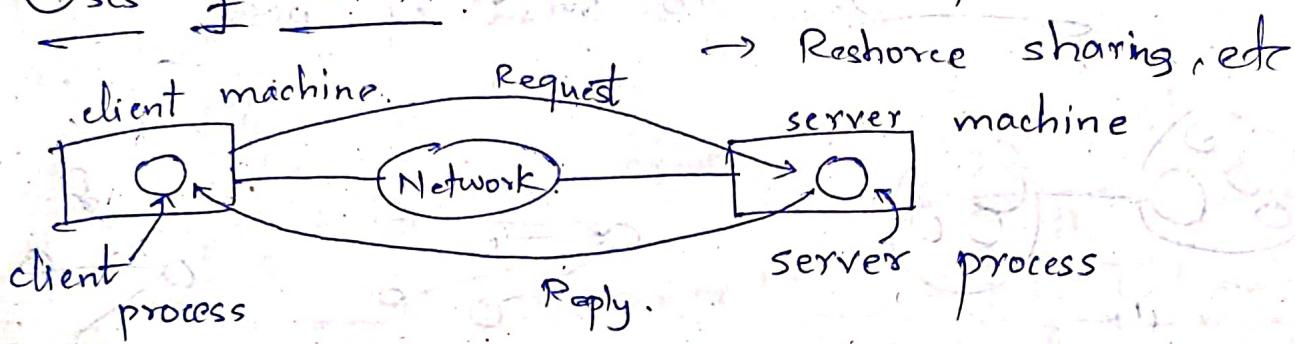


What is network? interconnection of devices.

Connected to network end systems (or) hosts.

Digital data  $\rightarrow$  parts / fragments  
 $\downarrow$   
packets

Uses of CNN: Business Applications



Network Essentials:

$\rightarrow$  Modem (Modulator + Demodulator):

Hardware device that converts data into a suitable format for transfer between two computers.

$\rightarrow$  Ethernet: system for connecting computers to form a LAN.

$\rightarrow$  Router: A device which forwards data packets along networks fragments

• Connected to atleast 2 networks commonly 2 LAN's

$\rightarrow$  Repeaters: Regenerate, replicate a signal.

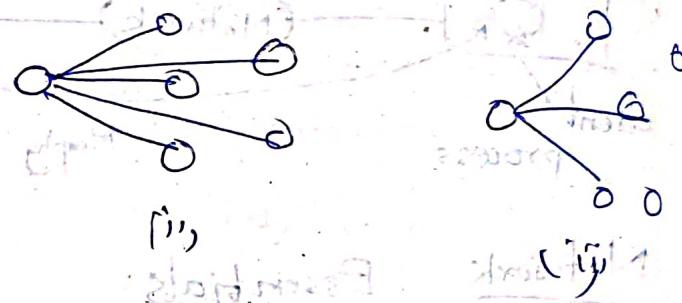
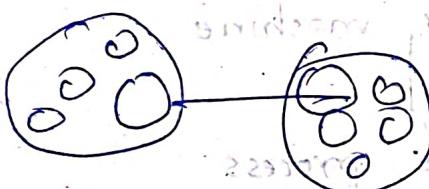
# Classification of Networks

## ① Transmission Technology

Point Unicasting: Exactly one sender & one receiver.

(ii) Broadcasting: To all hosts

(iii) Multicasting: To subset of hosts.



→ Personal Area Networks: (PAN's)

→ Local Area Network: (LAN's)

→ Metropolitan Area Network: (MAN's)

→ Wide Area Network: (WAN's)

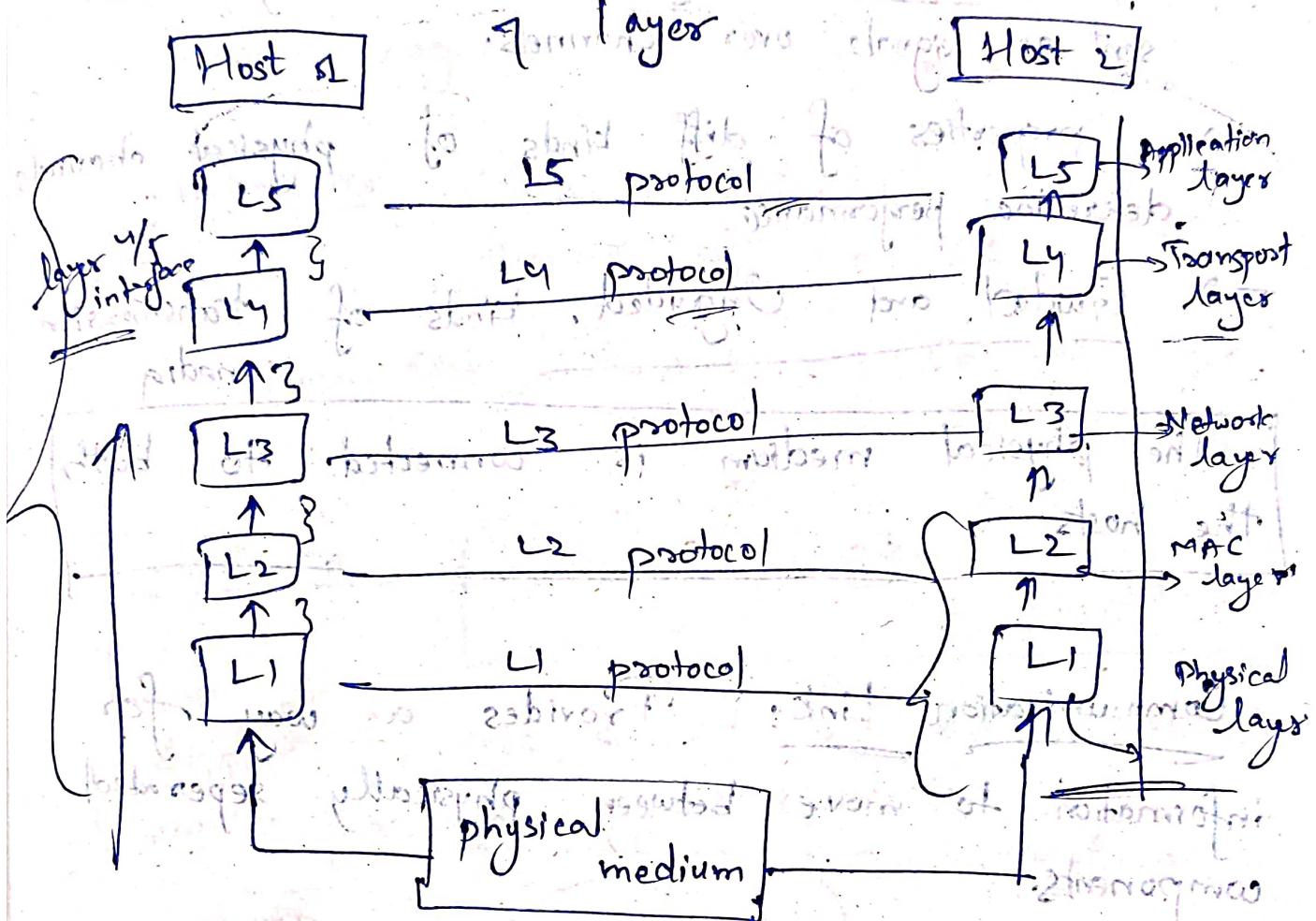
Network Software:

(Brain behind internet)

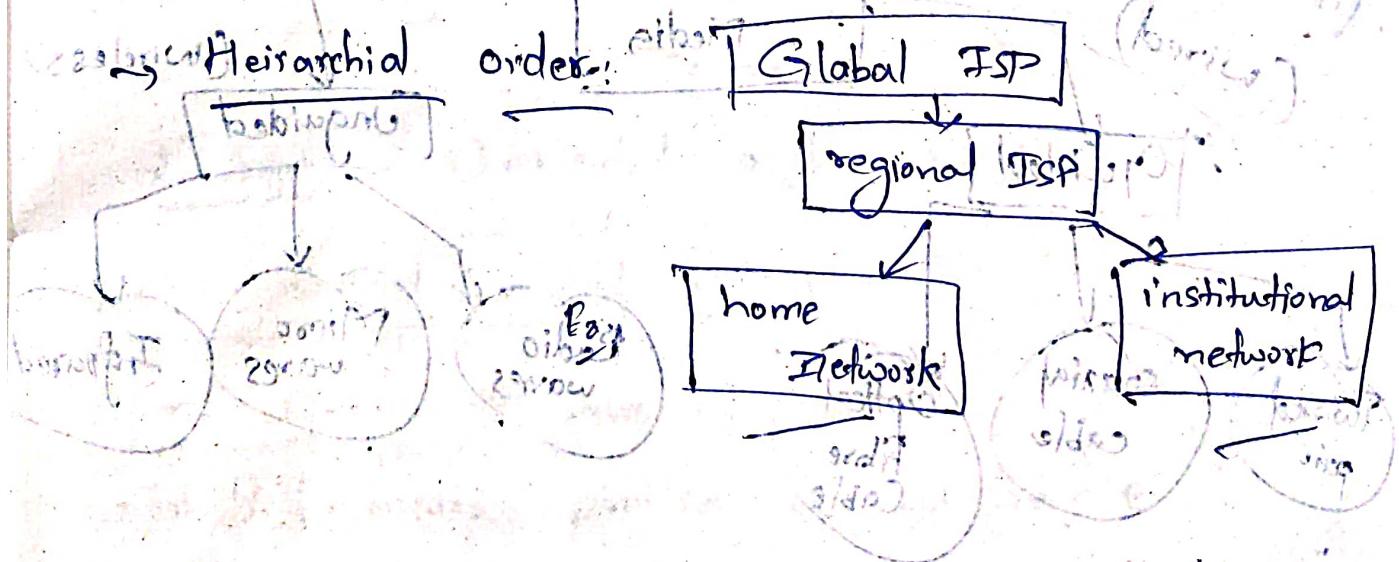
Protocol: Agreement between communicating parties on how communication is proceeded.

→ Violation makes it difficult.

To organise functionality, it is stacked as 7 layers.



### ISP - Internet Service Provider



12-01-2022

Multiplexing: Technique which combines multiple signals into one signal. → Transmission over a communication channel

↳ Large Bandwidth is saved. cost reduced, circuit complexity reduced and multiple signals can be sent over a single channel.

Analog: f division & wavelength division multiplexing.

continuous

Digital: Time division multiplexing

discontinuous

FDM (frequency division multiplexing):

- Bandwidth of communication channel should be greater than combined bandwidth of individual signals.
- Divides the bandwidth. → logical sub channels and each is separated by an unused bandwidth called guard band.

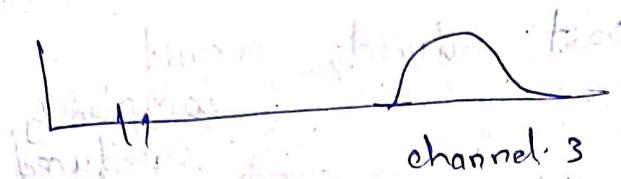
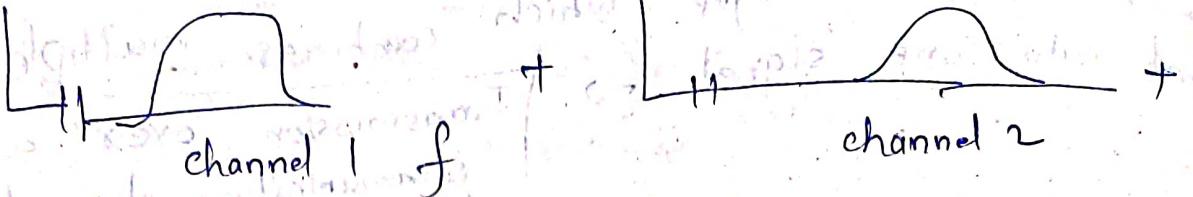
↳ A narrow frequency range which separates two signal frequencies.

Ex: 4 GHz for 100 people  

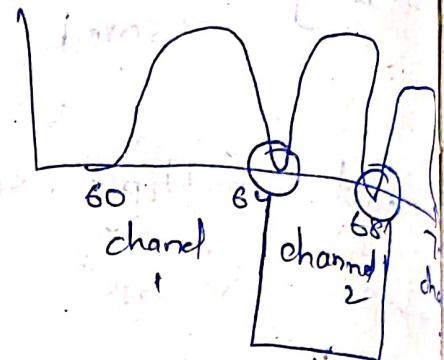
$$\text{One} = \frac{4 \times 10^9}{100} = 40 \text{ Hz}$$

but  $(35 \text{ Hz} - 40 \text{ Hz})$  guard band,  $(0-35 \text{ Hz})$

Used to separate 2 users and prevent overlapping.

