

Host address → 32 bits  
 → IPv4 address }  
 → IPv6 128 bit }  
 → MAC Address }  
 ↓ 48 bits.

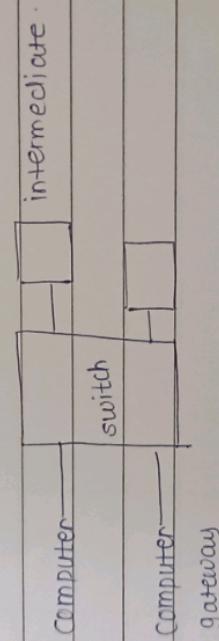
Ethernet Address  
 ↓

NIC → LAN Case 1.

TP-add

8 bit    8    8    8

decimal conversion.



Gateway to pass the message through switch  
 Subnet mask is used to create the network.

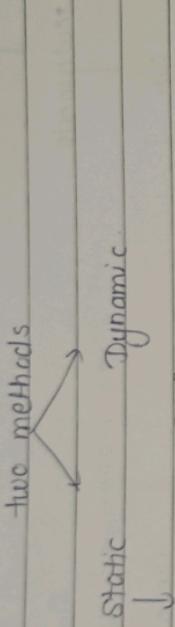
255.255.255.0 → 11111111.11111111.11111111.00000000  
 Host bit → 2<sup>8</sup> - 2 = 254 → maximum no. of computers can be connected.

256 - 2 ip's are reserved.

1st is network id  
 2nd is broadcast address.

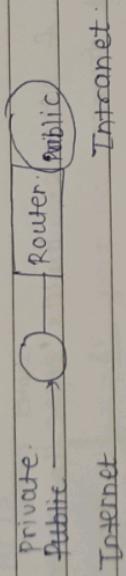
Switch - To connect number of Host to send packet within same Net.

- How to assign the address?



- 1) Assign IP address
  - 2) Subnet mask
  - 3) Gateway
  - 4) DNS switch
- MAC is by default

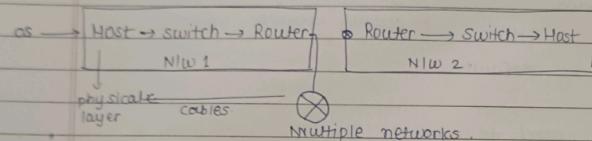
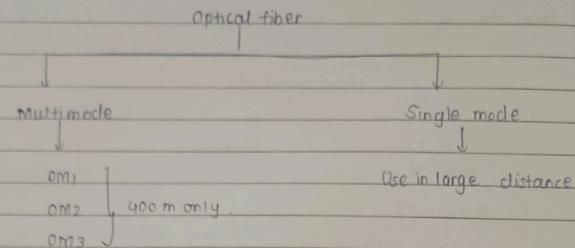
\* Router is used to connect the Net from LAN to LAN.



Dynamic DHCP server → Assign IP address immediately.  
lease address  
IP Address: --- Dynamic.

Static → comp class c  
192.168.0.1

Static IP address is assign to server  
Dynamic IP address is assign to LAN



Switch → Datalink & Network. Builds <sup>tables</sup> (MAC table or ARP table) to forward frames.

Books → Computer N/w up to down  
(Kurose)

Data Communication (frozen)

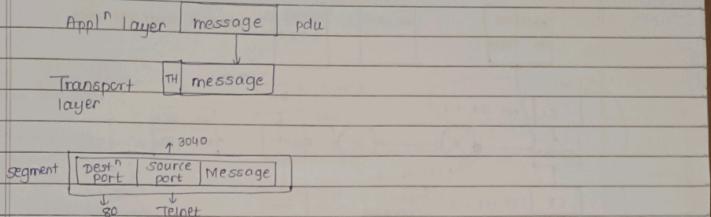
Q. Can client and server is Host ?  
Source                  host

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Date 19/07

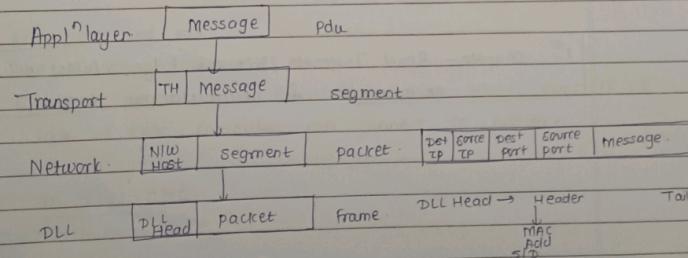
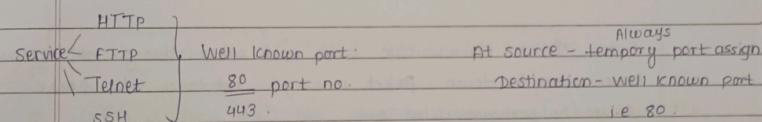
### TCP / IP Model

↓

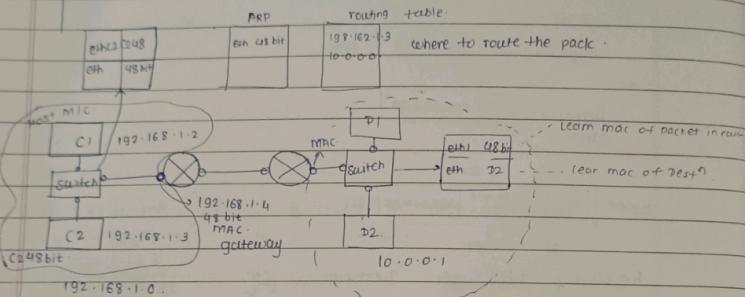
Google.com



Resolving DNS get Destination IP.



- To get destination or mac computer sent ARP request to gateway/Router
- Router also has its own ARP table and routing table.
- Mode of the switch is to learn the mac and bit the table & forward

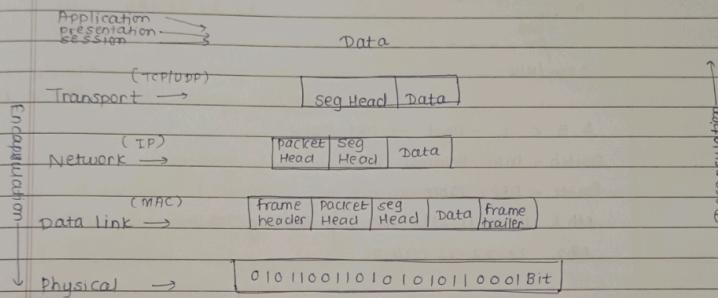


- Every device communicate with each other through mac address.
- Router doesn't touch packet message but touch N/W & MAC.
- Router only known N/W IP.
- Router is used to Route and take decision of routing interface
- ARP Request happen at DIL.

1<sup>st</sup> chapter - Read Internet, Network Edge, Access network, core encapsulation & decapsulation, Host.

