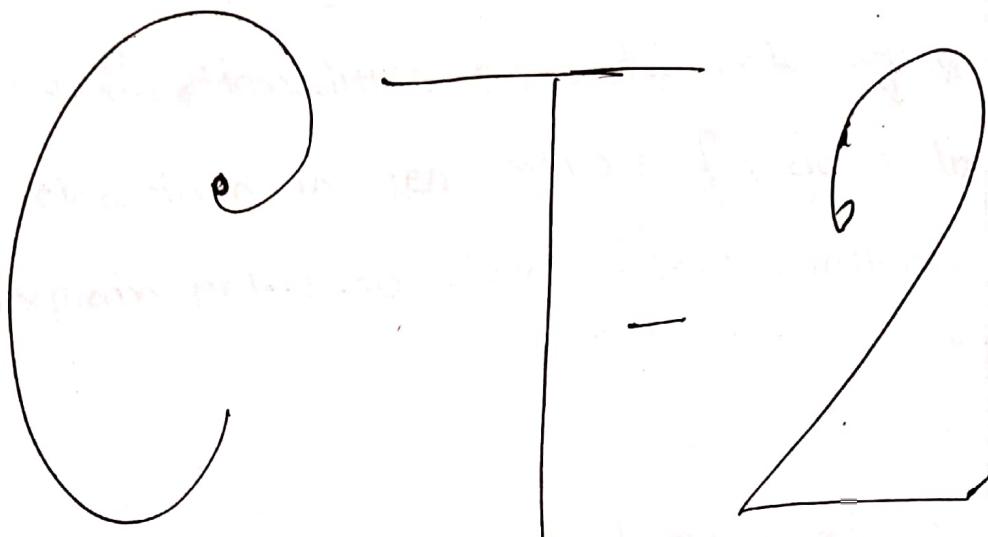


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Name : Moniruzzaman

ID : IT-18007

1

Q Define data link layer. What are its sublayers? 3

b. Explain the functionalities of data link layer. 5

c. Which mechanism is responsible for data link

layer? Explain point-to-point flow control. 6

2

Q Briefly explain the error control mechanism. 4

b. How many techniques are available in Data-link layer to control the errors by ARQ? 6

c. How many ways may involve in error control mechanism? List the types of errors in data transmission. 4

3

Q Briefly explain how to control errors? 7

b. List the network layer functionalities. 5

c. How many different kinds of network addresses in existence? 2

Sub :

4]

Define Network Layer.

@ Define Network Layer and its features. 6

@ Briefly explain Network Addressing. 6

@ Write down the protocols of network layer. 2  
Protocols used in network layer are IP, ICMP, ARP, RARP, etc.

5]

@ Define Network Routing. 2

@ Explain different types of routing. 6

@ Explain different routing protocols. 6

6]

Q DSA for answer go to last of page

@ What is tunneling? 3

@ Explain Packet fragmentation. 5

@ What does Internet Protocol version 4 (IPv4) means? Explain. 6

7]

Q answer last of page

@ What are routing algorithms? 3

@ What does Internet Control Message Protocol (ICMP) mean? Explain. 6

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Q) What does Internet protocol version 6 (IPv6)  
mean? Explain.

It is a new version of IP which has 128 bits to represent IP address instead of 32 bits.

It also supports 2^128 addresses which is approximately 3.4 x 10^38.

a) Explain the TCP/IP model/layers and answer.

b) How TCP/IP works?

c) Write down the importance of TCP/IP.

It is important for communication between different hosts.

It provides a standard way of communication between different hosts.

It also provides a standard way of communication between different hosts.

It is a reliable protocol which ensures delivery of data.

It is a connection-oriented protocol which means it establishes a connection before sending data.

It is a connectionless protocol which means it sends data without establishing a connection.

It is a stateless protocol which means it does not keep track of previous connections.

It is a connectionless protocol which means it sends data without establishing a connection.

It is a stateless protocol which means it does not keep track of previous connections.

It is a connectionless protocol which means it sends data without establishing a connection.

It is a stateless protocol which means it does not keep track of previous connections.

(Q1) a Answer to the question no A-1 ~~as per book~~ (b)

(a)

Data link layer is second layer of OSI Layered Model. This layer is one of the most complicated layers as it has complex functionalities and responsibilities.

Model, This layer is one of the most complicated layers as it has complex functionalities and responsibilities. Data link layer hides the details of underlying hardware and represents itself to upper layer as a simple interface.

Data link layer works between two hosts which are directly connected in some manner. This direct connection could be point to point or broadcast. Systems on broadcast network are said to be on same link. The work of data link layer tends to get more complex when it is dealing with multiple hosts on single collision domain.

Data link layer has two sub-layers:

□ Logical Link Control: It deals with protocols,

flow-control, and error control.

□ Media Access Control: It deals with actual

control of medium. How will one

get start of message or message

Answer to the que no: 01

Ans from Abhishek Kumar: Parab. No. 1

(b) What are functions of data link layer?

Data link layer does many tasks on behalf of upper layers. There are:

\* Framing: Data link layer takes packet from Network layer and Encapsulates them into frames. Then it sends each frame bit-by-bit. On the hardware.

at receiver end, data link layer picks up signals from hardware and assembles them into frames.