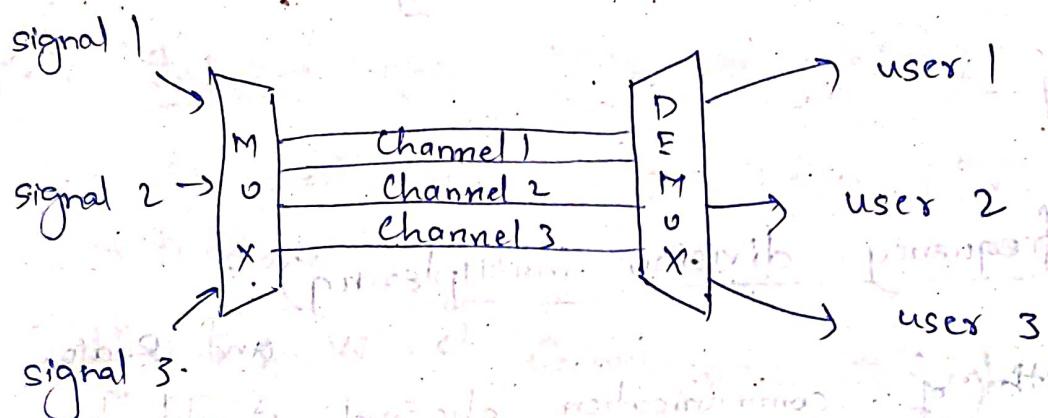


without guard band, overlapping happens.



**overlapping**

### Schematic Diagram:



→ transmit multiple signals

### Advantages:

→ FDM, demodulation process is easy.

→ Doesn't need synchronisation.

Disadvantages: Needs a large bandwidth for communication.

Applications: → FM, AM.

→ used in first gen cellular phone

→ TV Broadcasting

## WDM (Wavelength Division Multiplexing)

- A technique that increases the bandwidth of a communication channel (optical fiber) by simultaneously allowing multiple optical signals.
- Working principle of WDM is similar to FDM but here, we use optical signals instead of electric signals.

## Schematic Diagram (similar to FDM)

### 2 kinds of WDM Techniques: (WDM)

- Dense WDM (longer distance)

(one path with bandwidth of one wavelength band)

→ Dispersion Compensated WDM

(one path with bandwidth of one wavelength band)

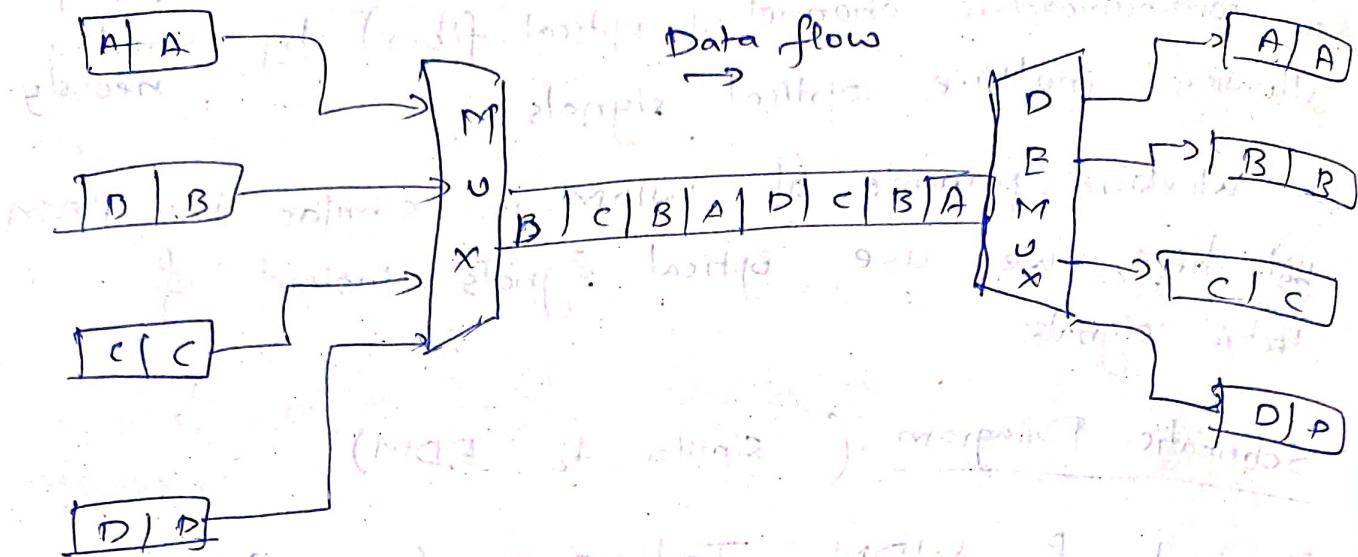
## Time Division multiplexing: (TDM)

- Multiple signals are combined and transmitted one after other on same channel

↳ Assigning a time slot strictly for one user

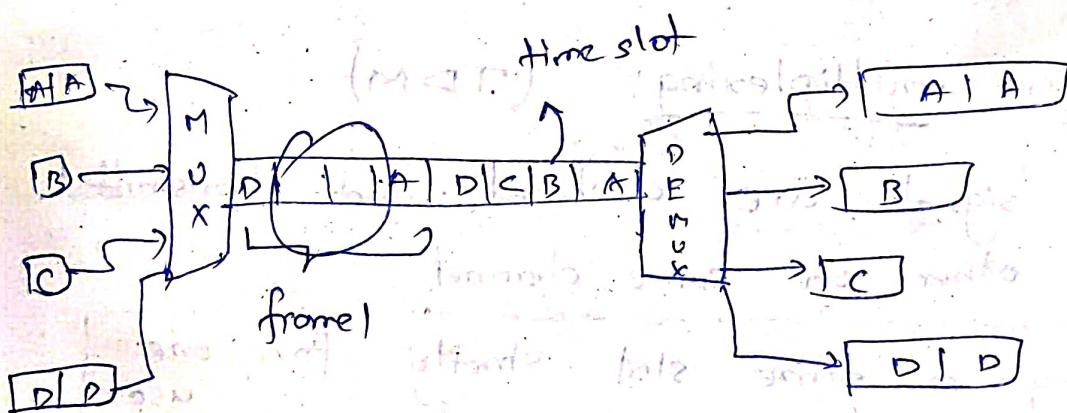
- All signals operate with same frequency but are transmitted at different times.

## Schematic Diagram



The users needs to be synchronised (so that one can be transmitted and then another one)

- 2 types:
  - Synchronous TDM (fixed time slot)
  - Asynchronous TDM (Not fixed slot time)

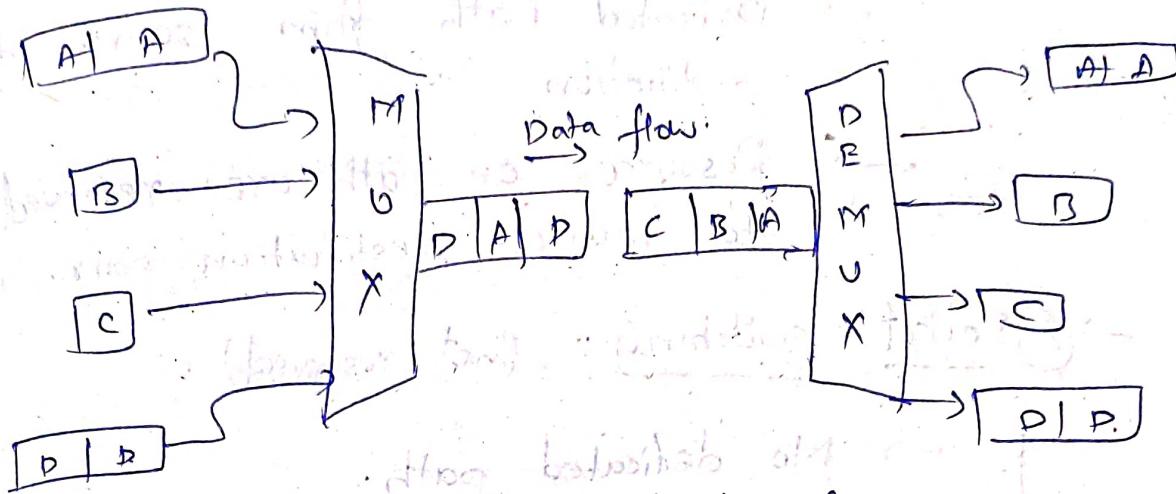


(Synchronous TDM)

$$\boxed{\text{number of time slot} = \text{number of transmitters.}}$$

## Asynchronous

time slots  $\neq$  number of transmitter



Advantages:

- full bandwidth for one user at one time.

→ TDM is flexible than FDM

→ crosstalk problem is less.

Disadvantages:

- synchronisation required.

## CDMA: Code Division Multiple Access

↳ Our information given  $\rightarrow$  We receive others  $\rightarrow$  Dot product is zero.  $\rightarrow$  orthognal codes.



→ This is called spreading and decoding.

13-01-2022

End systems connected by lines

→ ① Circuit switching: (reserved)

→ Dedicated path from source to destination

→ Resources on path are reserved to source-destination pair.

→ ② Packet switching: (not reserved).

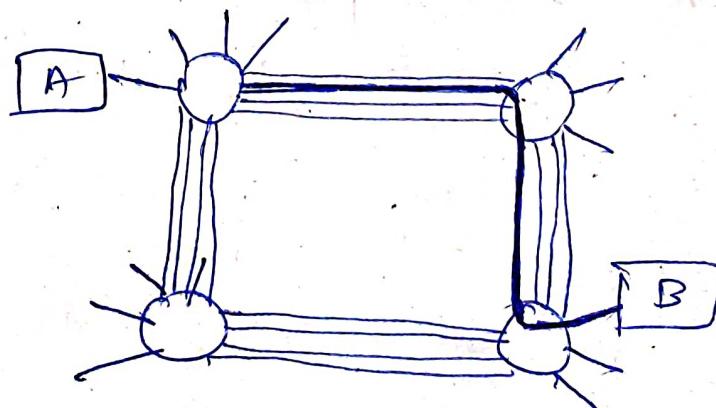
→ No dedicated path.

→ A switch/router forwards packets to another router/destination at the path.

→ Any message can take any path of the mesh of network

→ Gives the best of network path available but not guaranteed.

① Circuit switch network



4 circuits/paths available.

The connection between source and destination called as circuit.

- Resources such as bandwidth, buffers are blocked for communication on that circuit.
- Links are finite, hence very less users at once (Disadvantage).

Ex: Older Telephone Networks.

↳ Gives the best network (less delay) possible. (Advantage)

Ex: A circuit switch network transfer 640,000 bits between post A and post B. There are other links. It is following TDM which has 24 slots. Maximum supported speed is 1.536 MBPS. and time to establish a network is 500 ms. calculate the total time taken.

(A) Total time taken = Establishment + Transfer time

Total slots = 24

1 slot speed =  $\frac{1.536}{24}$  = 2 circuit speed

= 64 kbps (approx)

Total time =  $\frac{10}{64 \text{ kbps}} = 10 \text{ s.}$

Total time = 10.5 s

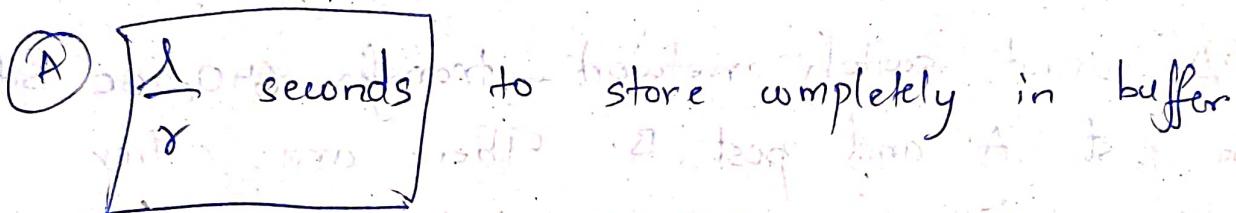
② packet switch network

Inte won't share partial amount of packet.  
There is a store and forward kind of system.

→ Before receiving the entire amount of packet, it is stored in buffer and queuing.

→ Only after receiving entire packet, it forwards.

Each packet is having  $\lambda$  bits and link capacity is  $r$  bits per second. How much time will it take?



and  $\frac{\lambda}{r}$  from buffer to destination

$$\boxed{\text{Total } \frac{2\lambda}{r} \text{ from source to destination}}$$

first packet:  $\frac{\lambda}{r}$  and subsequent ones follow kind of pipelining

Why can't bit by bit?

(A) Because when transferred bit by bit, the router or any other intermediate network component doesn't know the receiver nor the sender.

The reference to what destination the packet is sending to is. IP Address:

That IP address should be embedded into packet.

The input queue will be filled faster.

Output queue is a lot slower (6 times less) than input queue.

So there is a delay called queuing delay.