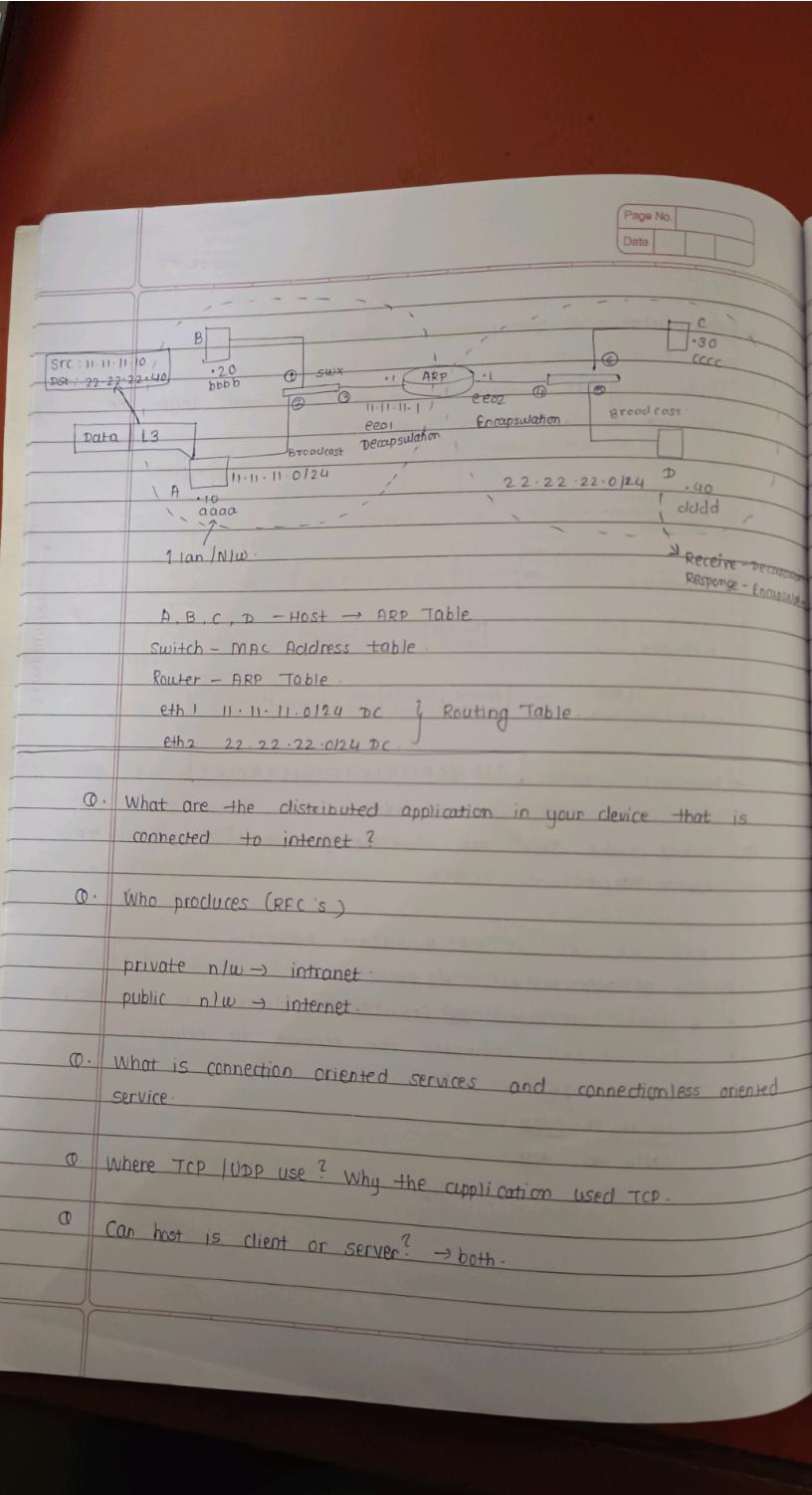
e. Encapsulation & Decapsulation in OSI Model.

When msg is forward toward transport layer the message is divided into segment.



Q. What is the Dest<sup>n</sup> MAC from the given Dest<sup>n</sup> IP, Source IP, Source MAC.

- 255.255.255.255 → Broadcast Address.
- Size of MAC Address = 48 bit.
- 6 octets are used for Broadcast Address → ff.ffff:ff:ff
- 124 → 254 Computer can connect to network.
- 125 → 126.
- 11.11.11.0124  
New Add Host



Q What are the codes used in application layer protocol?

Q How many imgs are sent by web browser?

• FTP - client server architecture.

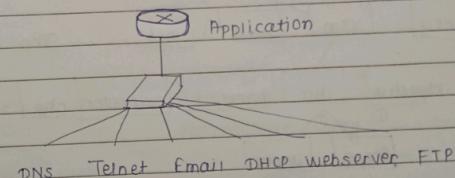
Q Delay. calculate?

• Types of delay (marks)

24/07

1. Nodal processing
2. Queueing (waiting for transmission)
3. Transmission
4. Propagation

• Application layer - OSI and TCP/IP models. Chapter 2



DNS - Resolve internet name to IP address

Telnet - Access to server and network devices

SMTP - Transfer of mail messages and

DHCP - Assigns IP address & other parameter to host

HTTP - Transfer file to create web pages

FTP - Transfer file between system

## Communication

Peer-to-Peer

client & server

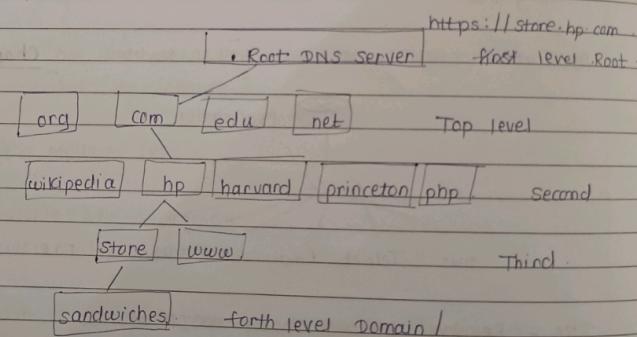
Q. what is

- o Commonly used protocols in Application layer
- 1. Domain Name system -
- 2. Hypertext Transfer -
- 3. Simple MTP
- 4.

o DNS = Domain Name System

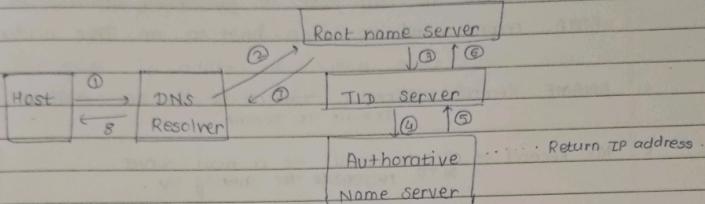
To translate a host name into an IP address.

13 root server in DNS.

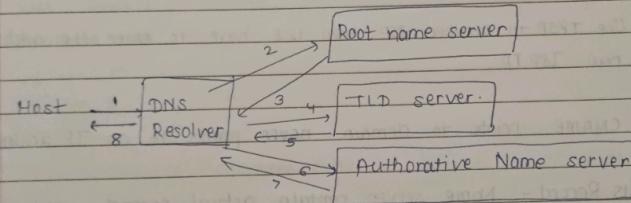


Q. URL given tell the levels.

Q Explain Recursive Query.



Q Iterative Query



Two types of queries : Recursive and Iterative.

Nonrecursive Query ①

What is the port used by DNS?

What server hierarchical?

What are the queries used by DNS to get webpage?

How many Root Server are there?

- A and AAAA Records -  
A record used to map host to an IPv4 address  
AAAA record used to map host to an IPv6 address
- CNAME Record - directs traffic to the new address  
Domain to domain.
- MX Record - directs email to a mail server.  
MTA responsible for querying MX.
- TXT Record - Resource type

Q. What is the uses of these records? What are other records.

One ISP to another ISP then we have to enter the address of new ISP IP

CNAME points to domain never points to an IP address.