\* Addressing: Data-link layer provides layer-2 hardware with its own MAC address. It uses hardware addressing mechanism. Hardware

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address is assumed to be unique on the link.

It is encoded into hardware at the time of manufacturing.

\* Synchronization: When data frames are sent on the link, both machines must be synchronized in order to transfer to take place.

\* Error Control: Sometimes signals may have encountered problem in transmission and the bits are flipped. These errors are detected and attempted to recover actual data bits. It also provides error reporting mechanism to the sender.

\* Flow control: Stations on same link may have different speed or capacity. Data link layer ensures flow control that enables both machine to exchange data on same speed.

\* Multi-Access: When more than one station want to send data on the shared medium, then access to the medium is controlled by some rules.

link tries to transfer the data, it has a high probability of collision. Data-link layer provides mechanism such as CSMA/CD to equip capability of accessing a shared media among multiple systems.

Answer to the que. no: 01

Ques. Ques. 1. Explain the function of Data Link Layer.

(C)

Data link layer is responsible for converting data bits from receiver into stream of bits over the underlying hardware. At the receiving end, Data link layer picks up data from hardware which are in the link-layer form of electrical signals, assembles them in a recognizable frame format, and hands over to upper layers. Data-link layer is responsible for implementation of point-to-point flow and error control mechanism.

Flow control: When a data frame (FCS+2 bytes) is sent from one host to another over a single shared link, it is required that the sender and receiver should work at the same speed. That is, sender sends at a speed on which the receiver can process and accept the data. What if the speed (hardware / software) of the sender or receiver differs? If sender is sending too fast the receiver may be overloaded, (swamped) and data may be lost. Two types of mechanisms can be deployed to control the flow: Stop and wait. This flow control mechanism forces the sender after transmitting a data frame to stop and wait until the acknowledgement of the data frame sent is received.

\* Sliding window: In this flow control mechanism, both sender and receiver agree on the number of data frames after which the acknowledgement should be sent. As we learnt, stop and wait flow control mechanism wastes resources, this protocol tries to make use of underlying resources as much as possible.

### Answer to the question no. 02

a) Error control: When data frame is transmitted, there is probability that data frame may be lost in the transit or it is received corrupted. In both cases, the receiver does not receive the correct data frame and sender does not know anything about any loss. In such case both sender and receiver are equipped with some protocol which helps them to detect & transmit errors such as

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loss of data-frame. Hence, either the sender retransmits the data-frame or the receiver may request to resend the previous data-frame.

Requirements for error control mechanism:-

\* Error detection: The sender and receiver, either both or any, must ascertain that there is some error in the transmit get of message.

\* Positive Ack: When the receiver receives a correct frame, it should acknowledge it.

\* Negative Ack: when the receiver receives a damage frame or a duplicate frame, it

sends a NACK back to the sender and the sender must retransmit the correct frame.

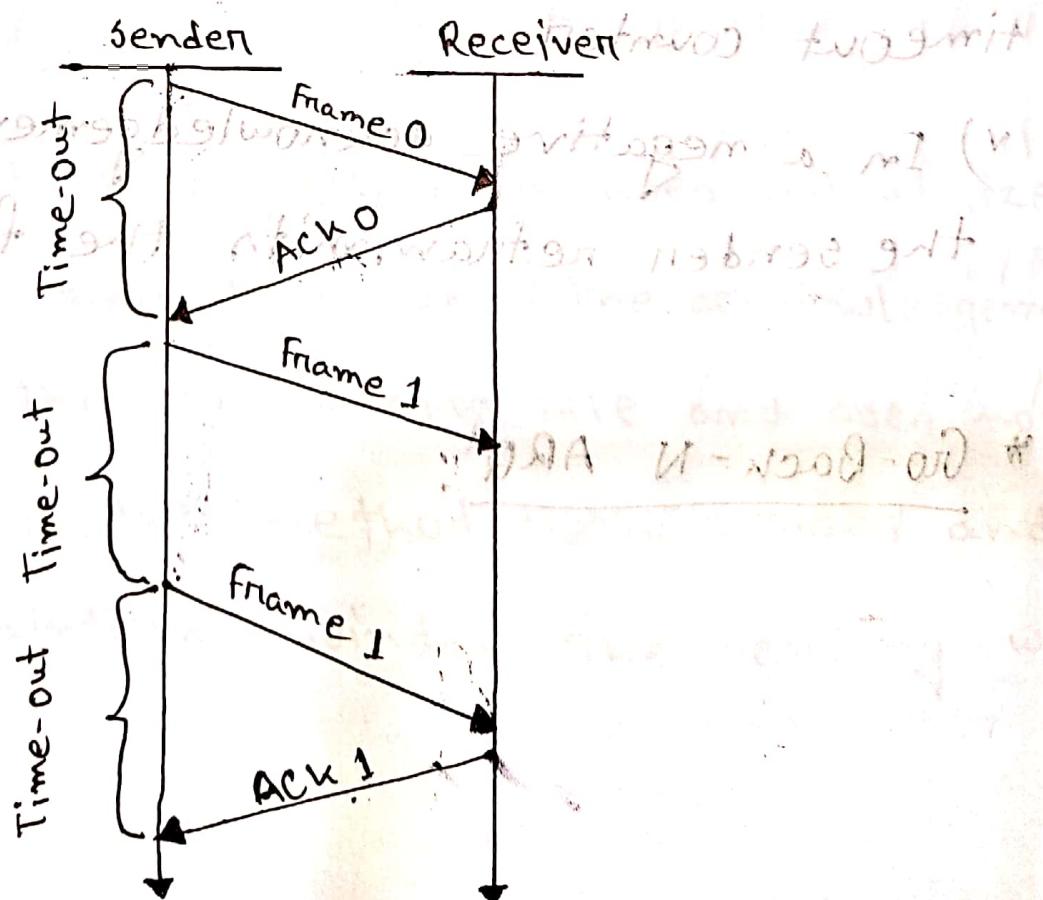
\* Retransmission: The sender maintains a clock and sets a timeout period. If an acknowledgement