Note: Internet is a packet switching network.

17-01-2022

### Statistical Multiplexing:

By: Users = 1 Mbps link.

→ User is active for 10% of time & user generate 100 kbps.

Circuit switching means 100Mbps reserved for each user all the time → can support 10 users simultaneously.

$$P(x \geq 35) = \sum_{i=11}^{35} C_i (0.1)^i (0.9)^{35-i}$$

↪ probability that 11 or more users are active.

simultaneously in packet switching  $\Rightarrow 0.0004$

⇒ We can support 35 users with  $P = 0.0004$

→ Circuit switching is used for time critical applications

↳ Videos, audio calls.

→ packet switching problems

↳ queuing delays  
↳ packet loss  
↳ not suitable for delay constrained applications.

## Layered Network Architecture:

Network organisation is a complicated task.

→ Divide and conquer

academic

finance

administration

- A layer provides a service (set of actions) to the immediate higher level.
- Each layer can be analysed and tested independently and newer technologies can be adopted easily.

## Open System Interconnection (OSI) model:

7 layers:

Application layer

Presentation layer

Session layer

Transport layer

Network layer

Data link layer

Physical layer

heirarchy

Most important

## Application Layers:

① user programs that work in hand

Ex: FTP, SMTP are protocols  
(file transfer) → (simple mail)

## Presentation layer:

↳ syntax and information transmitted

↳ translation

↳ Encoding data: Compression / conversion; encryption and decryption.

## Session layer:

↳ Allows to establish a link

↳ Dialogue control: Bidirectional pr, unidirectional

↳ Token management: Both sides do not attempt same operation. It gives token to perform such actions

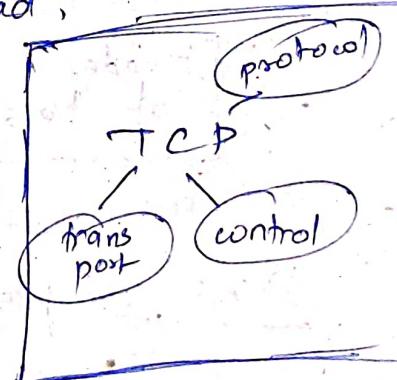
↳ Synchronisation: Pausing and resuming a download

## Transport Layer:

↳ Connection oriented services

↳ flow control

↳ guaranteed delivery



↳ Ensures data delivery is

↳ ~~data~~ free error

↳ in sequence

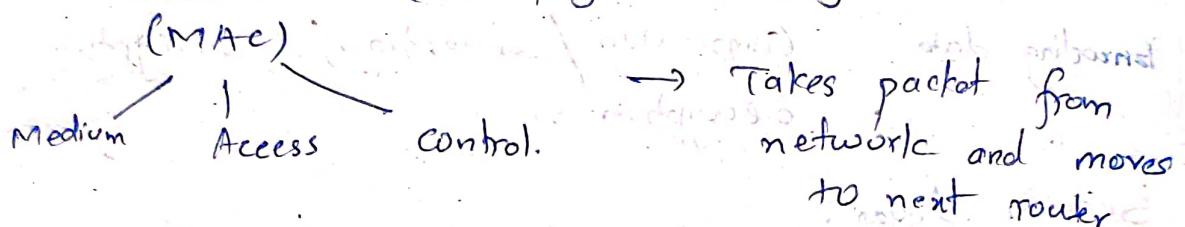
↳ no loss, duplication and corruption of packets

## ⑤ Network layer:

- ↳ Interface between host and network
- ↳ Routing
- ↳ Congestion and Deadlock
- ↳ Internetworking.

## ⑥

### Data-link and physical layer:



Physical layer: error free delivery

Physical layer: error correction policy

physical layer: Electric signals with created  
signal modulation.

## 18)

### Internet protocol stack:

Header of transport layer + payload = segment

### Delays in packet switching network

→ processing, queueing, transmission,  
propagation, delay.

Nodal delay = sum of all delays

(or)

end to end delay

→ delay from source to destination.

processing delay: Time required to examine the packet header. (unwrap)

- ↳ Determines where to direct the packet.
- ↳ check for errors.
  - ↳ In microseconds.

Queuing delay: If a router is busy in processing and transmitting a packet it will wait in queue.

No queuing delay if router is idle.

Order is microseconds to milliseconds

generally a random variable

Transmission delay: delay in pushing the packet into the link.

length of packet =  $l$  bits, and transmission rate =  $r$

$$\text{Transmission delay} = \frac{l}{r}$$

→ microseconds to milliseconds

propagation delay: Time required to propagate from one end of the link to the other end.

↳ depends on physical link between the routers.

$2 \times 10^8 - 3 \times 10^8$  m/s,  
(range)

→ depends on distance

$$\text{delay} = \frac{d}{s}$$

## Traffic intensity:

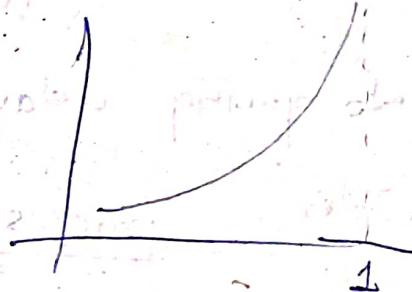
- ↳ Queuing delays are random
- ↳ indication of queuing delay

Let "a" be the average number of packets arriving at a queue.

Each packet length =  $L$ , transmission rate =  $R$ .

$$\boxed{\text{Traffic intensity} = \frac{aL}{R}}$$

If  $TI > 1$ , queue length increases to  $\infty$



If traffic intensity close to 1, there will be significant queue delay.

Throughput: Rate of successful message delivery

Instantaneous throughput is at rate at which

Host B is receiving data.

Suppose it takes  $T$  seconds to transfer  $R$  bits

Minimum of available limit is called throughput

$$\min(R_1, \dots, R_n) = \underline{\text{Throughput}}$$

Q1: ①  $N$  packets simultaneously transmitted (or, ② queued. packet length =  $L$ . Transmission rate =  $R$ .

What is average queuing delay for  $N$  packets.

① 1<sup>st</sup> packet delay = 0

2<sup>nd</sup> packet delay =  $\frac{l}{R}$

$$\frac{2l}{R}$$

Average delay =  $\frac{\text{sum of all packet delay}}{\text{no of packets}}$

$$= \frac{\frac{N(N-1)}{2} L}{2 R} \Rightarrow \left(\frac{N-1}{2}\right) \frac{L}{2R}$$

Ans

Q2: Path has 3 links.  $R_1 = 500 \text{ kbps}$

$$R_2 = 2 \text{ Mbps}$$

$$R_3 = 1 \text{ Mbps}$$

Throughput = ?

A. Throughput = 500 kbps

(ii). File size = 4 million bytes,

$$\text{Time taken} = \frac{\text{File size}}{\text{Throughput}} = \frac{4 \times 10^6 \text{ bytes}}{500 \times 10^3 \text{ bps}}$$

$$= 8 \text{ s}$$

③ packet length = 1000 bytes

propogation speed =  $2.5 \times 10^8$  m/s

length of link = 2500 km.

Transmission rate = 2 Mbps.

$$\text{propogation delay} = \frac{\text{distance}}{\text{speed}} = \frac{2500 \times 10^3}{2.5 \times 10^8} \text{ s}$$

$$= 0.01 \text{ s}$$

Note : propogation delay =  $d_p = \frac{m}{s}$

transmission delay =  $d_t = \frac{L}{R}$

$$\boxed{\text{end to end delay} = \frac{L}{R} + \frac{m}{s} = \text{d}_{\text{e2e}}}$$

If  $t=0$  at start, then  $t=d_{\text{trans}}$ , the last bit just ends A, just out of A placed on link

at  $t = d_{\text{trans}}$  if  $d_{\text{prop}} > d_{\text{trans}}$ , the first bit is on the link; if  $d_{\text{prop}} < d_{\text{trans}}$ , the first bit will be outside of A, closer to B.

Ex)  $s = 2.5 \times 10^8$  m/s  $L = 120$  bits  $R = 56$  kbps

distance = ?  $d_{\text{prop}} = d_{\text{transmit}}$   $m = ?$

$$m = \frac{L}{R}$$

$$m = \frac{120}{56 \times 10^3} \times 2.5 \times 10^8 \text{ m}$$

$$= \frac{1.1 \times 2.5 \times 10^7}{7.1}$$

Ex: packet length = 1500 bytes. Transmission link = 2 Mbps.  
no of packets in queue. What is queuing delay  
of the packet which is about to enter?

Transfer 4.5 packets needs to be done.

$\Rightarrow 4 \times 1500 + 750$  bytes needs to be transmitted.

$$\Rightarrow \text{Queuing delay} = \frac{[4 \times 1500 + 750]}{2 \times 10^6}$$

General expression:

$$\frac{[(L-\alpha) + nL]}{R}$$

### Traceroute:

→ runs in any internet host  
sends a packet to router 1 and router 1 sends back  
or transmit back packet and then calculate end-to-end delay.

Q1

### Networks under attack

malicious stuff  $\rightarrow$  malware

↳ deleting our files

↳ spyware: deletes security id, numbers, etc.

↳ compromised host can be part of organisation called as botnet.

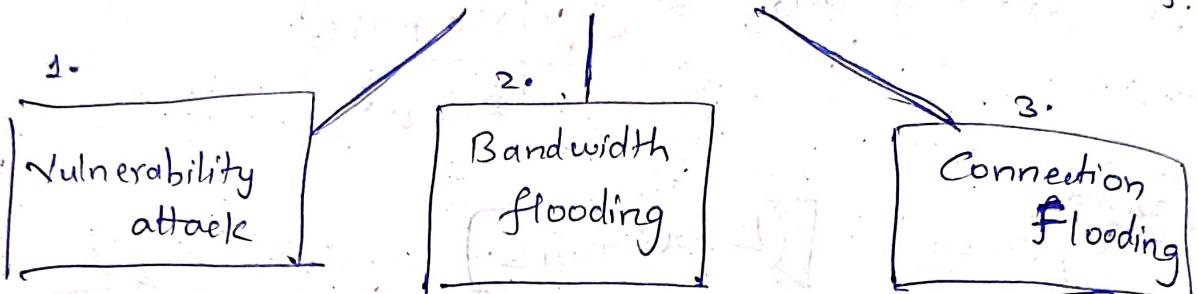
self replicating: clicking a link, entering details

virus: requires user interaction

worms: requires no user interaction

## Denial of Service Attacks (DoS attacks)

- ↳ brings down a service
- ↳ renders a service unusable by legitimate users.



bombarding with unnecessary packet like giving an ununderstandable packet which can't be decoded

→ can stop/ crash the host

sends or packets overflow and floods, so that actual users can't connect

using TCP which is connection based communication

(1<sup>st</sup> conn & 2<sup>nd</sup> comm)

Affecting the host so that it will not accept the legitimate connect users.

## Distributed DoS (DDoS)

leveraging botnets with 1000's of hosts and that then does attack on single host.

↳ Much harder to stop and attack.

## Packet sniffers

- ↳ can be wired (or) wireless obtain a
- ↳ an imposter passive receiver which can copy every packet that is transmitted.

Sniffed packets can be analysed offline for sensitive information (personal private information, trade secrets, passwords)

Wireshark is an example of packet sniffer.

→ analyze layer by layer protocol and give us a report (open source)

→ packet sniffers are passive:

do not inject  
packets into the  
channel

Difficult  
to test

→ One way to defense against it is cryptography

## Application Layer

Network Application management: Writing programs

that run on different end systems will communicate with each other over the network.

(Example in slides)

Network core devices do not function at the application layer

Applications use the service of network (Transport layers)

## Architecture of Application:

→ Client - Server architecture

→ peer-to-peer architecture

Application developer decides either of these 2

## Client - server Architecture:

Server: An end system that serves the requests from various hosts.

↳ always on.

Client: An end system that requests a server for content.

↳ can be ON-OFF or always ON.

Examples: e-mail, file transfer, etc.

## P2P Architecture:

→ End system has direct communication.

↳ peers.

Example: Skype, Torrents, etc.

## Advantages

→ Full distribution.

→ self scalable.

→ Cost effective.

## Disadvantages

→ ISP friendly: asymmetric data traffic.

→ Less security

→ Incentives: peers should share bandwidth

2/1

MAC

Frame : Header + payload in MAC LayerDatagram : Network Header + payload in Network LayerExample problem :① Optical fibre, core has  $\mu_{\text{core}} = 1.7$  and  $\mu_{\text{cladding}} = 1.2$ 

What is the speed of light inside the core?