- NS Record - Name server contains actual records.
- Class type field - Internet
- Record type field - SOA
- Zone file is always created on Master server.
- Name server that serves a particular zone file uses it to map IP address.
- The refresh time field.
- The negative cache.
- SOA is not often used.

- (e) For given IP address & subnet mask, find network and broadcast address for given device.
- (f) For given IP address & subnet mask, find network & broadcast address for given device.

#### DNS - Chapter 2.

- 1) Explain concept of DNS
- 2) Types of DNS server → Explain (Resolver, Root, TLD, Authoritative)
- 3) DNS Queries (3)
- 4) DNS Records Resource  
(AAAA, A, NS, CNAME, TXT, SOA, PTR)

#### Chapter 1.

Introduction to Computer Network

TSP

Internet

IP address public & private

What are the devices used in CN → switch, router, hub,

Host, end system, distributed applications running on host  
client server, p2p

medium → wired, wireless

TCP/IP layer and OSI layer difference.

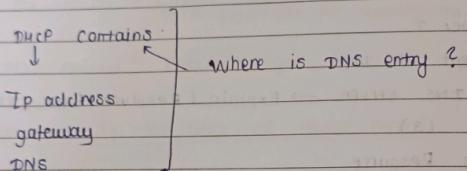
Layer architecture

ARP → How mac is obtained.

Which device is used at which layer.

o HTTP

- Q. What is RFC 2 → Request for comment.  
Q. private IP has no cost → free.  
public IP address has to pay cost because it is provided by ISP.

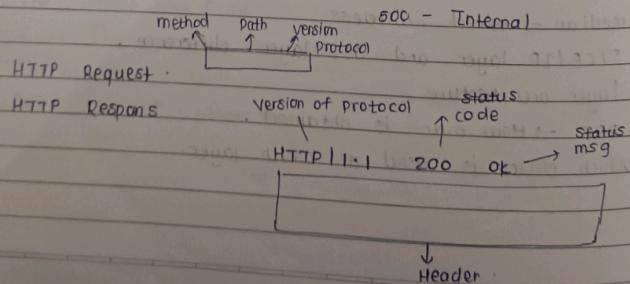


- Q. Which underline protocol used by HTTP? → TCP.

HTTP Method - GET, POST, PUT, DELETE. used for request.

HTTP Status codes - used for response by server.

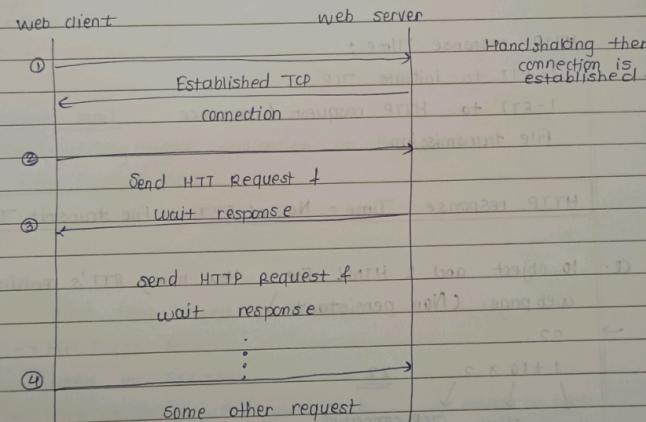
1xx : informational	200 - OK
2xx : success	201 - OK created
3xx : redirect	301 - Not modified (cached version)
4xx : client error	400 - Bad request
5xx : server error	401 - unauthorised. 404 - Not found 500 - Internal



Request - [Request Type] [URL] [HTTP version]

Response - [HTTP version] [status code] [status Phrase]

- ① status code → can be asked for match the pairs.



Socket → combination of [IP + Port]

client IP + port → client socket

server IP + port → HTTP / server

Internet connection  
Reliable data transfer  
Reordering of data

Two methods → persistent      with pipelining.  
HTTP connection      Non persistent

• RTT [round-trip time] = A time for small packet to travel from client to server and server to client.

HTTP response Time :

1-RTT to initiate TCP  
1-RTT to HTTP request & response  
File transmission

$$\boxed{\text{HTTP response Time} = \text{No. of RTT} + \text{file transmit Time}}$$

Q. 10 object and 1 HTML file. How many RTT's required to get web page. (Non persistent) .

→ 22.

$$1 + 10 \times 2 = 22$$

HTML objects      TCP connection  
file                  and request  
                        HTML

$$1 \text{ TCP} + 22 \text{ Images } 23 \times 2 = 46 \text{ RTT NP}$$

23 \* 1

24 RTT P

Part 31/107

- DNS - 53
- HTTP - 80
- SMTP - 25
- POP3 - 110
- Telnet - 23
- DHCP
- Telnet - 67
- FTP - 20, 21

Commonly used protocols.

## • Electronic mail

Most widely used application on the internet.

- for sending mail
  - for receiving mail

## 1) SMTP

- pure text based protocol
  - 7 bit ASCII format
  - Based on RFC 2821
  - Port 25 - non-encrypted port  
Port 465 - send message securly } works on port
  - Push protocol

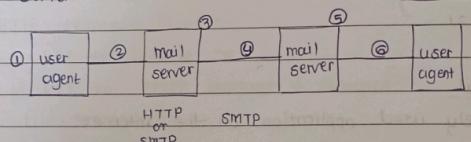
## Working of SMTP

220 - Service ready  
 221 - closing trans  
 250 - Requested action ok  
 354 - end with <endl> <crlf>

- Internet-mail system

Three major components - user agent, mail server & SMTP protocol

Transfer the email message from mail server <sup>user to</sup> and other mail server to another mail server. → main use of SMTP.



• Hello		101	
EHLO		211	
MAIL	code / command	214	
FROM		220	
RCPT TO	from client	221	Code from server
SIZE DATA		235	
QUIT		250	
VRFY		354	
EXPN		500	
HELP ..		510	

- Email Server Processes : MTA & MDA.

- Mail Transfer Agent - used to forward email

Rewrites msg from MUA or another MTA

- Page No. \_\_\_\_\_  
Date \_\_\_\_\_
- Mail Delivery Agent -
    - 1) Accept mail from MTA.
    - 2) place it into appropriate mail box
  - MUA - mail user agent .
- 2) POP3 - Post office Protocol (POP3) version 3
- Pull protocol
  - Uses TCP 110 .
  - Download-&-delete mood → retriver server - store locally - Delete , serve,
  - Download-&- keep mood .
  - works on two port  $\xrightarrow{\text{110 - Normal}}$   
 $\xrightarrow{\text{995 - secure}}$
- 3) IMAP - Internet Message Access Protocol .
- Email is not download , but can retained
  - Any received email is associated with users INBOX
  - Users can create message
  - two part  $\xrightarrow{\text{143 - Normal}}$   
 $\xrightarrow{\text{993 - secure}}$
  - Web based email - HTTP is used  $\xrightarrow{\text{Push}}$  (client to server)  
Pull protocol .
- Proprietary protocol -
- 1) IBM Lotus Notes
  - 2) Novell Groupwise
  - 2) Microsoft Exchange
- Web based -
- 1) Hotmail
  - 2) Gmail
- Email server process  
MUA , MTA .  
other internet email format .
- (Q. What is SMTP , POP3 , IMAP → draw diagram , gives scenario and show where it is used .)