of a data frame previously transmitted does not arrive before the timeout the sender transmits the frame again, thinking that the frame or its acknowledgement is lost in transmit.

### Ans. to the que no: 02

There are three types of techniques available which Data-Link layer may deploy to control the errors by Automatic Repeat Requests (ARQ).

#### \* Stop-and-wait ARQ:



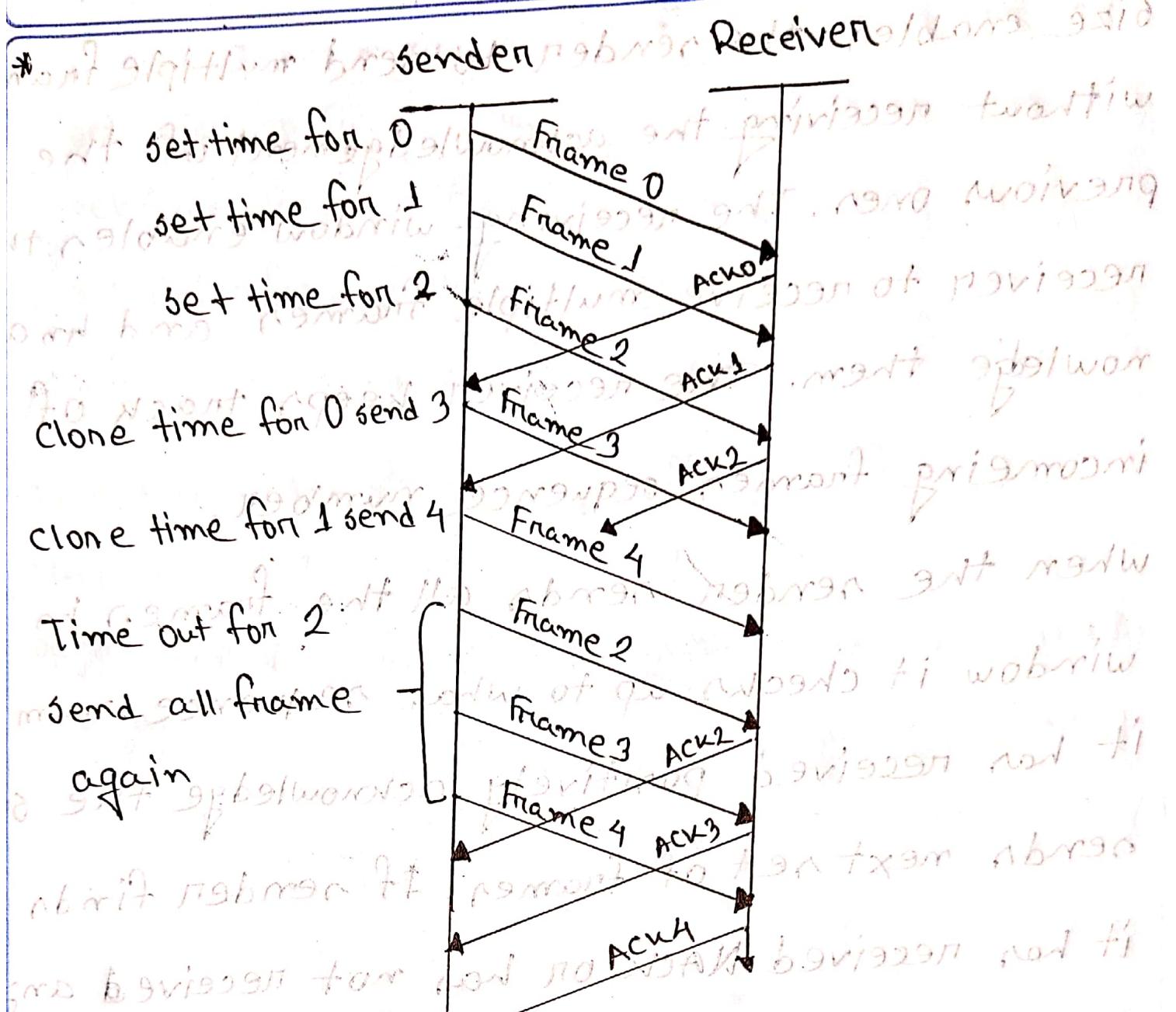
The following transition may occur in Stop-and-wait ARQ:

and transmission of frame and reception of acknowledgement

- i) The sender maintains a timeout counter.
- ii) When a frame is sent, the sender starts the timeout counter.
- iii) If acknowledgement of frame comes in time, the sender transmits the next frame in queue.
- iv) If acknowledgement does not come in time, the sender assumes that either the frame or its acknowledgement is lost in transit. The sender retransmits the frame and starts the timeout counter.
- v) In a negative acknowledgement is received, the sender retransmits the frame.