 $\mu_{\text{core}} = \frac{\text{speed of light}}{\text{speed inside core}}$ 

$$1.7 = \frac{3 \times 10^8}{s} \Rightarrow s = \frac{3}{1.7} \times 10^8$$

$$\boxed{s_{\text{core}} = 1.76 \times 10^8}$$

$$\boxed{s_{\text{cladding}} = 2.5 \times 10^8}$$

What is core-cladding interface angle?  $\Rightarrow$  critical angle.  
 so that total internal reflection happens.

$$\sin \theta_c = \frac{1.76}{2.5} = 0.704$$

$$\theta_c = \sin^{-1}(0.704)$$

but the interface angle  $= 90 - \theta_c = \theta$ .

② 10 sig with Multiplexed on single channel  
4000 Hz.

What is minimum bandwidth required for multiplexing  
a) with guard band

b) without guardband

A) without guardband  $\Rightarrow 10 \text{ sig} \times 4000 \text{ Hz}$   
 $= 40000 \text{ Hz} = 40 \text{ kHz}$

with guardband : no of bands required = 9

each guardband = 10% of 4000 Hz = 400 Hz

Total bandwidth =  $40 \text{ kHz} + 3600 \text{ Hz}$   
 $= 40.36 \text{ kHz}$

③ 4 channels  $\rightarrow$  2 channels = 200 kbps each

$\rightarrow$  2 channels = 150 kbps each

(no synchronisation bits)

Assume 4 bits from 200 kbps & 3 bits from  
150 kbps channel. 7 bits from 200  
3 bits from 150 kbps.

Questions:

a) What is size of frame in bits? 14 bits  
in frame

b) What is frame rate? 50000 frames / second.  $\Rightarrow f$

c) What is duration of frame?  $\frac{1}{50000} = 20 \mu\text{s}$

d) What is datarate of entire multiplexing?  
 $50000 \times 14 = 700,000$   
 $= 700 \text{ kbps}$

process communicating

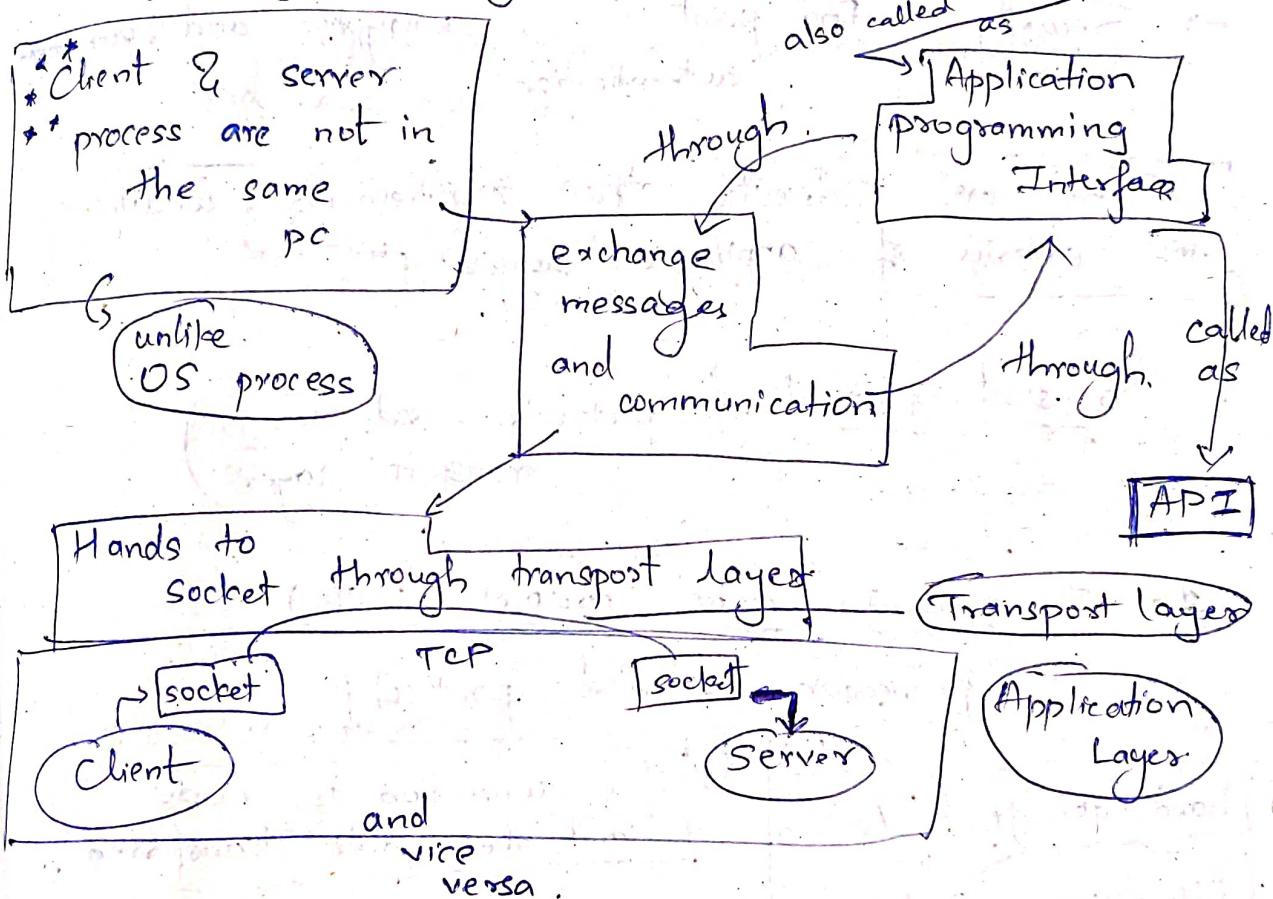
↳ A program running within the end system.

Client process → running on a client

server process → running on a server.

actual communicators.

→ sends and receives messages to and from transport layer through software interface ⇒ **socket**.



Application processes, sockets  
and underlying transport layer

Services of Transport layer (w.r.t Application layer)

→ Reliable data Transfer : Guaranteed Delivery.  
(Application layer's role ends after giving it to socket)

→ Throughput : ensures that minimum throughput is maintained.

Ex: Telephony → minimum throughput is required for transmission.

Video application service.

→ Timing : Maintaining maximum delay the application can tolerate.

Transport layer guarantees the packet be delivered in the given time.

→ Security : End point , encryption and decryption authentication

These are indirectly the requirements , conditions and criteria of application layers.

Transport protocols ( protocols used in transport layer )

### 1. Transmission control protocol (TCP)

↳ connection oriented protocol

handshaking

full duplex connection

client to server  
and vice versa

first connection is made  
and then data transferred

rearrangement of packets in  
the desired way

orderliness of  
packets

→ Reliable data transfer : no data corruption.

→ Maintains

congestion control

lowess the higher

highers the lower

## 2. User Datagram protocol ( UDP )

↳ connectionless protocol →

no handshaking

no congestion control

unreliable data transfer

no acknowledgement from the destination.

• TCP is complex to implement, time consuming and less data loss

• UDP is easier to implement, easier to connect, faster data transfer and can tolerate minute data losses.

Ex: Skype is using video service  
↳ uses UDP  
→ faster connection and data transfer

Youtube uses TCP  
→ slower than UDP  
can tolerate glitching (data loss)

Addressing process: Helps you identify what process is directed to what destination. process for communication.

Identify Host by IP Address

Identify processes by port numbers

Ex: Email: 25, web server: 80

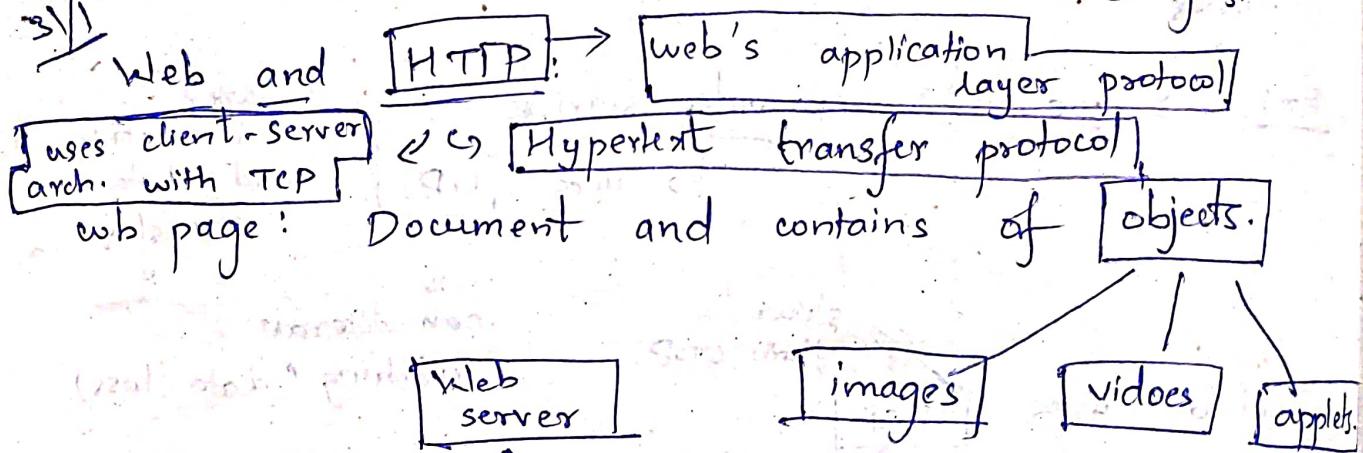
## Application Layer - Introduction defines:

↳ types of messages exchanged : request and response based.

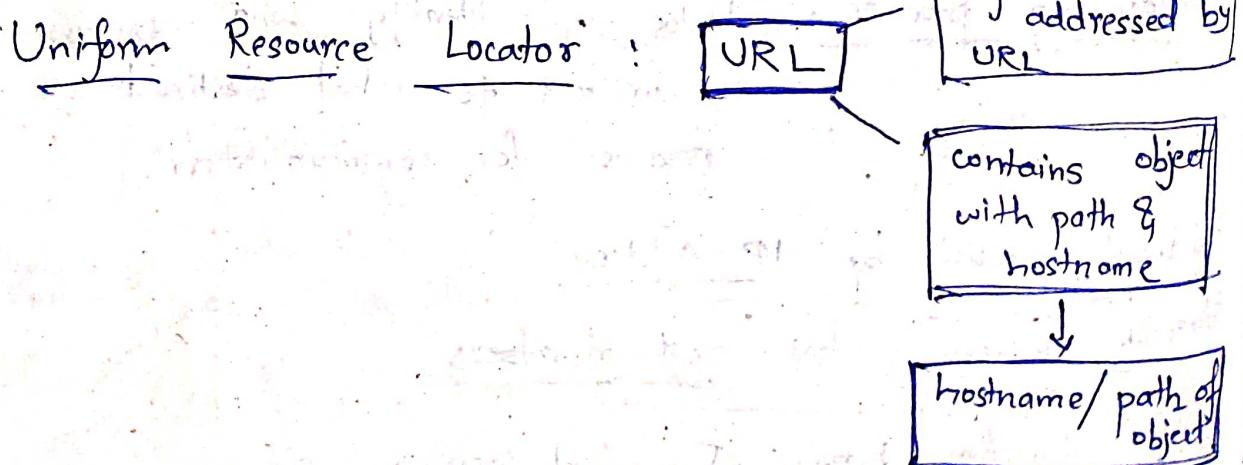
↳ syntax of various message types fields in messages, how fields are delineated.

↳ semantics of meaning of information in the field.

↳ Rules for determining when and how a process sends and receive messages.



• Client and server communicate by exchanging HTTP messages  
in Web : Browser - Opera, Chrome, etc.



## HTTP

- HTTP client initiates a connection with HTTP server → **Handshaking** → **Establishing a firm connection.** (using TCP)

- Once connection is established, client and server communicate through **socket interface**

- Client / server need not worry about packets.

(does not have any control) after sending through socket.

Client sends request and receives messages.

Server receives request and send messages

Stateless protocol: Server sends requested file without storing state information of client

Ex: Requesting for a file two times and server still sends 2 of them.

## HTTP connection

Let's say web page : 1. HTML and 10 pages.

2 ways :

**persistent**:

One TCP connection and send all files  
Overhead : minimal.

