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Comparison between OSI model & TCP/IP Architecture.

ISO OSI model	TCP/IP Architecture / Internet Model
① There are 7 layers	① There are 4 layers → in original Model
② OSI is a model & not a protocol	Now there are 5 layers
③ Was developed after TCP/IP	② TCP/IP is a protocol not a model
④ In OSI model was devised first & then the protocols were invented.	③ was developed before OSI model
⑤ Clearly distinguishes between Service, interface & protocol	④ In TCP/IP the protocol was developed first & then the architecture was developed
⑥ Do not support Internetworking	⑤ Does not clearly distinguish between service, interface & protocols
⑦ Supports both connectionless & connection oriented communication in network layer.	⑥ Supports Internetworking
⑧ OSI model is strictly layered	⑦ Supports both Connectionless & connection oriented communication in transport layer but also supports only connectionless communication in network layer
⑨ Vertical approach (Transport layer of one can directly talk to Transport layer of other)	⑧ TCP/IP Architecture is loosely layered
	⑨ Horizontal approach (Each layer is stored)

Network layer delivery

- ① Best effort delivery
(source to destination delivery)
- ② Does not guarantee transmission
- ③ Not reliable, does not provide error & flow control
- ④ Connectionless
- ⑤ packets or datagrams.
- ⑥ uses IP address

Transport layer delivery

- ① End to End delivery
(process to process delivery)
- ② Guaranteed transmission
- ③ reliable, provides error & flow control
- ④ connectionless or connection oriented
- ⑤ TCP segments
UDP → datagrams
SCTP → packets
- ⑥ uses port address.

Data Link Layer Delivery

- 1} Reliable
- 2} frames are used for information transfer
- 3} uses MAC address /phy link /HW address
eg AB : C1 : 23 : 14 : 36 : F2
- 4} Node to Node delivery (Hop - by - hop)

Service Point Addressing (Port Addresses)	Logical Addressing	Physical Addressing
1) This is the port no of the application	This is the IP address of the machine	This is the MAC address of the machine.
2) It distinguishes applications on the m/c's (e.g. HTTP & SMTP)	It distinguishes two different machines connected in Internet also called IP addressing	It distinguishes two different m/c's connected in LAN also called Hardware or Link or MAC addressing.
3) Also called port addressing	Not fixed	fixed unless MAC is changed.
4) Not fixed always	done at Network layer	Done at data link layer.
5) Done at transport layer	IP uses logical addressing	Ethernet, FDDI uses physical addressing.
6) TCP & UDP are used for port addressing Max port addr is 65535 Combination of IP & port addr is called socket addr.	32-bit addr \rightarrow IPv4 128-bit addr - IPv6 2^{32} addresses are possible 2) dotted decimal format e.g. 192.3.2.1/24	48 bit 2^{48} addresses are possible. Hex colon format e.g. AB:CD:11:22:FE:56
7) 16-bit addr.		
8) 2^{16} addresses are possible		
9) decimal format e.g. 753		

message

	Unit of communication (or data unit)	Way of communication	Addressess
Application	message or ALPDU	End-to-End	Application Specific address (socket Address)
Transport	Segments → TCP Datagram → UDP Packet → SCTP	end-to-end or process to process	Port Addresses
Network	Packets / Datagrams	source to destination (end-to-end)	IP Address or logical Address
Data link	frames	node to node	Physical Address or MAC Address or Hardware Address
Physical	bits	node to node	

related to IP protocol
network layer protocol

fig: TCP/IP stack

ALPDU: Application Layer
protocol Data Unit

TCP: Transmission Control
Protocol

UDP: User Datagram protocol

SCTP: Stream Control Transmission
Protocol.