### \* Go-Back-N ARQ:

Sub:



stop and wait ARQ mechanism does not utilize resources at their best. When the acknowledgement is received, the sender sits idle and does nothing. In go-back-N ARQ method, both sender and receiver maintain a window. The sending-win-

size enables the sender to send multiple frames without receiving the acknowledgement of the previous ones. The receiving window enables the receiver to receive multiple frames and acknowledge them. The receiver keeps track of incoming frame sequence number. When the sender sends all the frames in window it checks up to what sequence number it has received positively acknowledged. If sender finds that it has received NACK or has not received any ACK for a particular frame, it retransmits all the frames after which it does not receive any positive ACK.

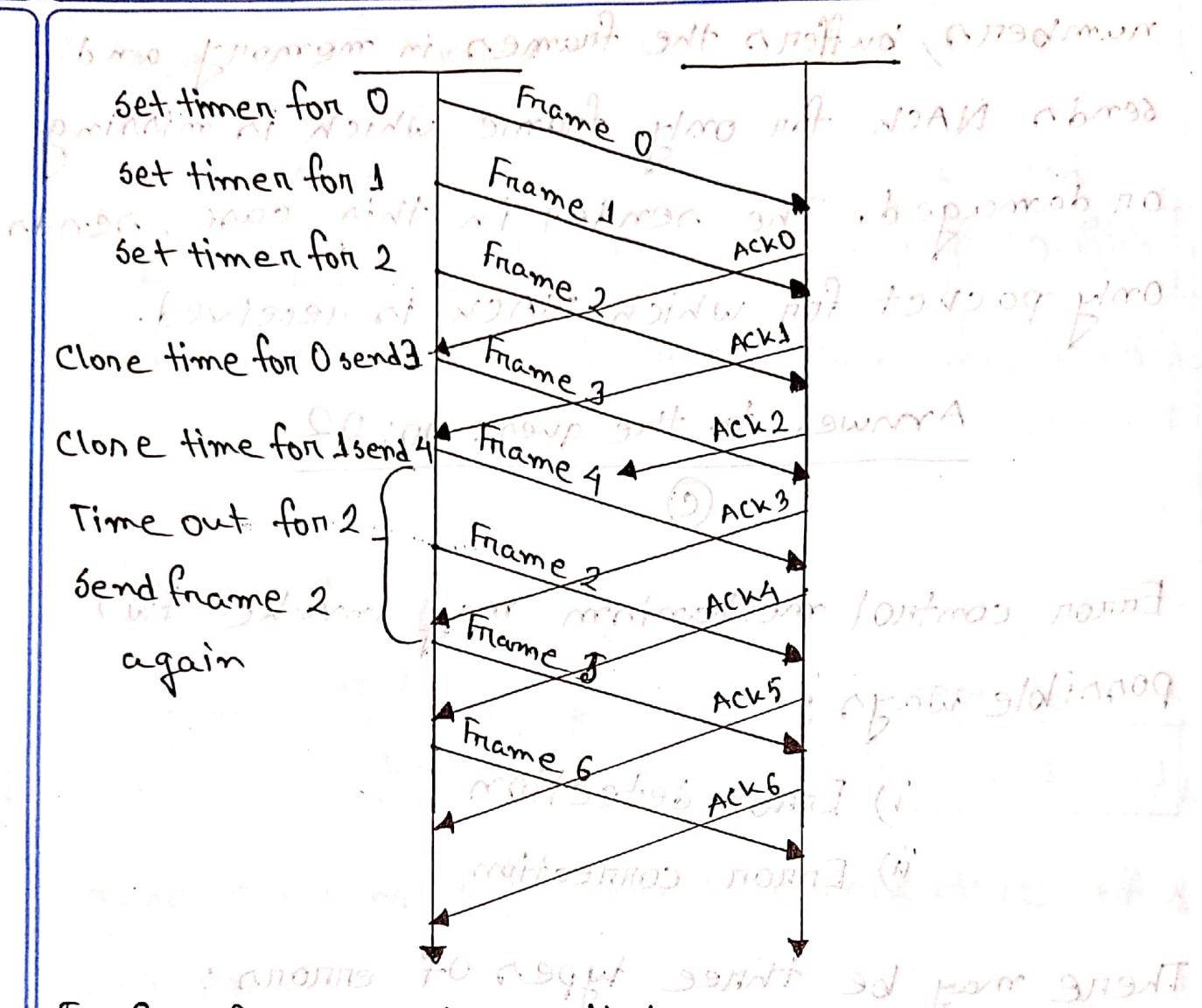
### \* Selective Repeat ARQ:

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In Go-Back-N ARQ, it is assumed that the receiver does not have any buffer space for its window size and has to process each frame as it comes. This enforces the sender to transmit all the frames which are not acknowledged. In Selective-Repeat ARQ, the receiver while keeping track of sequence

numbers, buffers the frames in memory and sends NACK for only frame which is missing or damaged. The sender in this case, sends only packet for which NACK is received.