### Ports

- Port - A 16 bit number that identifies the application process that receives an incoming message.  $2^{16} = 65536$  port
- Reserved port or well known ports (0 to 1023)  
Client can't use this ports. These are standard ports telnet, http, ftp.
- Ephemeral port (1024 - 49,151) (49,152 to 65,535)

### TANIA Ranges

- Well known ports - 0 to 1023
- Registered port - 1024 - 49,151
- Dynamic (or private) - 49,152 to 65,535

command - netsh int ipv4 show dynamicport tcp.]

What is the range of <sup>register</sup><sub>dynamic</sub> port?

- Socket is combination of an 32 bit IP address and 16 bit port number

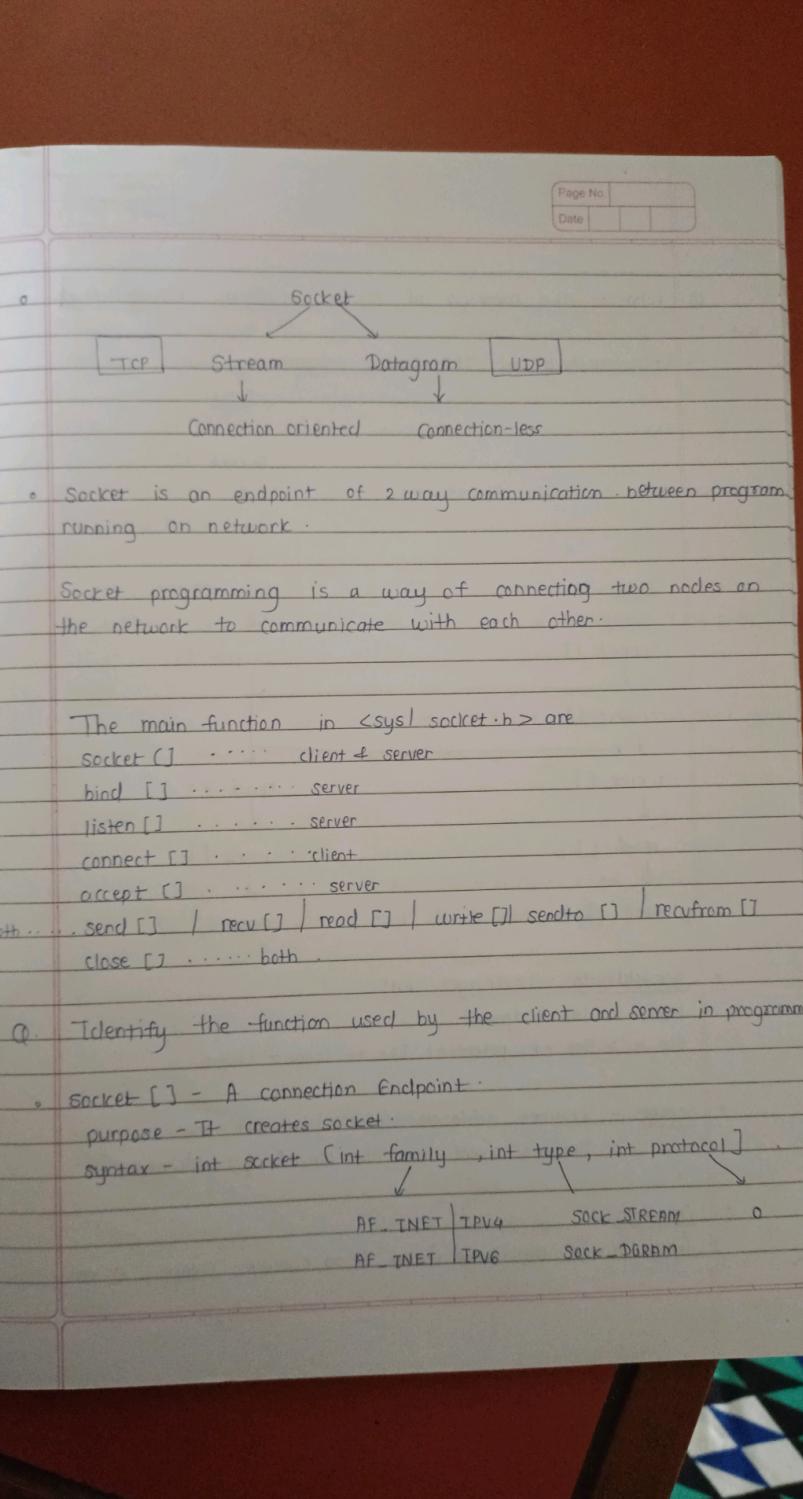
### Use of socket

- 1) as an endpoint for communication.
- 2) A socket is identified by an IP address concatenated with a port.

(Q) What is the use of socket?

(Q) Differentiate between socket and port?

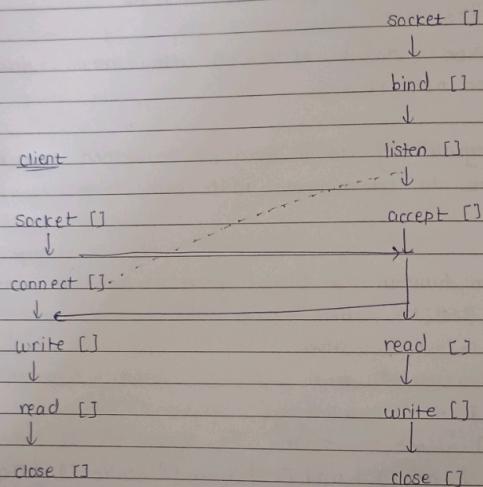
- Socket is interface between Application layer & transport layer



Q. What is the meaning of socket primitive TCP socket.

- Understanding of TCP socket programming.

Server



- getaddrinfo - Leverages DNS.

IP - a dns hostname

OLP - a list of potential IPs to connect / listen.

- Socket - creates a file descriptor, just like open.

Doesn't even use the OLP of getaddrinfo

- Connect -

Given a file descriptor & an IP to connect, create a connection.

- send & recv

Given a connected file descriptor, submit bytes to the OS for delivery bytes.

- close

Given ~~1~~ → ~~1~~ → ~~1~~, tell kernel that it can terminate this connection.

- bind

→ , tell the kernel to associate with given TCP port.

- listen

→ that has been binded to TCP port, tell the OS that you wish to start receiving connections.

- accept

→ that has already been activated via listen, create a new FD that can be used to communicate with individual client. By default this call block until client shows up.

#

import socket

#

server\_socket = socket.socket(socket

198 -  $\begin{array}{|c|c|c|} \hline 2 & 198 & 0 \\ \hline 2 & 99 & 1 \\ \hline 2 & 49 & 0 \\ \hline 2 & 24 & 0 \\ \hline \end{array}$  110 00 110  
class C

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- Transport Layer

$\begin{array}{l} \text{NIW} \quad \text{3 bits NIW} \\ \text{bits} \end{array}$       Host host host 1-126 /8  
 CLASS A  $2^7$  0NNNNNN Host host host 1-126 /8  
 CLASS B  $2^6$  10NNNNNN NIW host host 128-191 /16  
 CLASS C  $2^5$  110NNNNN NIW NIW host 192-223 /30  
 CLASS D 1110 MMMMM multicast group - - 224-239

CLASS E CLASS F Research.

Q-Binary 1st octet is given find out the class.

- Subnet Size - /24 to /30.

class C subnet mask /24

192.168.0.0 → NIW id.  
 :  
 192.168.0.10 → 128 subnet 1  
 and 192.168.0.10  
 192.168.0.128 → 255.255.255.0 → <sup>sub</sup> netmask  
 192.168.0.0 → NIW id.  
 192.168.0.140 → 128 subnet 2  
 192.168.0.255 → broadcast address  
 192.168.0.10 or  
 255.255.255.0  
 192.168.0.255 → Broadcast address

/25 Bit by bit and operation.