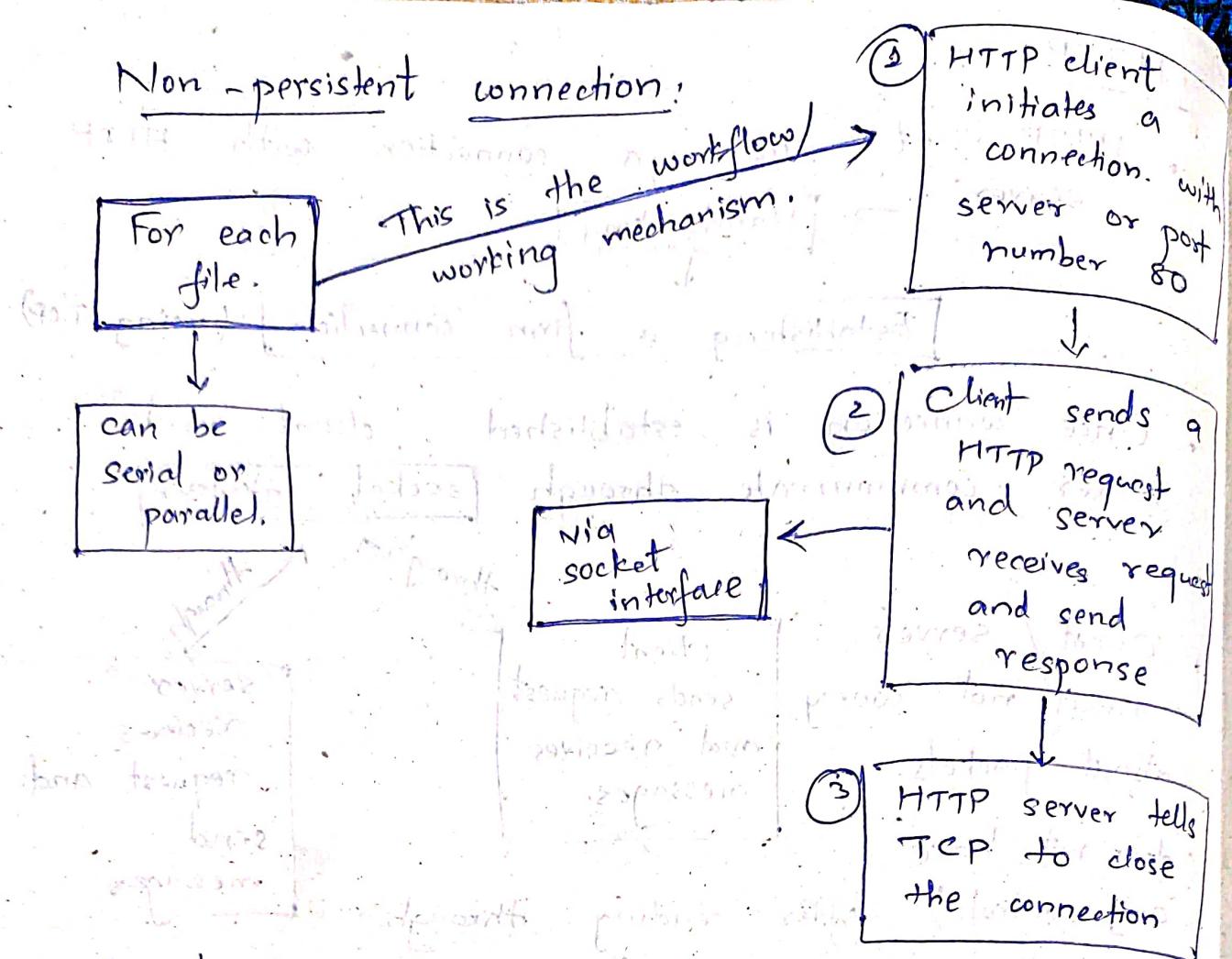
**non-persistent**:

Open TCP → take the connection, give to file → client  
→ close

Overhead : maximal

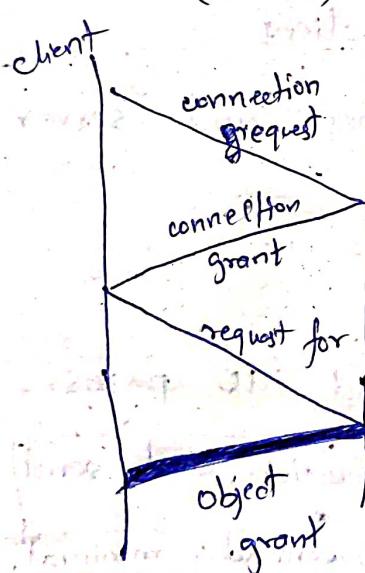
→ Opens 10 times. connection and closes

## Non-persistent connection:



## Round Trip Time: (Image in slides)

(RTT)



server: Total time for 1 object

receipt =

RTT (for connection) +

RTT (for object)

∴ Total time for 1 object = 2 RTT

10 objects = 20 RTT

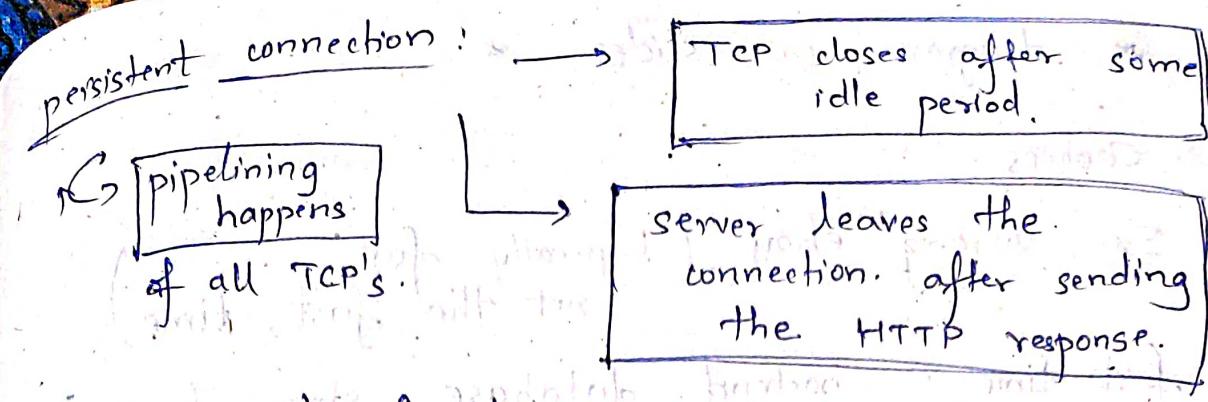
serial non-persistent connection.

Note:

Default mode

HTTP

persistent connection + pipelining



### HTTP Request format:

#### HTTP Request Message:

```

GET / somedir/page.htm HTTP/1.1
    path.
    Version.

Host: www.iits.ac.in
Connection: close
User-agent: Mozilla/4.0
Accept-language: En
  
```

- Methods: GET, PUT, POST, DELETE, etc.

- (diagram in slides)

1/2

### HTTP Response

HTTP/1.1.200 OK.

Connection: close.

Date: Sat, 07 Jul 2007 12:00:15 GMT

Server: Apache/1.3.0 (unix)

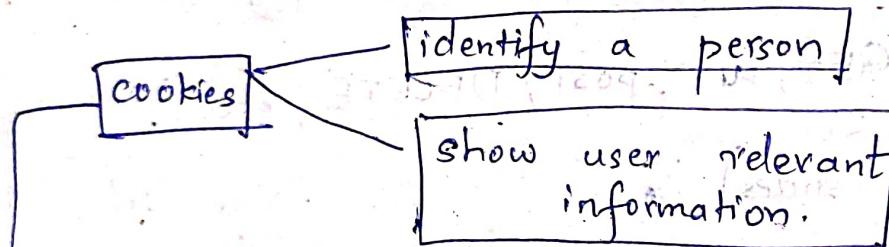
## \* diagram in slides \*

### Cookies :

Ex: visiting ebay (typically for not the first time)

first time : backend database store for some time and does backend processing.

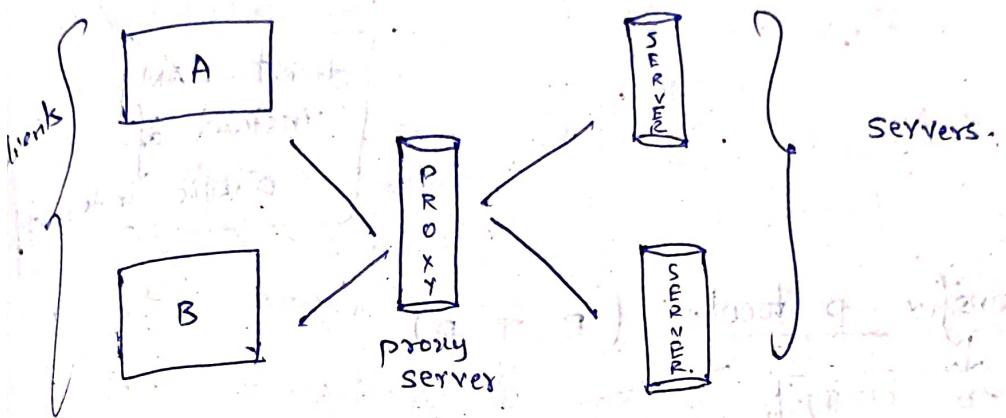
- Required to not to give a response only, rather to remember/the user (Ex: Amazon) identify to enhance the recommendations.
- There will be given an unique id / number for amazon server for further purposes.



- The unique number is shared with both Backend server and also client side.
- Downside! Selling of user data and misuse

Happens on HTTP Message only

- Web Caching (Cache) → concept of storing frequently used client memory.
- ↳ Optimising the request-response for 1000 of clients.
  - ↳ also called as proxy servers.



Initiates a TCP connection (handshaking) with the client and proxy server and if we get a hit, response is given immediately, but if it's a miss, it then opens a TCP connection with the proxy server and the server (analogous to main memory).

Ex: Given avg object size = 1 Mbps  
 Link = 15 Mbps      avg request rate = 15 objects per sec.

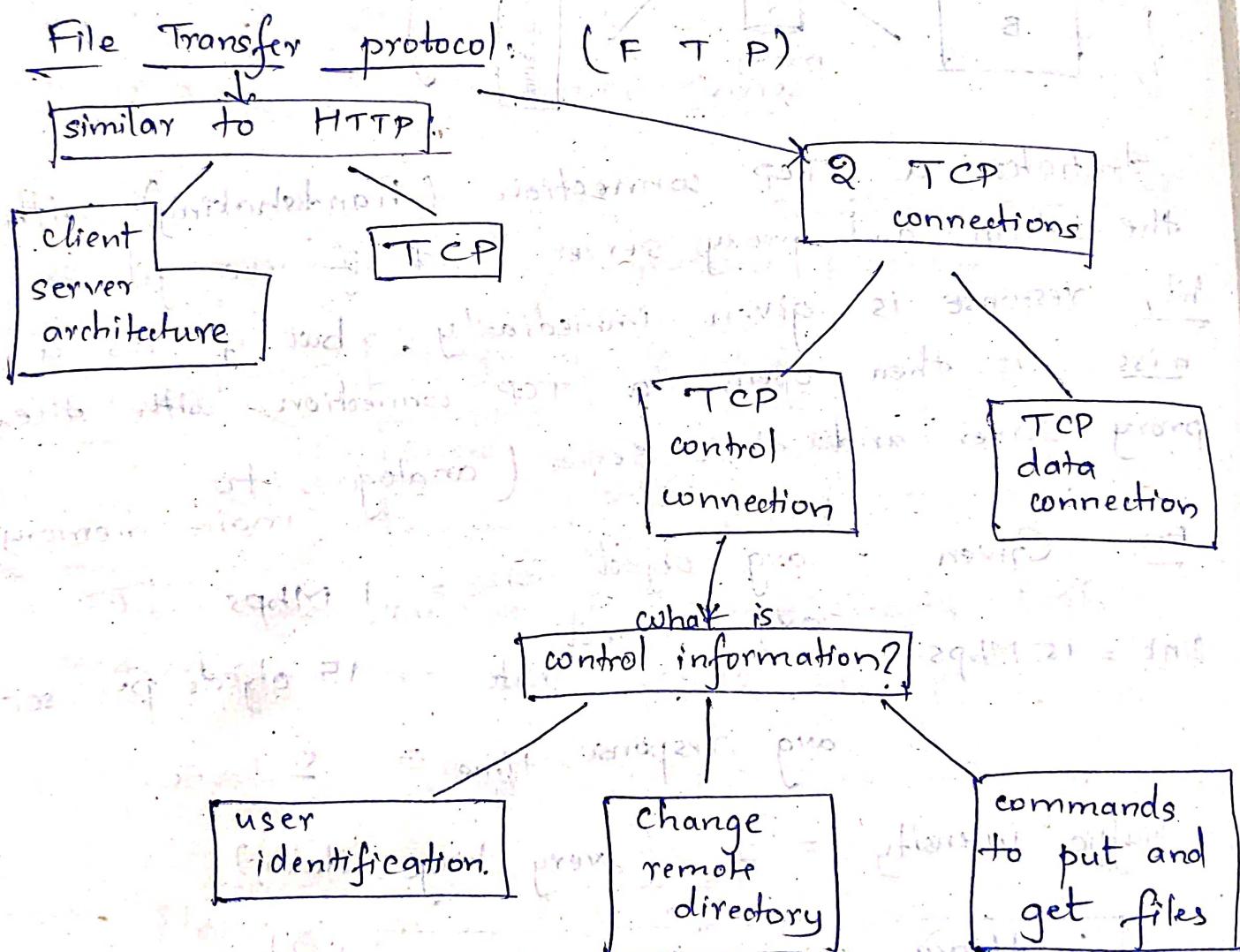
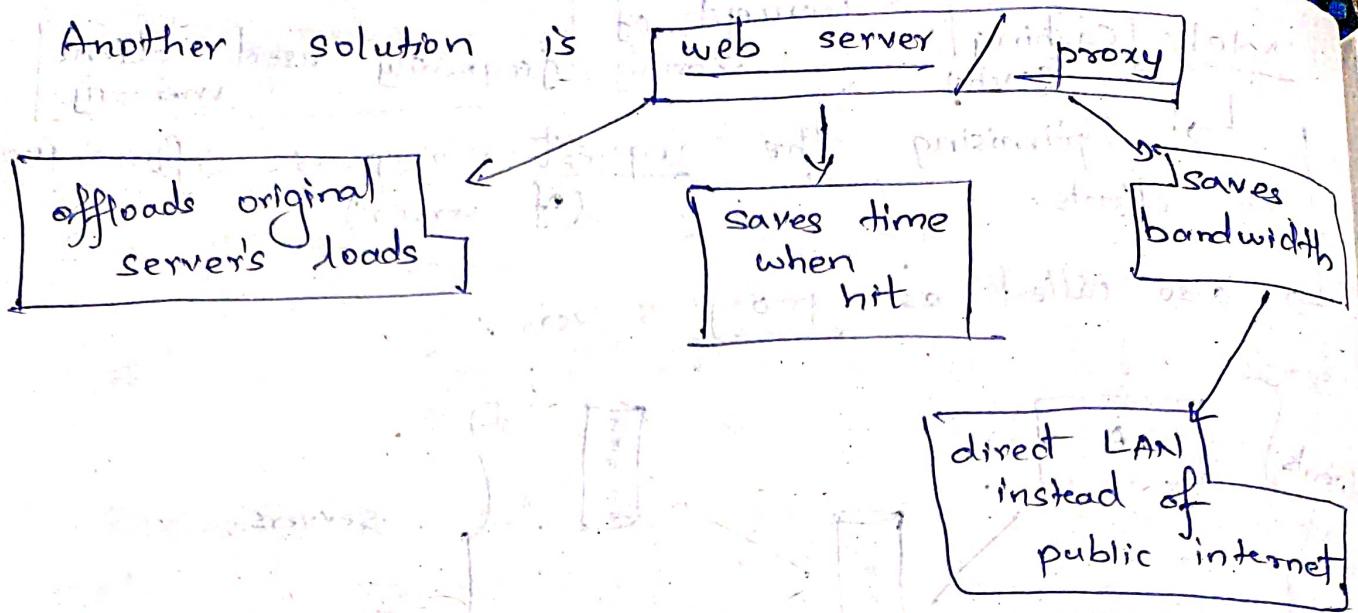
avg response time = 2 sec.