Answer to the que: no; 02

(C)

Error control mechanism may invoke two possible ways:

- i) Error detection
- ii) Error correction

There may be three types of errors:

\* Single bit error

Received

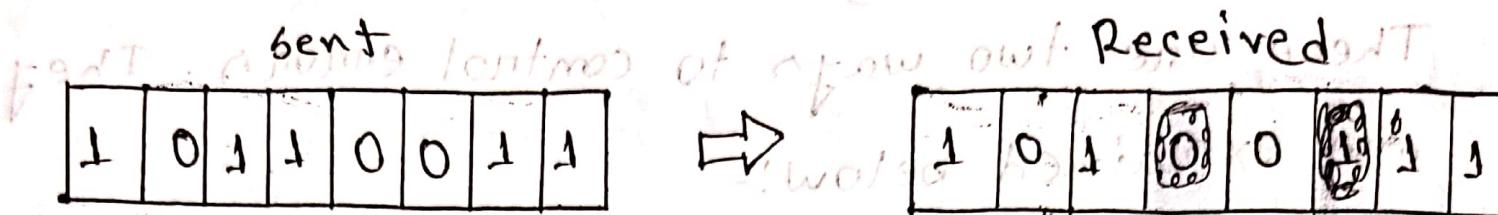
1	0	1	1	0	0	1	1
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1	0	1	1	0	1	1	1
---	---	---	---	---	---	---	---

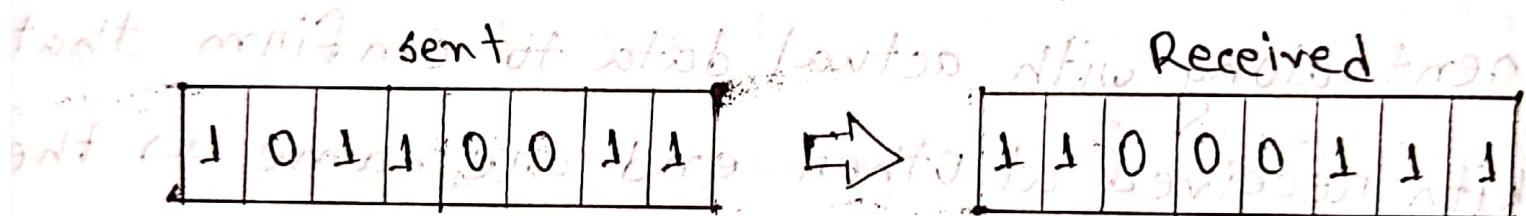
In a frame, there is only one bit, anywhere though, which is corrupt.

## \* Multiple bits error



Frame is received with more than one bits in corrupted state.

## \* Burst error



Frame contains more than 1 consecutive bits corrupted.

Consequently burst errors can affect multiple bytes at once.

Protocol for handling burst errors is to use Forward Error Correction (FEC).

FEC uses parity bits to detect and correct errors.

Parity bits are used to calculate the sum of all data bits.

If the sum is odd, then an error has occurred.

The parity bit is then set to make the sum even again.

This process continues until all errors are corrected.

Ans. to the que no: 03

(a)

There are two ways to control errors. They are explained below:-

### Error Detection

Errors in the received frames are detected by means of parity check and cyclic Redundancy check (CRC). In both cases, few extra bits are sent along with actual data to confirm that bits received at other end are same as they were sent. If the counter-check at receiver fails, the bits are considered corrupted.

Parity check : One extra bit is sent along with the original bits to make number of 1s either even in case of even parity, or odd in case of odd parity. The sender while creating a frame counts the number of 1s in it.

For example if even parity is used and number of 1s in even then one bit with value 0 is added this way number of 1s remain even. If the number of 1s is odd, to make it even a bit with value 1 is added.

The receiver simply counts the number of 1s in a frame. If the count of 1s is even and even parity is used, the frame is considered to be not-corrupted and is accepted. If the count of 1s is odd and odd parity is used, the frame is still not corrupted.

Cyclic Redundancy Check (CRC): CRC is a different approach to detect if the received frame contains valid data. This technique involves binary division of the data bits being sent. The division is generated using polynomials. The sender performs

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performs division operation on the bits being sent and calculates the remainder. Before sending go to the actual bits, the sender adds the remainder at the end of the actual bits. Actual data bits plus the remainder is called a codeword.

The sender transmits data bits as codewords.

At other end, the receiver performs division operation on codewords using the same CRC divisor. If the remainder contains all zeros the data bits are accepted, otherwise it is considered as there some data corruption occurred in transit.

### Error Correction:

In the digital world, error correction can be done in two ways:

- Backward Error Correction: When the receiver detects an error in the data received, it requests back the sender to retransmit the data unit.

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Forward Error Collection: When the receiver detects some error in the data received, it executes error-correcting code, which helps it to auto-recover and to correct some kind of errors. The first one, Backward Error Correction is simple and can be efficiently used where retransmitting is not expensive. For example, fiber optics. But in case of wireless transmission retransmitting may cost too much. In the latter case, Forward Error Correction is used.

