in order of increasing path cost
    - The algorithm is iterative.
    - The algorithm can be implemented by maintaining a set of N permanently labeled nodes, which consists of those nodes whose shortest paths have been determined.
    - At each iteration the next closest node is added to the set N and the distance to the remaining nodes via the new node is evaluated.

**Dijkstra Algorithm:**
**Finding shortest paths in order**

- \( N \): set of nodes for which shortest path already found
Initialization: (Start with source node s)
- \( N = \{s\}, \text{ } Ds = 0, \text{ } \text{“s is distance zero from itself”}\)
- \( Dj=Csj \text{ for all } j \neq s, \text{ distances of directly-connected neighbors} \)

Step A: (Find next closest node i)
- Find \( i \in N \) such that
- \( Di = \min Dj \text{ for } j \in N \)
- Add i to N
- If N contains all the nodes, stop

Step B: (update minimum costs)
- For each node \( j \in N \)
- \( Dj = \min (Dj, Di+Cij) \)
- Go to Step A

**Execution of Dijkstra's algorithm**

<table>
<thead>
<tr>
<th>Iteration</th>
<th>N</th>
<th>D2</th>
<th>D3</th>
<th>D4</th>
<th>D5</th>
<th>D6</th>
</tr>
</thead>
<tbody>
<tr>
<td>Initial</td>
<td>{1}</td>
<td>3</td>
<td>2</td>
<td>5</td>
<td>(\mu)</td>
<td>(\mu)</td>
</tr>
<tr>
<td>1</td>
<td>{1,3}</td>
<td>3</td>
<td>2</td>
<td>(\mu)</td>
<td>(\mu)</td>
<td>3</td>
</tr>
<tr>
<td>2</td>
<td>{1,2,3}</td>
<td>3</td>
<td>2</td>
<td>4</td>
<td>7</td>
<td>3</td>
</tr>
<tr>
<td>3</td>
<td>{1,2,3,6}</td>
<td>3</td>
<td>2</td>
<td>4</td>
<td>5</td>
<td>3</td>
</tr>
<tr>
<td>4</td>
<td>{1,2,3,4,6}</td>
<td>3</td>
<td>2</td>
<td>4</td>
<td>5</td>
<td>3</td>
</tr>
<tr>
<td>5</td>
<td>{1,2,3,4,5,6}</td>
<td>3</td>
<td>2</td>
<td>4</td>
<td>5</td>
<td>3</td>
</tr>
</tbody>
</table>

**Shortest Paths in Dijkstra’s Algorithm**

**Why is Link State Better?**
- Fast, loopless convergence
- Support for precise metrics, and multiple metrics if necessary (throughput, delay, cost, reliability)
- Support for multiple paths to a destination
  - algorithm can be modified to find best two paths

**Distance vector v/s Link state**

<table>
<thead>
<tr>
<th>Functionality</th>
<th>Distance vector</th>
<th>Link State</th>
</tr>
</thead>
<tbody>
<tr>
<td>Primary principle</td>
<td>Sends larger updates, about the complete network information only to neighboring routers</td>
<td>Sends smaller updates, about the link state of neighbors, to all routers</td>
</tr>
</tbody>
</table>
Assignment Questions

1. Why is packet switching more suitable than message switching for interactive applications? Compare the delays in datagram packet switching and message switching.  
   
   (Jan 10, 6M) (Dec 12 10 M)

2. Differentiate between virtual circuits and datagram subnets.  
   
   (July 09, 10M)(AUG 02,Feb 06 6M) (July 07, 5M)

3. Define routing. Explain Bellman-Ford routing algorithm with necessary illustration. What are the drawbacks of this algorithm?  
   
   (Aug 06, 10M) (June 12 10 M)

4. Explain Bellman-Ford routing algorithm with necessary illustration. What are the drawbacks of this algorithm?  
   
   (Aug 06, 10M) (June 12 10 M)

   Find the shortest path for below diagrams. Destination node is E for Fig1 and Destination node is 6 for Fig2
5. Using Dijkstra’s algorithm find the shortest path between A and D.

6. Find the shortest path between A and E for the below diagram.

6. Compare the Bellman–Ford algorithm and Dijkstra’s algorithm for finding the shortest paths from a source node to all other nodes in a network.

7. Good news spreads fast; bad news propagates slowly in Bellman Ford algorithm. Explain with an example.