255.255.255.00000000  
 and 192.168.0.20  
 255.255.255.128  
 192.168.0.0

or 192.168.0.255 → Broadcast Address.

192.168.0.140 ← 2<sup>nd</sup> subnet

255.168.0.128

192.168.0.128 → start address of subnet 2

$$\begin{array}{ccccccc}
 \text{Subnet} & 128 & & 8 & 8 & 8 & 4 \text{ bit } 2^4 = 16 \\
 & & & 255 & 255 & 255 & 1111 \\
 255 \cdot 255 \cdot 255 \cdot 240 & & & \downarrow & \downarrow & \downarrow & 0000 \\
 \text{No. of Subnet} - 16 & & & 128 & 04 & 12 & 16 \\
 \text{Host per Subnet} - 2^4 - 2 = 14.
 \end{array}$$

192.168.0.250

- 255 · 255 · 255 · 240

192.168.0.240 → N/w id.

0 · 0 · 0 · 239

192 · 168 · 0 · 250

192.168.0.255 → BN id

If subnet is 130 2 computer can be connected

$$\text{class A} \quad 10 \cdot 0 \cdot 0 \cdot 10 \quad 125 \rightarrow 2^{17} / 2^{28} \text{ per subnet address} \quad 255 \cdot 0 \cdot 0 \cdot 0 \\ 8+8+1=1$$

$$\text{Class B} \quad 172 \cdot 16 \cdot 0 \cdot 10 \quad 125 \rightarrow 2^9 / 128$$

$$\text{class C} \quad 192 \cdot 168 \cdot 0 \cdot 10 \xrightarrow{\text{add}} 125 \quad \rightarrow 2^8 | 123.$$

## Transport Layer

TP Layer Process to process communication

Nw Layer Host to Host communi

DLL Node to Node communi

- Transport layer is responsible
  - i. service point and addressing

- Process level addressing

- multiplexing & demultiplexing

(① where multiplexing & demultiplexing occurs)

- Segmentation, Packaging & Reassembly → host

- Connection Establishment & Management and Termination

- Acknowledgment

- flow control

- Network layer, Transport layer

- Process to process achieve through client-server paradigm.

process on the local host - client,

need services from a process on remote host - server.

for communication we must define.

1) local host - TP

2) local process - port

3) Remote host - TP

4) Remote process - port

0-65536. (16 bit).

✓ MSS - Maximum segment size - TPL  
✓ MTU - maximum Transfer unit - NL

Page No.	
Date	

Provide -

- ✓ - aggregate data from different applications into a single Stream - Multiplexing.
- distributing that data to different application - Demultiplexing.

Segmentation - host , fragmentation - every node.

• Popular Application of UDP .

- ✓ ① Multimedia streaming .
- ② What are the applications supported by UDP .
  - DNS TFTP RTP
  - SNMP NFS
  - BOOTP NTP

③ Difference between TCP & UDP .

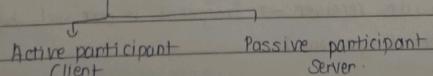
\* Establishing connection between sender & receiver → connection oriented .

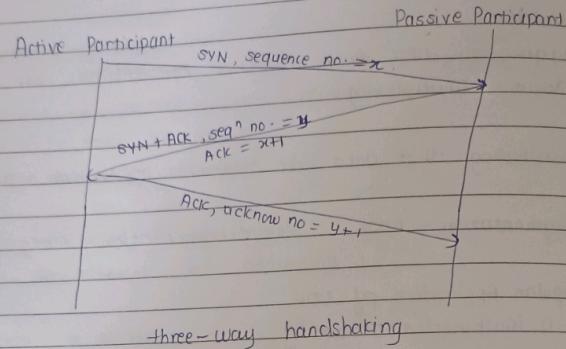
✓ UDP data Reassembly .

- TCP Protocol  Provide a reliable , in-order , byte stream abstract flow and congestion control .

✓ • TCP Segment Reassembly

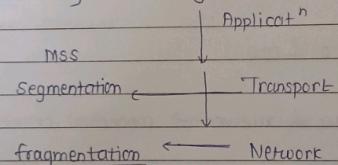
• Connection creation :





Q If seq<sup>n</sup> n what is the acknowledgement no?  $\rightarrow y+1$

- Three-way handshaking



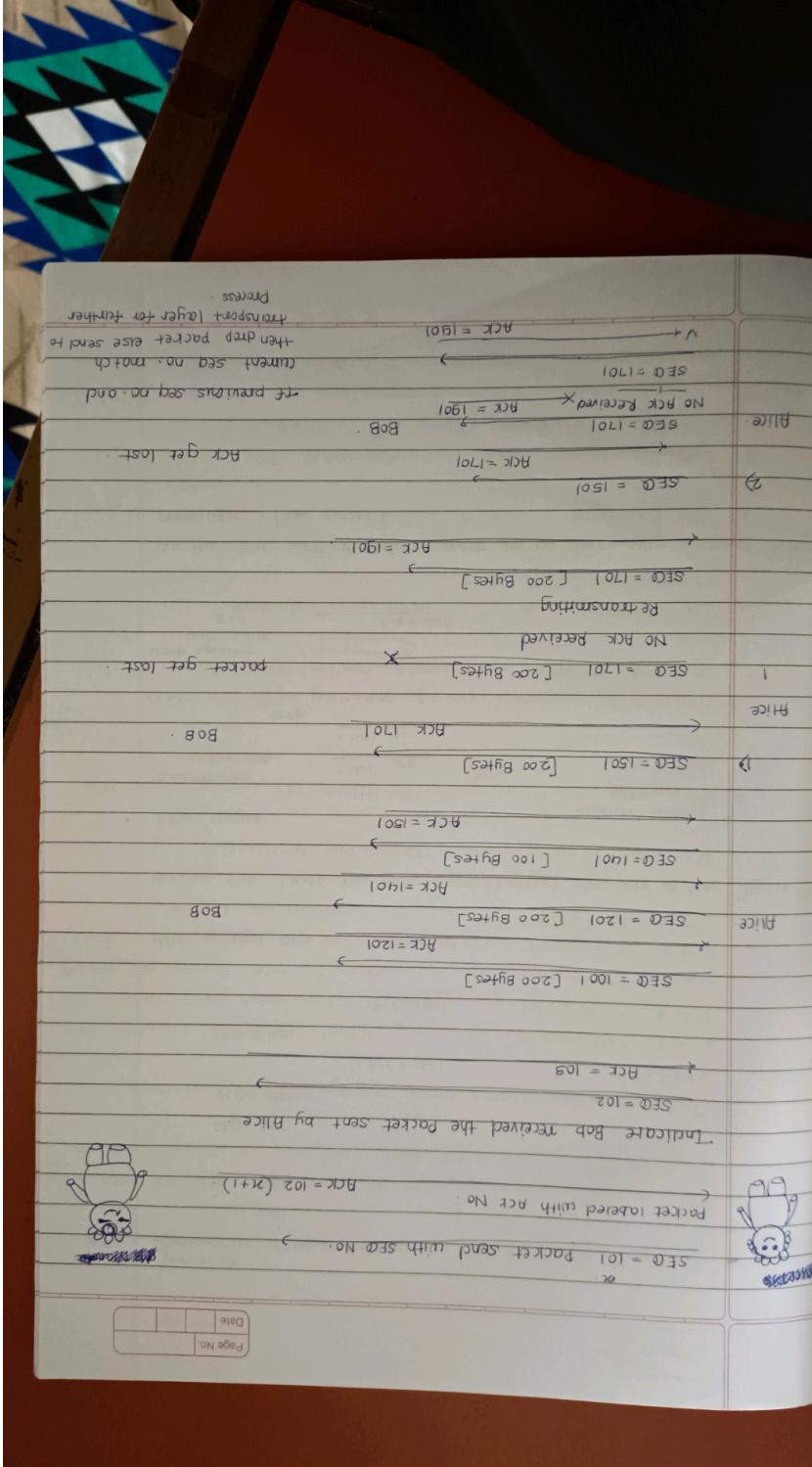
Q why segmentation is done at transport layer & why fragmentation is done at N/w layer?