Answer to the question 3

(b)

Devices which work on Network layer mainly focus on routing. Routing may include various tasks aimed to achieve a single goal.

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There can be various protocols for IP routing.

- Addressing devices and network.
- Populating routing tables on static routers.
- Internetworking between two different subnets.
- Delivering packets to destination with best in all possible ways based on traffic effort.
- Provides connection oriented and connection less mechanism.
- Queuing income and outgoing data and then forwarding them according to quality of service constraints set for those packets.

Ques. No: 03  
Answer to the que. no: 03

(c)

(d)

There are different kinds of network address which are IPX, AppleTalk etc.

- IP
- IPX
- AppleTalk

Ans: to the question no: 04

(a)

Network layer takes the responsibility for routing packet from source to destination within or outside a subnet. Two different subnet may have different addressing schemes or non-compatible addressing types. Same with protocols, two different subnet may be operating on different protocols which are not compatible with each other.

Network layer has the responsibility to route the packets from source to destination, mapping different addressing schemes and protocols with its standard functionalities, layer 3 can be provided various features as:-

- Quality of service management.

- Load balancing and link management

- Security

- Internation of different protocols and flow management with respect to subnets with different schema.

- Different logical network design over the physical network design.

- L3 VPN and tunnels can be used to provide end-to-end dedicated connectivity.

Answer to the question no: 04

Layer 3 network addressing is one of the major task of Network layer. Network addresses are always logical i.e. there are software based addressing which can be changed by appropriate configurations. A network address always points to host / node / server or it can represent a whole network. Network address is always

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configured on network interface card in generally.

• IP address of the host is mapped to the MAC address of the machine for Layer 2 communication.

IP addressing provides mechanism to differentiate between hosts and network. Because IP addresses

are assigned in hierarchical manner, a host

always resides under a specific network. The

host which needs to communicate outside its

subnet, needs to know destination network

address, where the packet / data is to be sent.

Host in different subnet need a mechanism to

locate each other. This task can be done by

DNS. DNS is a server which provides Layer 3

address of remote host mapped with its domain

name on FQDN. When a host acquires the Layer

3 address via broadcast.

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3 address (IP-address) of the remote host

it forwarded all its packet to its gateway.

A gateway is a router equipped with all the

information which leads to route packets to

the destination host. Router take help

of routing tables, which has the following

information:

• Method to reach the network Router upon receiving a forwarding request, forwards, packet to its next hop (adjacent router) towards the

destination. The next router on the path follows the same thing and eventually the

data packet reaches its destination.

Network address can be of one of the

following :-

- Multicast

- Broadcast

• Anycast

• Unicast

A router never forwards broadcast traffic by default.

Multicast traffic gets special treatment as it is most often used for video streams or audio with highest priority. Anycast is just similar to unicasting, except

that the packets are delivered to the nearest destination when multiple destinations are available.

Ans. to the que no: 04

The following are examples of protocols operating at the network layer:-

• CLNS - Connectionless - mode Network Service

• DDP - Datagram Delivery Protocol

• EGP - Extension Gateway Protocol

• ICMP - Internal Group Management Protocol

• IPv4 / IPv6 - Internet Protocol

Answer to the ques: no: 05

(a)

When a device has multiple paths to reach a destination, it always selects one path by preferring it over others. This selection process is termed as Routing. Routing is done by special network devices called routers or it can be done by means of software processes.

Answer to the ques. no: 05

(b)

Different types of routing is given below :-

- Unicast Routing - Most of the traffic on the internet and internets known as unic平 data on unicast traffic is sent with specified destination. Routing unicast data over the

internet is called unicasting. It is the simplest form of routing because the destination is already known. Hence, the router just has to look up the routing table and forward the packet to next hop.

#### • Broadcast Routing :-

By default, the Broadcast packets are not routed and forwarded by the routers on any network. Routers create broadcast domains.

But it can be configured to forward broadcast

in some special cases. A broadcast message is destined to all network devices.

#### • Multicast Routing :-

Multicast Routing is a newer version of routing of broadcast with significant difference and challenges. In broadcast routing, packet data are sent to all nodes even if they do not want it.

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But in multicast routing, the data is sent to

only nodes which want to receive the packet.

The router must know that there are nodes,

which wish to receive multicast packet then

only it should forwarded. Multicast routing works

spanning tree protocol to avoid looping. Multicast

routing also uses reverse forwarding technique

to detect and discard duplicates and loops.

Anycast Routing: bengfituad ad mod fi hbd

Anycast packet forwarding is a mechanism where multiple hosts can have same logical address. When a packet destined to this

logical address is received, it is sent to the host which is nearest in topology.

Anycast routing is done with help of DNS

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server. Whenever an ~~Any~~ packet is received it is enquired with DNS to where to send it.

DNS provides the IP address which is the nearest IP configured on its subnet (if

Ans: to the que: no:05

(C)

To avail of static routing one need

• Unicast Routing Protocols :-

• Broadcast and Multicast Routing protocols

There are two kinds of routing protocols

available to route unicast packet:

i) Distance Vector Routing Protocol -

Distance vector Routing Protocol is simple routing

Protocol which takes routing decision on the number of hops between source and destination.