Comparison of Network Topologies

Topology	Total No. of Links	Privacy	Installation	Cost	Fault Identification and Isolation	Line Configuration	Application
Mesh	$\frac{n(n-1)}{2}$	Yes	Difficult	Expensive	Easy	P2P	Regional Telephone Offices
Star	$n - 1$	No	Easy	Less Expensive	Easy	LANS, High Speed LANs	
Bus	n	One	Difficult	Expensive	Easy	Multicast	
Ring	n	No	Difficult	Least Expensive	Difficult	Not used in large networks	

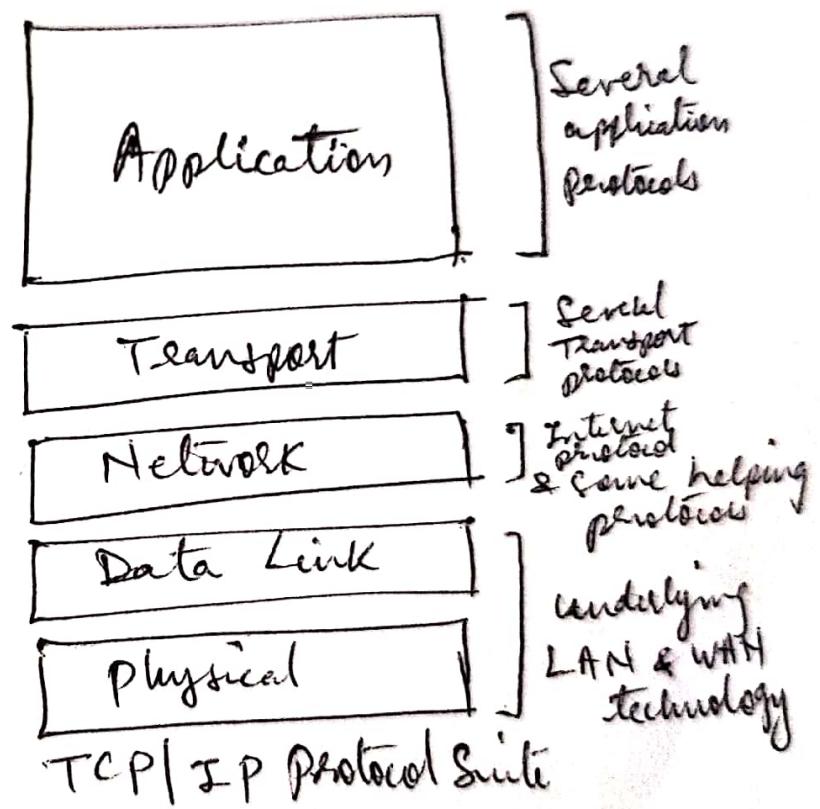
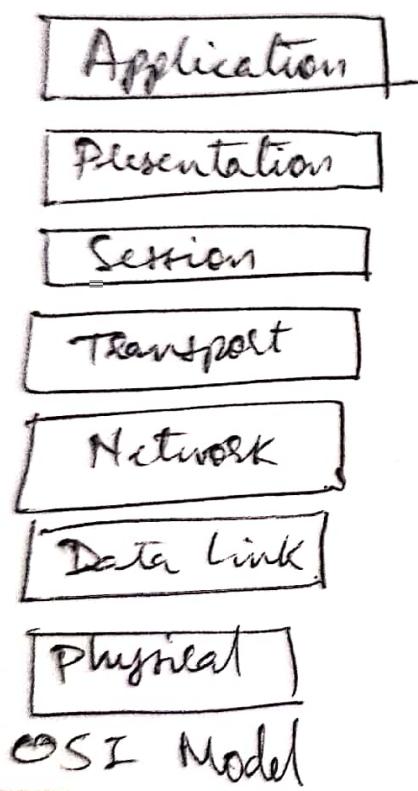
Single backbone with n drop lines