Traffic intensity = 1 (very huge delay)

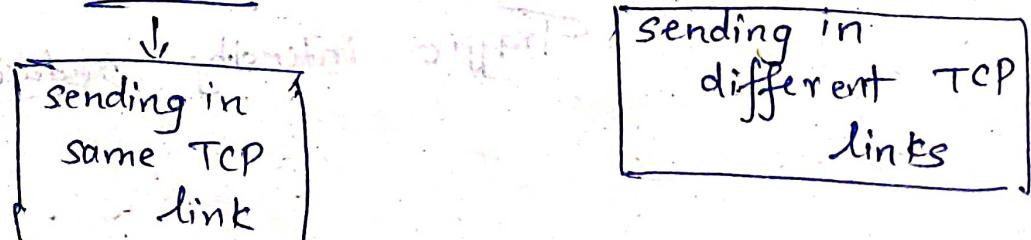
∴ Hence, proxy server is needed

Simple fix: Increase link. to 100 Mbps or so.

Traffic intensity reduces.



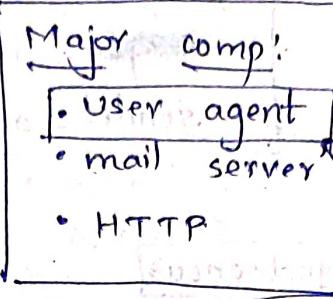
FTP is said to control information, out-of-band while HTTP is inband



Commands and Replies in slides \*  
(not important)

## Electronic mail (E-mail):

Asynchronous communication medium.



tool used to communicate.  
Ex: Outlook.

could be client

→ doesn't expect other person to be attentive.

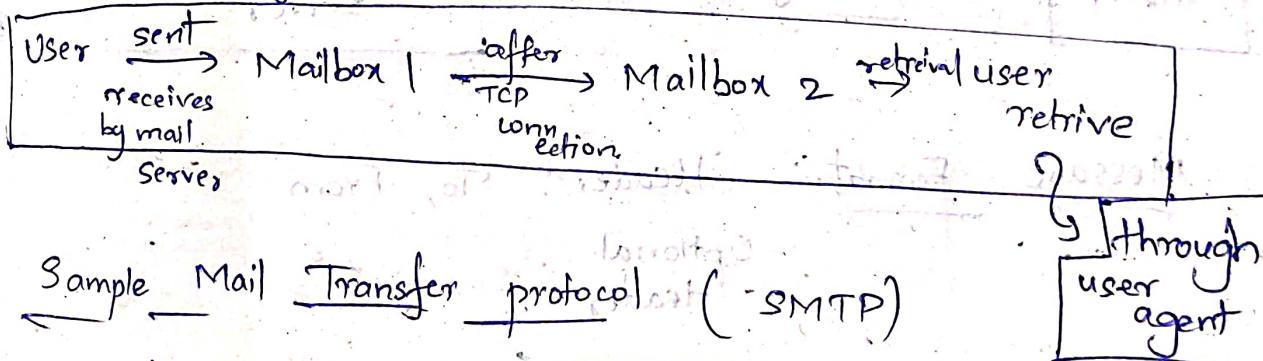
user uses this tool to push the message onto mail server

- Once handshake is completed between user agents, then the mail goes to the mailbox of other user

No information yet!

- When he's online, he receives from mail server 2.

Diagram in slides \*



## Sample Mail Transfer protocol (SMTP)

transfers messages from user's mail server to recipient's mail server

serves as both server and client

sender as client

receives as server

indicating client server architecture

Nice  
view - true

SMTP is asynchronous coz:

If receipt's mail down, client reattempts to send

If n iterations later, if not received, it notifies the sender and dispose the mail

2/2

SMTP

asynchronous

Sequence of

Operations:

End to End TCP connection

restricts the body to 7-bit ASCII

problems.

only text messages

Disadvantages:

Messages sometimes doesn't get placed in immediate mail server

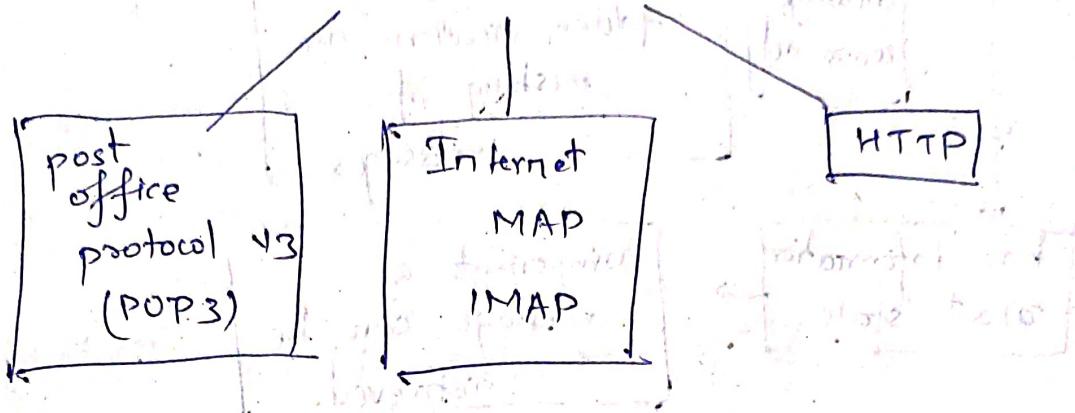
Message Format: Header: To, From

Optional Headers: subject

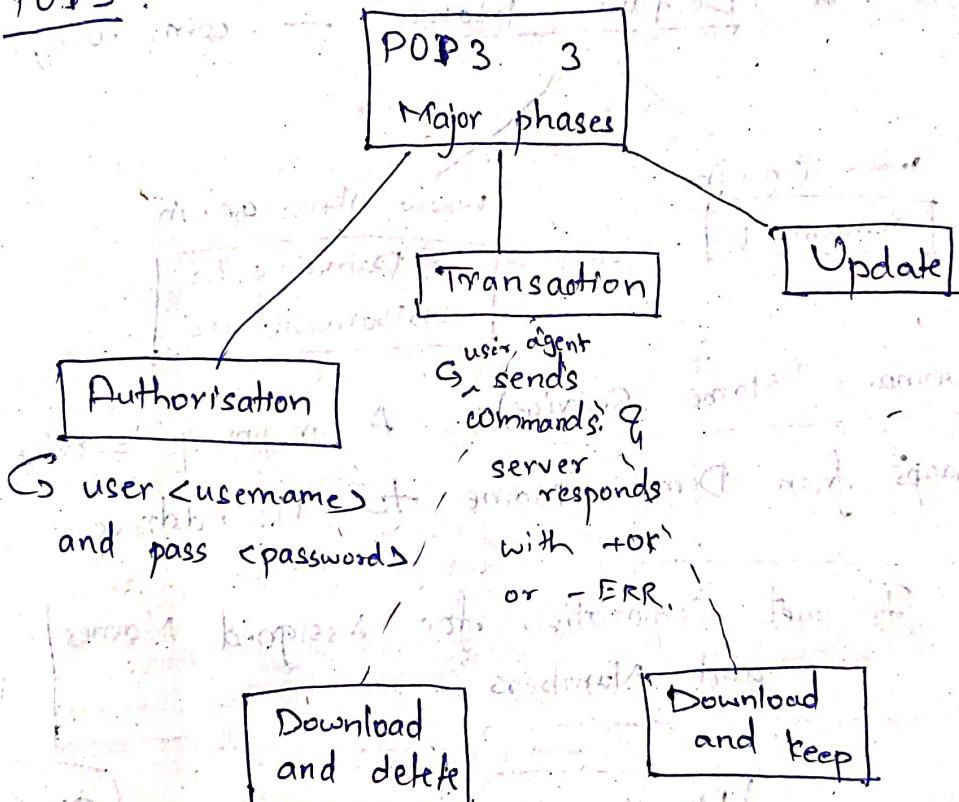
SMTP	HTTP
• push	• pull
• 7 bit ASCII rule	• No restriction.
• Only Text	• can encapsulate each object in own HTTP message

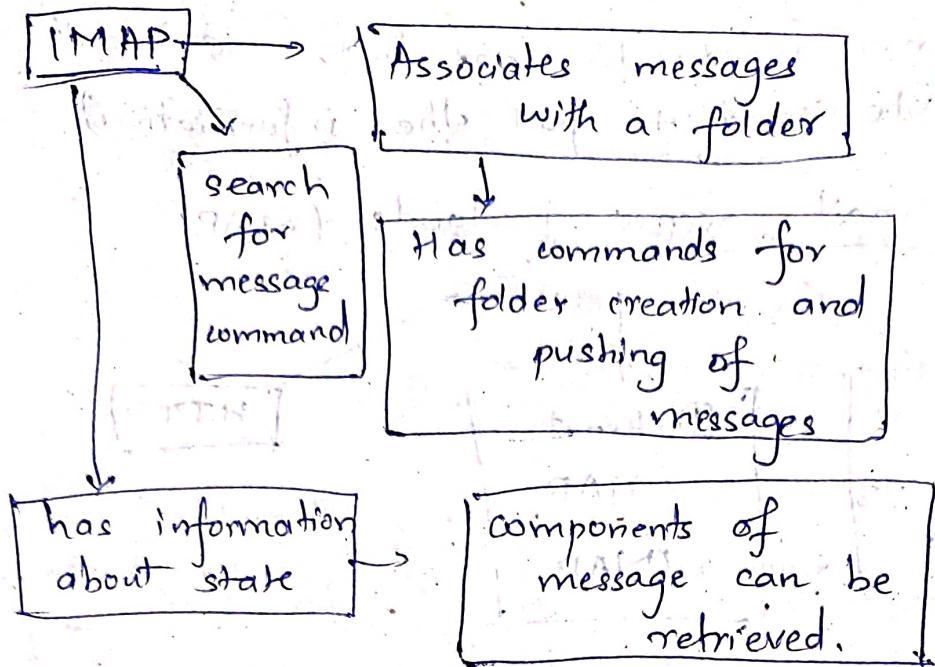
From B's mail server to mail box, we can't use push. (We need to pull the information)

Here comes mail access protocols (MAP)



### POP3:





What is a Domain Name? — .com, .org, .edu

www.itm.in

Domain: in

www.itsitm.ac.in

Domain: in  
subDomain: ac

DNS (Domain Name Service)

A mapping service which maps from Domain Name to IP Address.