A router with less number of hops is considered

as the best route. Every router advertises its best routes to other routers. Ultimately,

all routers build up their network topology based on the advertisements of their peer routers, for example Routing Information Protocol.

### ii) Link State Routing Protocol :-

Link state protocol is slightly complicated protocol than Distance vector. It takes into account the states of links of all the routers in a network. This technology helps routers build a common path of the entire network. All routers then calculate their best path for routing purposes.

For example, Open shortest path first (OSPF) and Intermediate system to Intermediate system (IS-IS)

### Multicast Routing Protocol :-

Unicast routing protocols use graphs

while Multicast Routing Protocols use trees, i.e.

Spanning tree to avoid loop. The optimal

tree is called shortest path spanning tree.

DVMRP - Distance Vector Multicast Routing Protocol  
Protocol for distribution of broadcast traffic

Protocol

It maintains shortest path to each node

MOSPE - Multicast Open Shortest Path First

Protocol for distribution of broadcast traffic

CBT - Core Based Tree

Protocol for distribution of broadcast traffic

PIM - Protocol Independent Multicast

Protocol Independent Multicast is commonly used

now. It has two flavors:-

PIM Dense Mode - This mode uses source based

tree. It is used in dense environment where there are many hosts attached to a single link.

PIM Sparse Mode - This mode uses shared trees.

It is used in sparse environment

such as WAN.

Scanned with CamScanner

Ques 4 Answer to the question

If they are two geographically separate networks, which want to communicate with each other, they may deploy a dedicated line between or they have to pass their data through intermediate networks. Tunnelling is a mechanism by which two or more same networks communicate with each other, by passing intermediate networking complexities. Tunneling is configured at both ends. When the data enters from one end of tunnel, it is tagged. This tagged data is then routed inside the intermediate or transit network to reach the other end of tunnel. When data exists the tunnel its tag is removed and delivered to the other part of the network.

Both ends seem as if they are directly connected and tagging makes data travel through transit network without any modification.

Ans. to the question no 06

(b) To provision QoS FA

Most Ethernet segment have their maximum frame size (transmit threshold) fixed to 1500 bytes.

A data packet can have more or less packet

length depending upon the application. Devices in the transmit path also have their hardware

and software capabilities which tell what amount of data device can handle and what amount size

of packet it can process. If the data packet size is less than or equal to the size of packet the

transit network can handle, it is processed

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naturally. If the packet is too large, it is broken into smaller pieces and then forwarded. This is packet fragmentation. Each fragmentation contain the same destination and source address and routed through transit path easily.

At the receiving end it is assembled again. If a packet with DF (don't fragment) bit set to 1 comes to a router which can not handle the packet because of its length, the packet is dropped.

When a packet is received by a router with its MF (More Fragment) bit set to 1, the router then knows that it is a fragmented packet and parts of the original packet is on the way.

If packet is fragmented too much small, the overhead is increase. If the packet is passing at the broadcast and shows to the receiver.

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fragmented too large, intermediate router may not be able to process it and it might get dropped

Answer to the que no: 06

(C)

Internet protocol version 4 (IPv4) is the fourth revision of the Internet protocol and a widely used protocol in data communication over different kinds of networks. IPv4 is a connectionless protocol used in packet-switched layer networks, such as Ethernet. It provides the logical connection between network devices by providing identification for each device. There are many ways to configure IPv4 with all kinds of devices including manual and automatic configurations, depending on the network type. IPv4 is based

On the best effort model. This model guarantees neither delivery nor avoidance of duplicate delivery, these aspects are handled by the upper layer transport. IPv4 is 32-bit addressing schema used in TCP/IP host addressing mechanism. IP addressing enables every host on the TCP/IP network to be uniquely identifiable. IPv4 provided hierarchical addressing scheme which enables it to divide the network into subnetworks, each with well-defined number of hosts. IP addresses are divided into many categories:

Class A - it uses first octet for network addresses and has three octets for host addressing.

Class B - first two octets are used for network addresses and last two octets are used for host addressing.

Class C - first three octets are used for network addresses and last one octet is used for host addressing.

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- Class B - it uses first two octets for network address and last two for host addressing.
- Class C - it uses first three octets for network addressing and last one for host addressing.
- Class D - it provides flat IP addressing scheme. It is contrast to hierarchical structure of above three.
- Class E - It is used as experimental. IPv4 also has well-defined address spaces to be used as private addresses, and public addresses.

Though IP is not reliable one; it provides Best-Effort-Delivery mechanism.

Effort-Delivery mechanism.

Loop for each station yet, both work in parallel.

No switching and no routing but still no FDDI.

Answer to the question: no; 07 i - 07 d

has had no sign **(a)** feel bad about being alone

The routing algorithms are as follow:

**Electrons** are not starting until family now 41-2000

## Flooding -

Flooding -  
food not there food becomes unmarketable

Flooding is simplest method packet forward

and after received, the routers send it

When a packet is received, the routers send it to all the interfaces except the one on which it came (addition of interface fib).

it was received. This creates too much burden

on the network and lost lots of duplicate

Figure 7: Bitstreaming as source of the problem  
packets wandering in the network.

This passage on Webb's Bright-line test  
will often be used to avoid

time to live (TTL) can be used to avoid infinite flooding by setting its value. There exists another approach for flooding, which is called selective flooding.