$n = \# \text{ of nodes}$

$P_2 P$ = Point to point

IBM token ring. Not required in token LANs

Comparison Between OSI & TCP/IP protocol Suite



Comparison between TCP & UDP

Transmission control protocol (TCP)	User Datagram protocol (UDP)
① provides process to process service.	① provides process to process service.
② TCP is connection oriented protocol	② UDP is connectionless protocol.
③ The unit of communication is segment	③ The unit of communication is user datagram.
④ provides reliable service	④ provides unreliable service
⑤ Implements Error control using checksums, acknowledgement and time outs.	⑤ Implements minimal level of error control using checksum.
⑥ The use of checksum in TCP ^{segment} datagram is optional. Mandatory	⑥ The use of checksum in UDP datagram is optional.
⑦ TCP Header is 20 Bytes without options. TCP header can extend upto 60 Bytes with options	⑦ UDP Header is 8 Bytes
⑧ Since Header occupies more bytes, TCP incurs more overhead	⑧ Less overhead in UDP as header is 8 bytes.
⑨ TCP provides flow Control	⑨ UDP does not provide flow control
⑩ TCP provides Congestion control	⑩ UDP does not provide congestion control
⑪ TCP is a byte stream service (Data is transmitted as stream of Bytes)	⑪ UDP is a message service (Data is transmitted as messages)

TCP

- (12) TCP provides more ~~options~~ services with the use of options
- (13) TCP provides full duplex communication
- (14) TCP uses pseudoheader for checksum calculation
- (15) value of protocol field for TCP in pseudoheader is 6
- (16) TCP offers slow delivery due to ACK mechanism. Hence not suitable for audio & video transmission.
- (17) TCP segments are encapsulated in IP packets.
- (18) Well Known ports used by TCP:

<u>Port</u>	<u>Protocol</u>
20 & 21	FTP
23	TELNET
25	SMTP
53	DNS
67	BOOTP
79	Finger
80	HTTP

UDP

- (12) UDP provides less services as there are no options.
 - (13) UDP does not provide full duplex communication
 - (14) UDP uses pseudoheader for checksum calculation.
 - (15) Value of protocol field for UDP in pseudoheader is 17
 - (16) UDP offers fast delivery hence suitable for audio & video transmission.
 - (17) UDP Segments are encapsulated in IP packets.
 - (18) Well known ports used by UDP:
- | <u>Port</u> | <u>Protocol</u> |
|-------------|-----------------------------|
| 53 | DNS |
| 67 | Bootps \Rightarrow DHCP-S |
| 68 | Bootpc \Rightarrow DHCP-C |
| 69 | TFTP |
| 161 | SNMP |
| 162 | SNMP (Trap) |

TCP

- TCP is not suitable for interactive traffic
- TCP provides sequence numbers
- TCP ~~does not provide~~ Timestamping
- TCP does not support multicasting
- TCP supports port numbers
- TCP does not support mixing
- Retransmission in TCP upsets timestamping & playback in real time traffic

UDP

- UDP is more suitable for interactive traffic.
- UDP does not provide sequence numbers.
- UDP does not provide timestamping
- UDP supports multicasting
- UDP supports port numbers
- UDP does not support mixing
- No retransmission in UDP

RTP

- RTP is most suitable for interactive traffic.
- RTP provides sequence numbers.
- RTP provides timestamping
- RTP does not support multicasting.
- RTP does not support port numbers
- RTP supports mixing.
- No retransmission in RTP

* marked features of UDP & RTP are combined to handle real time traffic on Internet.

Link State Routing Versus Distance.

Vector Routing :

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Distance Vector Routing

- Neighbouring routers exchange the vectors stating the distances to the desired destination.
- Metric is number of hops
- Less convergence speed
- use Bellman Ford algorithm to compute shortest path
- Neighboring nodes adapt eventually to changes in n/w topology
- eg. RIPv

Link State Routing

- Each node floods the information about the state of the links that connects to its neighbours.
- metric is cost.
- High Convergence Speed.
- use Dijkstras algorithm to compute shortest path.
- each node floods the information regarding changes in n/w topology
eg: OSPF

Stop-and-wait ARQ

- Transmits one frame at a time
- If a frame is in error then it is retransmitted
- Sequence number is one bit both for I-frame & ack frame.
- No of Sequence numbers = $2^1 = 2$ (i.e 0,1)

Go-Back-N ARQ

- Transmits number of frames equal to the window size W_s
- If a frame is in error then ~~than~~ all frames are retransmitted
- Sequence numbers ~~are~~ ^m bits for I-frame & ~~one~~ ^m bit for ack frame.
- No of Sequence numbers = 2^m
if $m=3$ then No of Sequence numbers = $2^3 = 8$
i.e. 0,1,2,3; 4,5,6,7.