ICANN: Internet Corporation for Assigned Names and Numbers

Domain Names

translation

Absolute

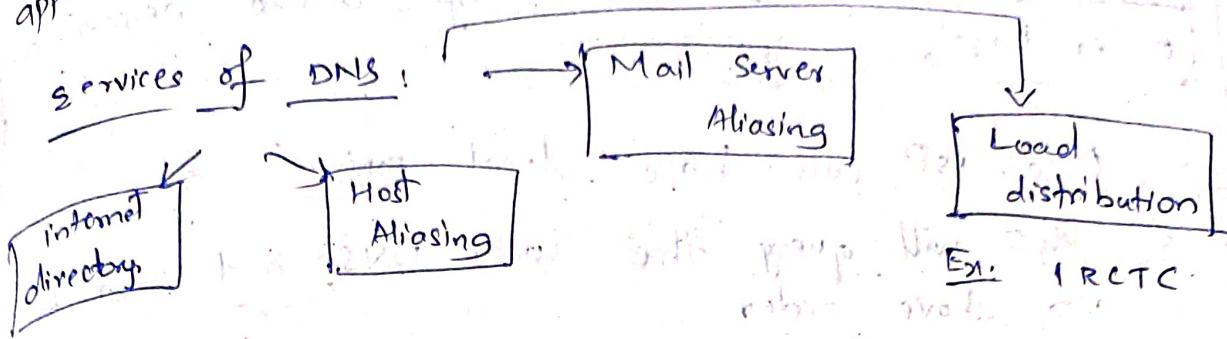
(Ends with)

Relative

(Interpreted based on context)

Application Layer protocol used by other applications

→ Client server architecture, uses UDP at transport layer  
 DNS adds an additional delay to the internet  
 applications that use it

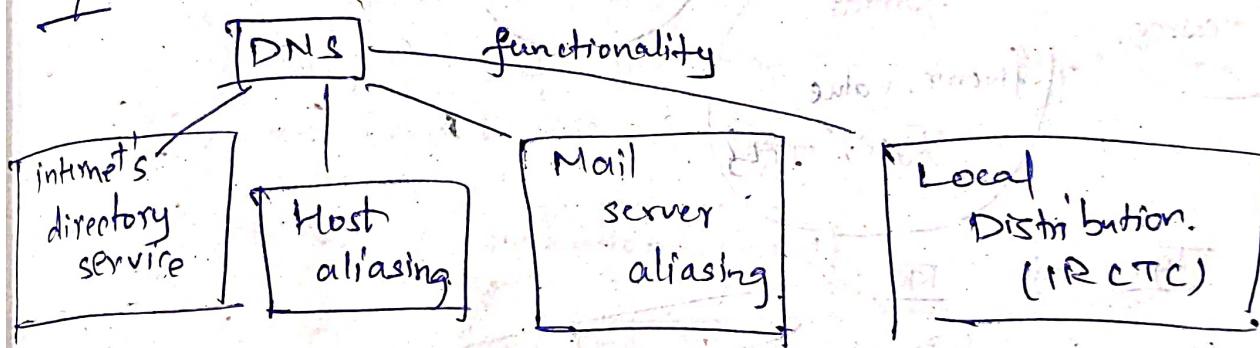


### problems:

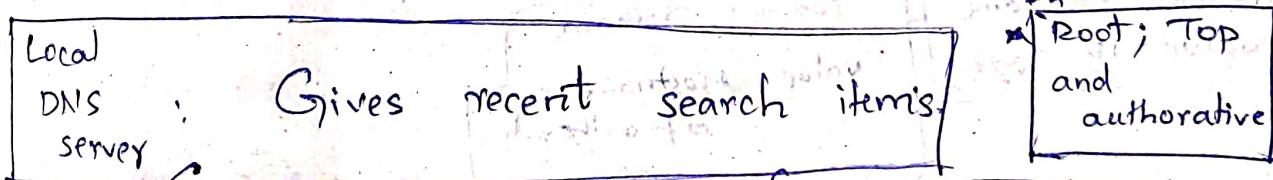
- A single point of failure
- Huge Traffic Volume
- Distant centralized database
- Maintenance

↳ huge delay

3/2:



DNS work by: Recursive and Iterative Query.



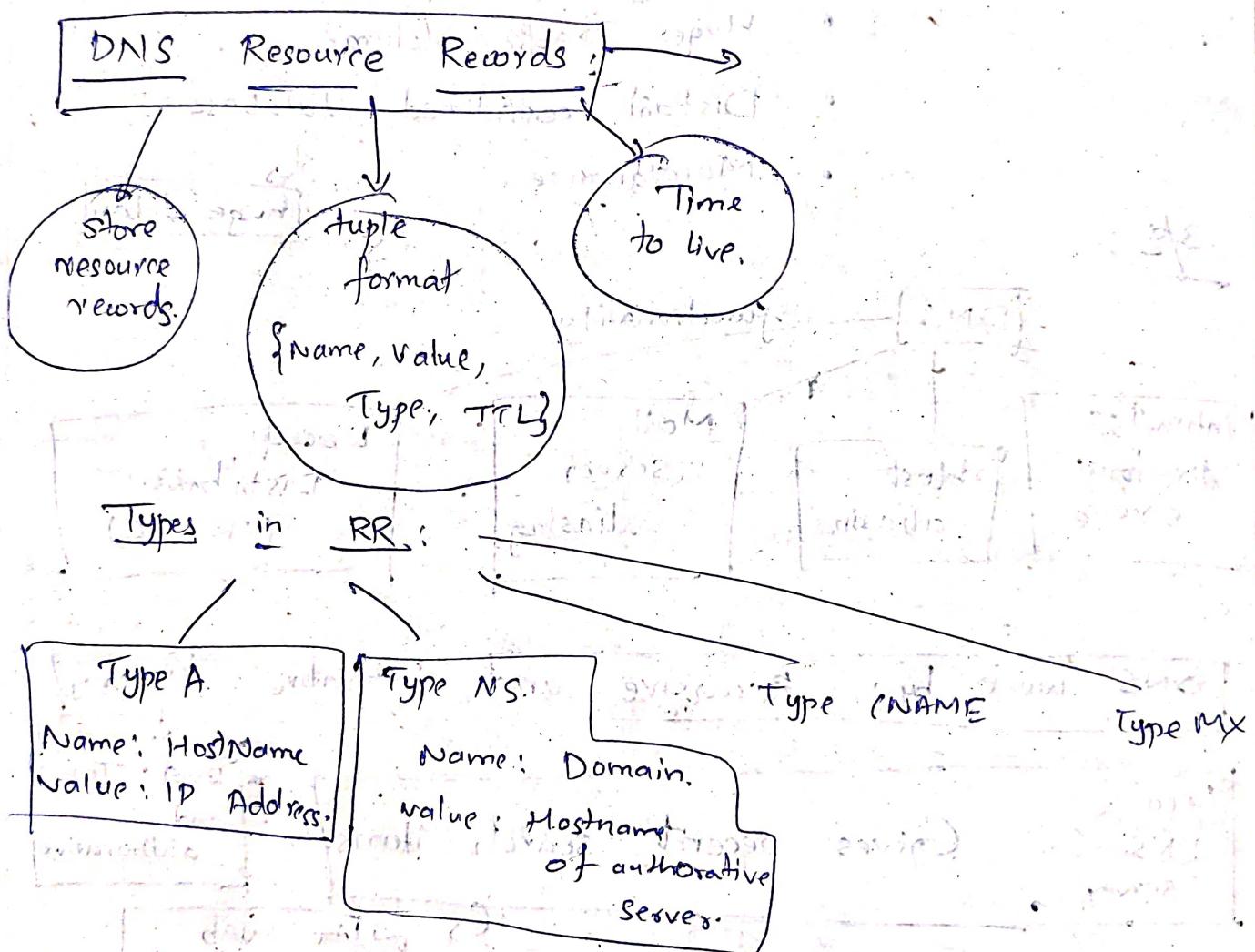
→ If located, we need not go to top level and root DNS and authoritative DNS.

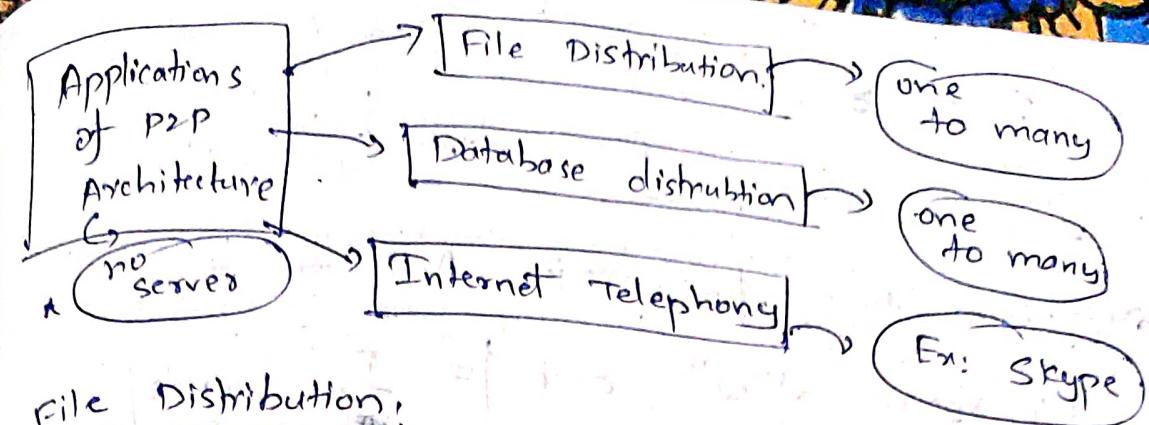
\* Diagram in slides A

Request to root, root gives IP Address to next hierarchy order.

LOCAL DNS have IP address of above hierarchical order DNS's.

- Local ISP can have local DNS
  - Host will query the local DNS and that takes to above order
  - Cache DNS replies.



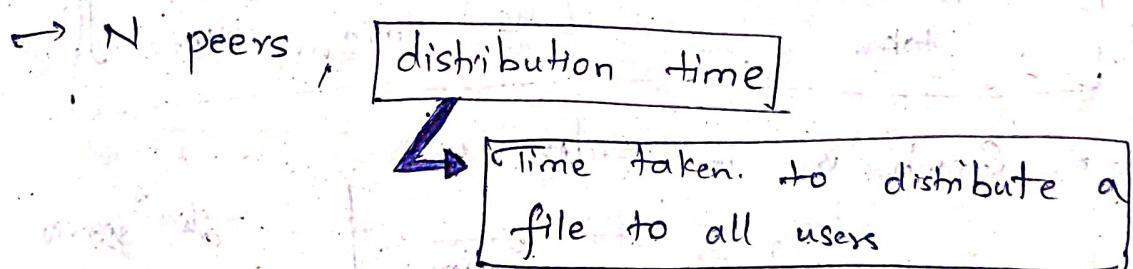


### File Distribution:

↳ Each peer can redistribute any portion of file to any other peer.

Ex: BitTorrent

↳ Scalability: Many peers can be there at one time.  
 ↳ In. Server client arch., one might not able to support 1M to 2M users, but in P2P arch., if the algorithm is right, one can support without limit



### Distribution time for client server architecture:

file size =  $F$  bits      users =  $N$

upload rate =  $u_s$  bits/sec

$$\text{Time taken} = \frac{NF}{u_s}$$

(Minimum)

If download limit is there, then  $d_{min} = \min\{d, d_2, \dots\}$

$$\therefore \text{Minimum distribution time} = \frac{F}{d_{min}}$$

$$\therefore D_{cs} \geq \max \left\{ \frac{NF}{d_{us}}, \frac{F}{d_{min}} \right\}$$

Distribution time for P2P Architecture:

File shared atleast once

$$= \frac{F}{u_s} \text{ seconds.}$$

peer with atleast download rate takes time

$$t < \frac{F}{d_{min}} \text{ seconds.}$$

Total upload rate =  $u_1 + u_2 + \dots + u_N$ .