Flooding to reduce the overhead on the network.

In this method, the router does not flood

out on all the interfaces, but selective ones

Sub:

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Shortest path = ~~shortest path~~ shortest path

Routing decisions in networks are mostly taken on the basis of cost between source and destination.

Hop count plays major role here. Shortest path is a technique which uses various algorithms to decide a path with minimum number of hops.

Common shortest path algorithms are:

- Dijkstra's algorithm

- Bellman Ford algorithm

- Floyd Warshall algorithm

Ans: to the ques no: 07

(b)

ICMP in network diagnostic and error reporting protocol, ICMP belongs to IP protocol suite and uses IP as carrier protocol. After constructing ICMP packet, it is encapsulated in IP packet.

Because IP itself is a best-effort non-reliable protocol, so in ICMP we maintain virtual link for reporting errors or about network information back to the originating host. If some error in the network occurs, then it is reported by means of ICMP. ICMP contains dozens of diagnostic and error reporting messages.

ICMP-echo and ICMP-echo-reply are the most commonly used ICMP messages to check the reachability of end-to-end hosts. When a host receives an ICMP-echo request, it is bound to send back an ICMP-echo-reply. If there is any problem in the transmit network, the ICMP will report the problem.

Q. What is the difference between ICMP and IP?

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Some points from Ans. to the question no 7

Ques. No. 7. (c) Explain IP version 6.

Internet protocol version 6 (IPv6) is the most recent version of the Internet Protocol (IP), the communications protocol that provides an identification and location system for computer on networks and location system for computers routers traffic across the internet. Devices on the Internet are assigned a unique IP address for identification and location definition with the rapid growth of the Internet after commercialization in the 1990s. it becomes evident that far more addresses would be needed to connect devices than the IPv4 address space had available.

Exchanges of IPv4 addresses gave birth to a next generation Internet Protocol version 6, IPv6 addresses its nodes with 128 bit wide address providing,

plenty of address space for future to be used on entire planet or beyond. IPv6 has introduced Anycast addressing but has removed the concept of broadcasting. IPv6 enables devices to self-acquire an IPv6 address and communicate within that subnet. This auto configuration removes the dependency of Dynamic Host Configuration Protocol (DHCP) servers. This way, even if the DHCP server on the subnet is down, the hosts can communicate with each other. IPv6 provides new feature of IPv6 mobility. Mobile IPv6 equipped machines can roam around without the need of changing their IP addresses. IPv6 is still in transition phase and is providing various services to set up a better life.

expected to replace IPv4 completely in coming years. At present, there are few networks which are running on IPv6. There are some transition mechanisms available for IPv6 enabled networks to speak and roam around different networks easily on IPv4. These are:

- i) Dual stack implementation
- ii) Tunneling
- iii) NAT-PT

Ans to the question no: 08

(a)

TCP/IP functionality is divided into four layers, each of which include specific protocols:

- i) The application Layer -

Provides applications with

standardized data exchange. Its protocols include the Hypertext Transfer Protocol (HTTP), File Transfer Protocol (FTP), Post office protocol (POP 3), Simple Mail Transfer Protocol (SMTP) and Simple Network Management Protocol (SNMP). At the application layer, the payload in the actual application data.

### ii) The transport layer -

is responsible for maintaining end-to-end communication across the network. TCP handles communications between hosts and provides flow control, multiplexing and reliability. The transport protocol include TCP and User Datagram protocol (UDP) which is sometimes used instead of TCP for special purposes.

iii) The Network layer - also called the internet layer, deals with packets and connects independent networks to transport the packet across network boundaries. The network layer protocols are the IP and the Internet control Message Protocol (ICMP), which is used for error reporting.

iv) The physical layer - also known as the network interface layer or data layer, consists of protocols that operate only on a link - the network component that interconnects nodes on hosts in the network. The protocol in this layer include Ethernet for local area networks (LANs) and the Address Resolution Protocol (ARP).

Answer to the question 8

(b) TCP/IP uses the client-server model of communication in which a user on machine (client) is provided a service (like sending a webpage) by another computer (a server) in the network.

Collectively, the TCP/IP suite of protocols is classified as stateless, which means each client request is considered new because it is unrelated to previous requests.

Being stateless frees up network paths so they can be used continuously. The transport layer itself, however, is stateful. It transmits a single message and its connection remains in place until all the packets in a message

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have been received and reassembled at the destination. The TCP/IP model differs slightly from the seven-layer Open System Interconnection (OSI) networking model designed after it. The OSI reference model defines how applications can communicate over a network.

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(C)

Importance of TCP/IP

TCP/IP is nonproprietary and as a result is not controlled by any single company. Therefore, the Internet Protocol suite can be modified easily. It is compatible with all operating systems, so it can communicate with any other system.

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The Internet Protocol suite is also compatible with all types of computer hardware and networks.

TCP/IP is highly scalable and is a routable protocol, can determine the most efficient path through the network. It is widely used in current Internet architecture.

③

91/92T Go to question 1

It functions as the first program of 91/92T

notional. Programs like PC, Etc. but it is not  
of fibres. It has static load balancing and  
prioritises traffic slotted queue with fibres  
also known as traffic prioritised queuing