Maximum Transport unit (MTU) 1500 bytes.

→ In between node at router fragmentation is done.  
Host perform segmentation.

If DF=1 at router  $\rightarrow$  Drop packet from router to router.

If DF=1 at receiver  $\rightarrow$  fragment is received by receiver do not frag  
(DF=1 Do not fragment)



Window size limits how much unacknowledged data can be sent. By 10 PPT  
 Typecasts on receiver  
 Q. How window size is determined.  
 Window size →  
 $RCV = 1801$   
 $SEG = 1601$   
 $SFA = 1001$   
 $SEG = 1501$   
 $RCV = 1401$   
 $SEG = 1201$   
 Cumulative acknowledgement

Window size sent in each segment can be dynamically updated through  
 connection (flow control).  
 Sender effective window size →  
 $RCV = 1501$   
 $SEG = 1601$   
 $RCV = 1401$   
 $SEG = 1201$   
 $RCV = 1201$   
 $SEG = 1001$   
 Flow control

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Digital Sequence Number

+ TCP connection starts with 3-way handshaking, includes 4 events:

TCF is bidirectional - both peers can send data. Both have an B2C & T2C track by test seed.

A  $\hookrightarrow$  B SYNchornize with my initial sequence Number of Y  
A  $\Rightarrow$  B received your SYN, I acknowledge that I am ready  
for [Y+1]

Focus - way handwriting with fin flags. Graceful connection closing

- Long range connection raising one-way with RST flags $\rightarrow$  B something went wrong sending RST flag.
- The window principle: why it is adjusted. size of window is 1.

$$\text{segmentWindow} = \text{AdvertisedWindow} - (\text{LastByteSent} - \text{LastByteReceived})$$

$$\text{PDU Window} = \text{Max RCV Buffer} - (\text{LastByteSent} - \text{LastByteReceived})$$

$$\text{How much time window is available?}$$

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Lenovo

• Initial Advertised Window is 300.

• Receiver capability - Advertised window.

Q What is sender window size and receiver window size?  
and how much data is remaining to send from the sender?

• Find out effective window?

Sender { Last byte sent = 2600  
Last byte written = 3000

Receiver { Last Read byte = 1700  
Last Received byte = 1700  
Advertised Window = 2400  
Next expected = 2401

EffectiveWindow (SenderWindow) = 200.

= Advertised window - outstanding bytes.

• Frame control

RDT - Reliable Data Transfer.

RDT = 2.0  
RDT = 2.1  
RDT = 3.0

RDT = 1.0 Error free channel; No error checking necessary.  
RDT = 2.0 Channel with error bit  $\rightarrow$  corrupted packet.

RDT = 2.1 ACK, NAK + GEC

- Handling Bit Errors  
 • Error detection  
 • Correct ACK / NAK packet  
 • Option 1  
 • Handler interrupts corrupt ACK / NAK = ACK.  
 • Receiver misses data packet.  
 • Option 2 . PDT 2.0 without sequence number.  
 • handle interrupt corrupt ACK / NAK .  
 • generate interrupt corrupt ACK / NAK .
- Version 2.1 . with seq<sup>n</sup> no . Seq + ACK + NAK .  
 Version 2.2 with Seq + ACK without NAK .  $\rightarrow$  Corrupted packet
- Bug fix - with Seq<sup>n</sup> no .  
 Pack to made = a  
 Pack to made /  $\left\{ \begin{array}{l} \text{Seq } \\ \text{number} \end{array} \right.$   
 Pack to made = 0  
 Bug fix - with Seq<sup>n</sup> no .  
 Version 2.0 with ACK + NAK .  $\rightarrow$  corrupted packet
- Version 2.1 . with Seq + ACK + NAK .  
 Version 2.2 with Seq + ACK without NAK .  $\rightarrow$  Corrupted packet
- Lost packets  
 Package lost , Handling lost packets  $\rightarrow$  package lost .  
 Version 2.0 with Seq + ACK + Timeout .
- Performance Problem with stop-and-wait (wait too long) .
  - Transmission Delay  $\approx 64 \text{ KB} \times 8 \text{ bits/B} = 5 \text{ ms}$
  - $T_t = \frac{\text{Transmission Delay}}{10 \text{ Mbps}} = \frac{5 \text{ ms}}{10 \text{ Mbps}} = 0.5 \text{ ms}$
  - $T_t + RTT = T_t + \frac{\text{Round Trip Time}}{2}$

- 19/10/08

  - Pipelined protocol - error handling mechanism
  - Why seq no. is small in selective repeat protocol.
  - TCP connection setup.
  - Three way handshaking.
  - TCP sequence number.
  - What is ACK & SEQ?
  - Selective Acknowledgment algorithm (SACK)
  - TCP Header + UDP Header
  - Find out sequence number.
  - Source port, destination port, seq number, Ack number, Header length Reserved bits, URG bit, ACK bit, PSH bit, RST bit.
  - Header length (4 bytes).
  - what is the size of TCP header.
  - 0101 → 5 (5d)
  - 1010 → A (10d)
  - 1111 → F (15d)
  - concept of scaling factor -
  - if Hex value will be given. using Hex 1 bit, 2 bit
  - find IP, Port, flag set, length.
  - UDP Header contains four fields :-
  - checksum calculation  $\rightarrow$  find sum and negate the sum.

- Underlying channel perfectly reliable

  1. No bit errors
  2. No loss of packets
  3. Separate EMS for sender, receiver.
  4. Sender sends data into underlying channel
  5. Receiver reads data from underlying channel
  6. Receiver perfectly reads data from channel

Sender

Receiver

RDT 2-a: EMS Specification.