System delivers  $F$  bits to each  $N$ . Min dist time

$$t = \frac{NF}{u_{total}}$$

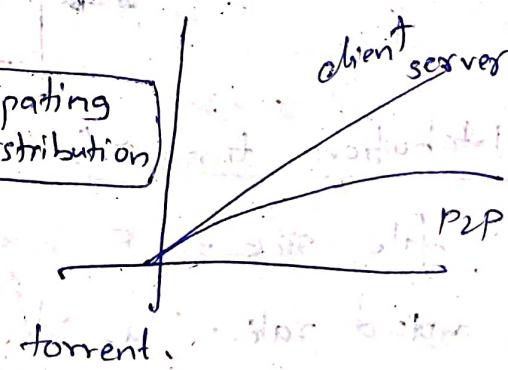
$$\therefore \text{Min time taken} = \max \left\{ \frac{F}{u_s}, \frac{F}{d_{min}}, \frac{NF}{u_{total}} \right\}$$

\* Graph is in slides

Bit Torrent:

nodes participating in distribution

Collection of peers in distribution of file called



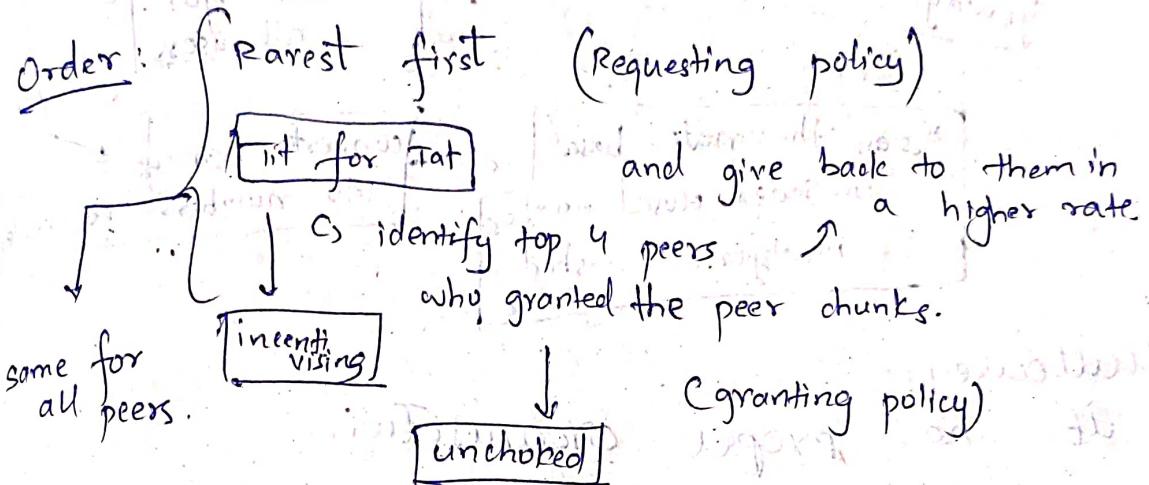
- peers in torrent download equals chunks of file (typically 256 kbps)
- peers accumulates more and more chunks over time
- once completed, they may leave or continue
- peers may leave with subset of files.

Each torrent has a node called tracker

When peers join the torrent, registers with tracker.

Each peer periodically updates tracker about its presence.

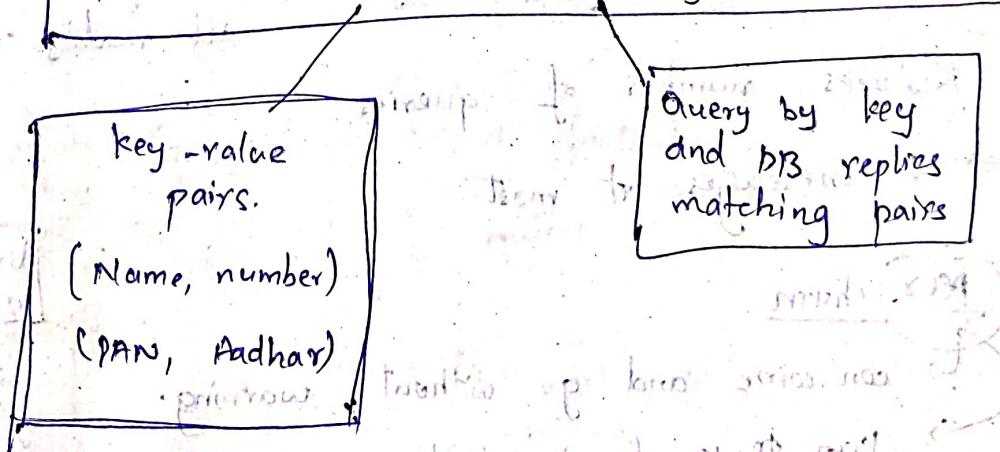
A new peer requests for chunks that does not current there with her.



New peer: optimistically unchoked

## Distributed Hash Tables

↳ Huge Database to stored among number of peers in a distributed way.



How Transfer occurs?

Distribution

Theore on both sides

- Assign **identifier** to each node  
↳ integer in  $[0, 2^n - 1]$  for some fixed  $n$ ,  
by integers
- (key, value) pairs are also identified using hash function
- always successor**
- available all for users**
- Assign the node based on index closest number  
identifier = hashed value
- converts string to numbers i.e. index

challenge:

= if no proper organization.

Circular DHT

Each peer only aware of predecessor and successor.

Dependency greatly reduced

instead of remembering all nodes.

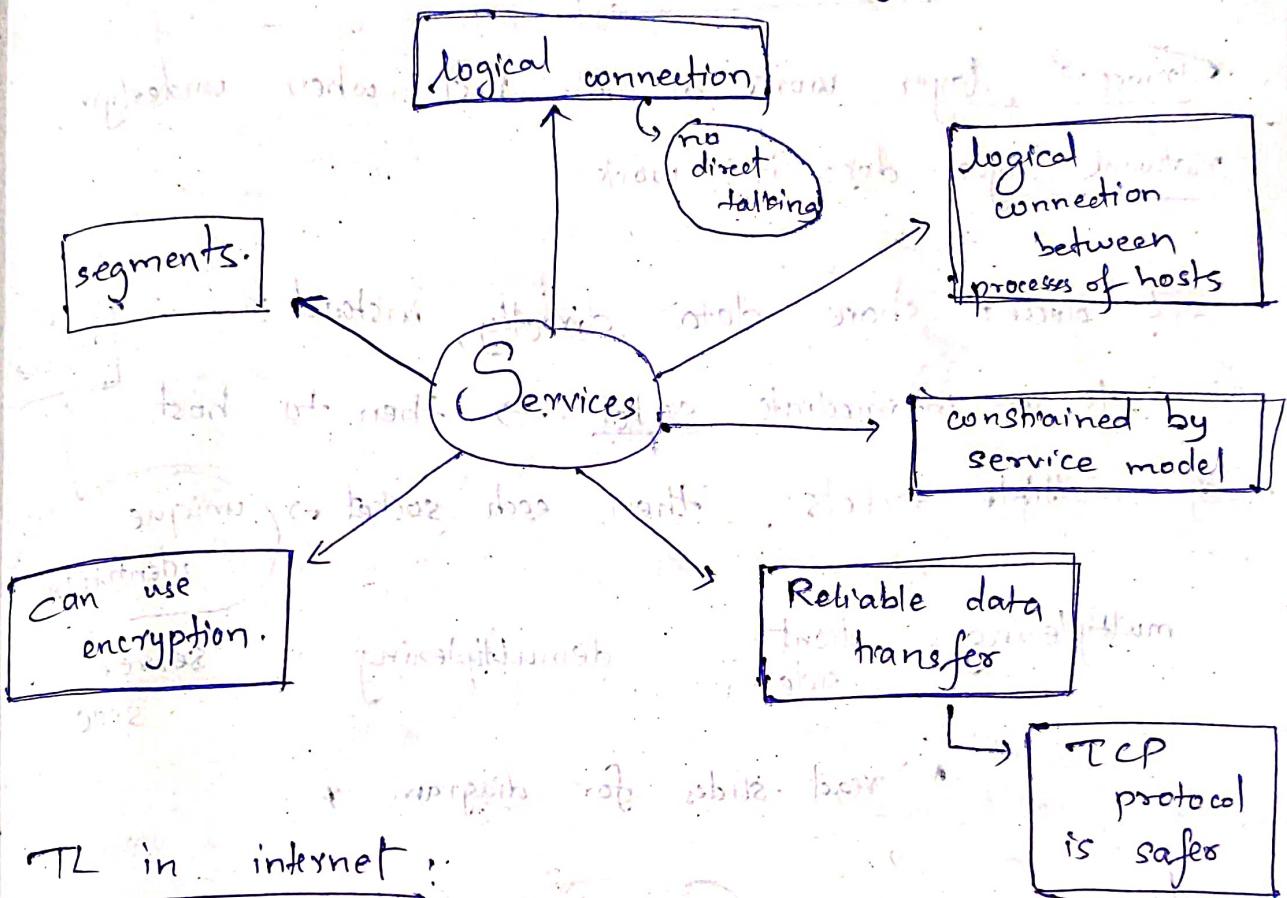
Reduces number of queries.

→  $N$  messages, at most

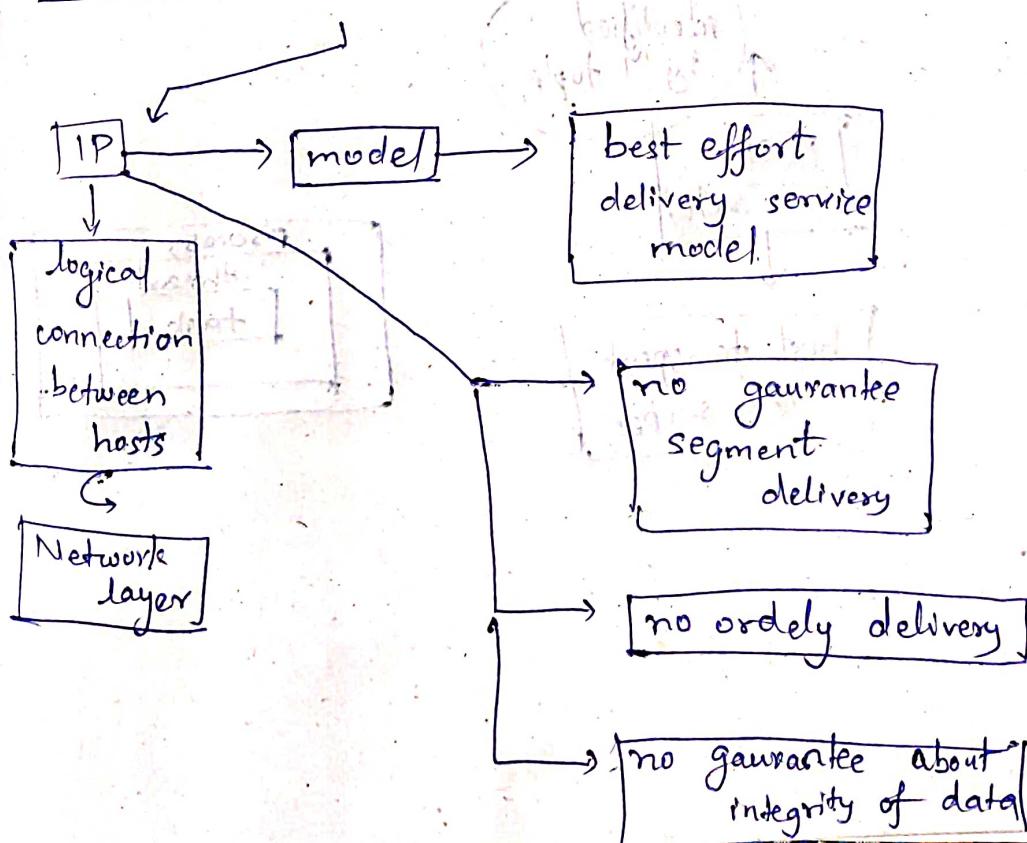
- peer churn:
- can come and go without warning.
- keep track of immediate predecessor and successor
- When neighbour leaves, peer updates its neighbour.
- live messages

7/2 TRANSPORT LAYER

→ Reliable communication method, (without corruption or data loss) controlling the transmission rate of transport layer entities in order to avoid or recover from, congestion within the network, (flow rate / congestion control)



### TL in internet:



UDP

→ process data delivery and error checking  
to process

TCP

reliable data transfer

correct and in order

congestion control

Transport layer works at its best when underlying network layer does its work

TL doesn't share data directly, instead:

→ to an intermediate socket → then to host

If multiple sockets, then each socket →

multiplexing : client side

demultiplexing :

server side

\* read slides for diagram \*

8/1

Identified by 4 tuple.

TCP  
socket

host transport  
the socket

process  
thread  
task

Connection less

no handshaking

UDP:

used in real time applications

Finer application service.

DNS

stateless

no connection establishment and no connection state

buffers

congestion control parameters

multiplexing, error checking.

Small packet header overhead

Example

Tcp : 20 bytes

UDP : 8 bytes

no sequencing

principles of Reliable Data Transfer:

Reliable

Data

Transfer

Safe

RDT 1.0 :

Assumptions

No packet loss

No bit errors