Q: What is difference between RDT 2.1 & RDT 2.0? (Seq. no. is discarded)

A: RDT 3.0: Selector (Timer is introduced)

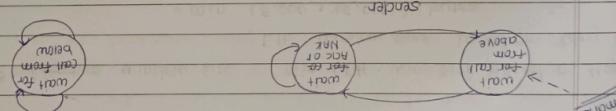
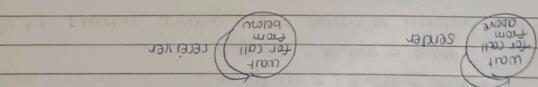
Implementation Rule 2.2 and 3.0

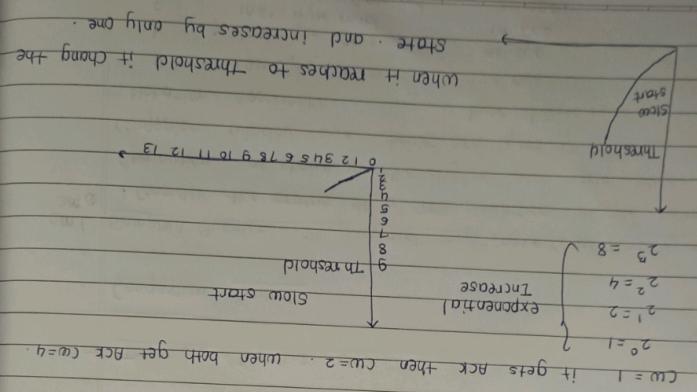
Sample Question - Following central agency for congestion control

  - Consider the status of a TCP connection at the source and destination as shown in the figure and take below. Let the congestion window size be 15,000 bytes.
  - Consider the status of a TCP connection at the source and destination as shown in the figure and take below. Let the congestion window size be 15,000 bytes.
  - After transmission 10.5 + byte read 15,000
  - b. Lost byte sent 30,000
  - a. Lost byte ACK 20,000
  - c. Lost byte ACK 15,000
  - d. Lost byte ACK 20,000

Conclusion - If we consider any edge congestion control

3M/





c) Congestion control in TCP

$$\begin{aligned}
 \text{c) } \text{Effective window size} &= \min(\text{Eff. Win. Size based on flow control}, \\
 &\quad \text{Eff. window size based on Cong. Control}) \\
 &= \min(5,000, 15,000) \text{ bytes}
 \end{aligned}$$

$$\begin{aligned}
 \text{b) i) Effective window size} &= \text{Advertised window size} - (\text{Last byte sent} - \text{Last byte received}) \\
 &= 25,000 - (30,000 - 10,000) = 15,000
 \end{aligned}$$

$$\begin{aligned}
 \text{Effective window size} &= \text{Congestion window} - (\text{Last byte sent} - \text{Last byte received}) \\
 &= \text{Max Ret Buffer} - (\text{Last Byte Received} - \text{Last Byte Read}) \\
 \text{a) ii) Minimum of congestion window} &= 15,000 - (30,000 - 20,000) = 5,000
 \end{aligned}$$

$$30,000 - (45,000 - 20,000) = 15,000$$

$$\text{Max Rev Buffer} - (\text{Last Byte Received} - \text{Last Byte Read}) = 25,000$$

$$\text{b) i) Calculate Ret Buffer - (Last Byte Received - Last Byte Read)} = 5,000$$

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$$\text{b) i) Calculate Ret Buffer - (Last Byte Received - Last Byte Read)} = 5,000$$

$$\text{Page No}$$

$$\text{Date}$$

- Two states:
    - i) TCP has two state: slow start and congestion avoidance
    - ii) A window size threshold governs the state transition
  - ATM (Additive increase of multiplicative decrease).
  - 1 to 20, packet loss, 20 to 10, 10 to 30, packet loss, 30 to 15, 15 to 20, packet loss, 20 to 10, 10 to 16 packet loss.
  - Slow start:
    - Tinitial CW is 1
    - Slow start phase: 1, 2, 4, 8, 16, 32, 64
    - When it is about to reach  $2^{n-1}$  window size reduces to 1
    - First time reach threshold, New threshold is set to half of previous:  $40/2 = 20$ .
    - When packet lost it will go to half of previous window size.
    - Slow start + ATM is used here.
  - Fast Recovery (No threshold).
    - Slow start + ATM is used here.
    - Slow start until congestion window reaches congestion threshold or advertised window.
    - Ex. 20 packets are send, if packet is lost, 2nd and 27 packets are required to send 30 packets.
    - RTT's are measured to send 30 packets.
    - ① 1.
    - ② 2, 3.
    - ③ 4, 5, 6.
    - ④ 5.

RTT	Seq no. of packet
⑤	6, 7
⑥	8, 9, 10
⑦	11, 12, 13, 14
⑧	⑯, 16, 17, 18, 19
⑨	15, 16
10	17, 18, 19
11	20, 21, ⑯, 23
12	22, 23
13	24, 25, 26
14	⑯, 28, 29, 30
15	27, 28
16	29, 30,

Q. Slow start + Congestion avoidance.  
packet lost : 5, 15, 22, 27

RTT	Seq No. of packet
1	1
2	2, 3
3	4, ⑯, 6, 7
4	5 cong. Threshold = 2
5	6, 7
6	8, 9, 10
7	11, 12, 13, 14
8	⑯, 16, 17, 18, 19 cong. Threshold = 2 (Threshold = $cw = 2$ )
9	15
10	16, 17
11	18, 19, 20
12	21, ⑯, 23, 24 cong. Threshold = 2
13	22
14	23, 24
15	25, 26, ⑯ cong. Threshold = 1 (Threshold = $cw / 2 = 1$ )

16	27
17	28, 29
18	30

The throughput =  $\frac{(30 \text{ packet} * 1 \text{ KB/packet})}{(18 \text{ RTTs} * 100 \text{ ms/RTT})} = 136533 \text{ bits/sec}$

27-08

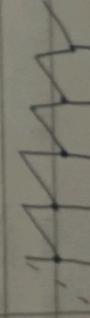
### \* Fast Recovery

C. Solve same problem using fast Recovery 5, 15, 22, 27. How many RTTs will be required to sent 30 packets.

①	1
2	2, 3
3	4, ⑤, 6, 7
4	5, 6
5	7, 8, 9
6	10, 11, 12, 13
7	14, ⑥, 16, 17, 18
8	15, 16
9	17, 18, 19
10	20, 21, ⑦, 23
11	22, 23
12	24, 25, 26
13	⑧, 27, 28, 29, 30
14	27, 28
15	29, 30

161 Exponential increase

2nd Additive increase



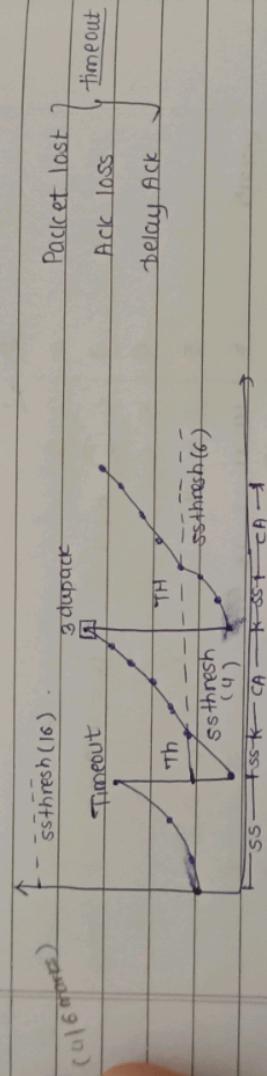
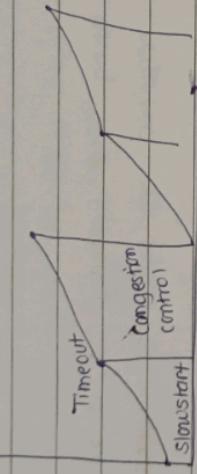
### TCP Tahoe ( 4 - 6 marks )

Show start and congestion avoidance.

TCP Tahoe treats the two signals used for congestion detection,

time-out three duplicate segment ack.

and to 1 mss



### TCP Reno

two signals of congestion

time-out the arrival of three duplicate ACKs

T+ behaves like the slow start, in which and grows exponentially and start with the value ssthresh plus 3 mss (instead of 1)

Q. Same problem → Difference between TCP Tahoe & Reno?

a. What is the threshold value in TCP Tahoe and Reno?

" TCP Tahoe : ss + f1 + fast Retransmit "

Step 1 - Assume that the cwnd = 8 and sender has sent Segments 31-38, and 31 is lost.

Step 2 - Receiver will reply with 7 duplicate ACKs of segment 30 (indicating the it is waiting for 31).

Step 3 - On receiving 3 duplicate ACKs, window cwnd is set to 1 if 31 is retransmitted, and ss phase is started.

Step 4 - Restart in slow start by sending segment 39.

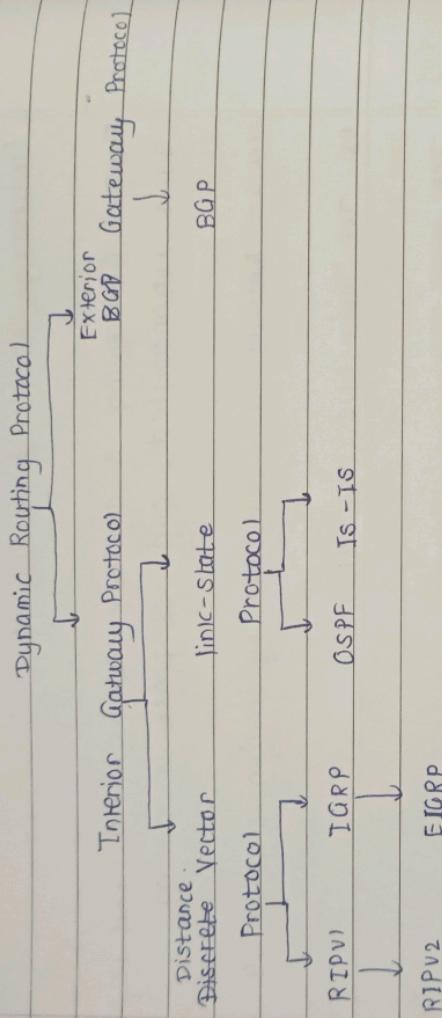
Ex - TCP Tahoe (1/2) , (2/2)

- TCP state Transition Diagram (do it own)  
Connection establishment and connection reestablishment  
(do it own)
  - web caching and DNS caching - chapter 2.
  - CDN
  - Bit torrent in p-to-p architecture.

## Chapter 34

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- Static Vs Dynamic Routes:
- Routing Protocol
  - Q. Diagram is given where BGP and IGP used?



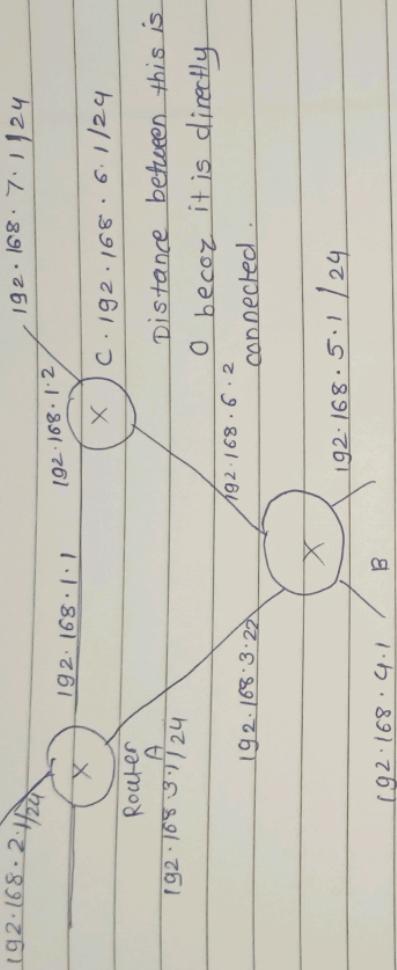
### 1) Distance Vector Routing Protocols:

Provides two characteristics → Distance → how far destination is.  
Vector → direction of the next-hop  
 $R_1$  and  $R_2$  should be in one subnet.

- 2) Link State Routing Protocol.
- 3) Interior Gateway Protocols
- 4) Exterior gateway protocol is used to do this:  
ASX and ASy

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Date	

a. How Asx provide autonomous information to Asy?



- Metrics - is a variable assigned to routes as a means of worst or from most preferred to least preferred.

- Hop Count -

- Bandwidth

- Load

- Autonomous System (AS)
    - collecting of routers whose prefixes & routing policies are under common administrative protocol.
    - com. (Interior Gateway Protocol)
- Dynamic IGP :- Protocol used to distribute routing information within an AS.

Dynamic EGP :- [Exterior Gateway Protocol]

• ETC

- weighted failure queue
- round robin
- priority
- first come, first serve
- Packet Scheduling : FCFS

3. Random

2. marking

1. drop  $\searrow$  priority

Buffer Management

24-09

\* Output port queuing

- queuing and loss due to off buffer overflow
- Head-of-the-line (HOL) blocking: (deadlock)
- queuing delay or loss due to input buffer overflow

\* Input port queuing

- expediting
- multistage switch

• Switching via intermediate network or cache

- 32 Mbps
- bus connection: switching speed limited by bus bandwidth
- via shared bus
- datagram from IP port memory to off-port memory
- Switching via bus

- Queue Operation
    1. Packets are stored in the order they arrive
    2. After the transmission of packets
  - Prioritized Traffic
    - guaranteed Queueing
      - Two priority classes = 0 high priority, 2 low priority
    - packet arrival and Transmission
  - Queue Classification
 

0 High priority	2 low priority class
pkt 1	pkt 2
pkt 3	pkt 4
pkt 5	pkt 6
  - Scheduling Policies: Round Robin Discipline
    - places all sorted into classes similar to file priority queueing
    - Same priorities
    - Consider of two classes Round Robin Queueing
    - class 1
    - classes 2
    - classes 3
    - classes 4
    - classes 5
  - Weighted Fair Queueing
    - gives 1 - class 1 - part 1, class 2, part 4
    - class 2 - part 2 - part 5
    - class 3 - part 3 - part 6
    - class 4 - part 4 - part 7
    - class 5 - part 5 - part 8
  - Polling Mechanism
  - Fragmentation
    - What is network neutrality?
    - Technical: how an ISP should share allocation of its resources  $\rightarrow$  (Access to the Internet, the Content)
    - Social, Economic principles
      - protecting free speech
      - encouraging innovation, competition
      - reinforced legal rules and policies
    - Different countries have different "rules" on network neutrality
  - IPv4 Header format
    - IPv4 checksum is applied to only header for entire segment.
    - Version -
    - Header length - min 20 bytes ... max 60 bytes.
    - Calculating version, header length, IP address, Source and destination address - find 2.
    - When an IP datagram fragmented?
    - DF Bit -
      - DF = 1 - no fragmentation
      - DF = 0 - allow for fragmentation.
    - MF Bit
      - MF = 0 - last fragment
      - MF = 1 - given permission for more fragmentation in process regular.
  - Fragment Offset - 13 bit field
  - Size of datagram is given which is the no of fragments can be done!

### Time to live

concept of scaling factor  
fragment offset value field = fragment offset / 8.

Time to live : value is decremented by 1 when packet travelling from one hop to another.

If the value of TTL becomes 0 before reaching the destination, the datagram is discarded.

### Protocol Field

- protocol number of TCP is 6, -1-2, TCP is found 6 and UDP 17

### Header checksum

Computation of header checksum includes IP header only.  
At each intermediate device checksum is checked.

Source IP

Destination

Options - This field is used for several purposes.

1. Routing
- 2.