	Stop-and-wait ARQ	Go-Back-N ARQ	Selective Repeat ARQ
Transmission Efficiency	$\eta_{SW} = \eta_0 \cdot (1 - P_f)$ $= \frac{(1 - \frac{n_0}{n_f}) \cdot (1 - P_f)}{1 + \frac{n_0}{n_f} + 2 \frac{(t_{prop} + t_{rec})R}{n_f}}$ <u>Neglecting header & CRC overhead</u> $\eta_{SW} = \frac{(1 - P_f)}{1 + 2 \frac{(t_{prop} + t_{rec})R}{n_f}}$	$\eta_{GBN} = \frac{1 - \frac{n_0}{n_f} (1 - P_f)}{1 + (W_s - 1)P_f}$ <u>Neglecting header & CRC overhead</u> $\eta_{GBN} = \frac{(1 - P_f)}{1 + (W_s - 1)P_f}$	$\eta_{SR} = (1 - \frac{n_0}{n_f})(1 - P_f)$ <u>Neglecting header & CRC overhead</u> $\eta_{SR} = (1 - P_f)$
	$L = 2 \frac{(t_{prop} + t_{rec})R}{n_f}$ = size of the pipe = multiple of frames = delay BW product	In Go-Back-N $W_s = L + 1$	
	Then, $\eta_{SW} = \frac{(1 - P_f)}{1 + L}$	$\eta_{GBN} = \frac{1 - P_f}{1 + (L + 1 - 1)P_f}$ $\eta_{GBN} = \frac{1 - P_f}{1 + LP_f}$	$\eta_{SR} = 1 - P_f$

- Comments:
- 1) Go-Back-N ARQ is worse than Selective repeat ARQ by a factor of $(1 + LP_f)$ & Stop-and-wait is worse than Selective repeat by a factor of $(1 + L)$
 - 2) If $LP_f \ll 1$ then $\eta_{GBN} = \eta_{SR}$
 - 3) If P_f approaches 1 then LP_f becomes L & $\eta_{GBN} = \eta_{SW}$

Comparison of Stop-and-Wait, Go-Back-N and Selective repeat ARQ's.

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Stop-and-Wait ARQ	Go-Back-N ARQ	Selective Repeat ARQ.
1) only one frame is transmitted at a time.	Number of frames transmitted depend on the window size.	Number of frames transmitted depend on the window size.
2) one bit is reserved in the header of I-frame ^{ack frame} _{for sequence numbering} .	m-bits are reserved in the header of I-frame ^{ack frame} _{for sequence numbering} .	m-bits are reserved in the header of I-frame ^{ack frame} for sequence numbering.
3) Not an example of Sliding Window protocol.	<ul style="list-style-type: none"> Example of Sliding window protocol. maximum window size is $W_S = 2^m - 1$ Transmitter window size is W_S and receiver window size is only one frame. 	<ul style="list-style-type: none"> Example of Sliding Window protocol. maximum window size is $W_S = 2^{m-1}$ Transmitter window size is W_S and receiver window size is W_R & they are equal. $W_S = W_R$
4) whenever a frame is in error it is retransmitted.	whenever a frame is in error, that frame & all subsequent frames are retransmitted	whenever a frame is in error only that frame is retransmitted.
	$W_S > \text{Delay } \times \text{BW product}$	$W_S > \text{Delay } \times \text{BW product}$

Parameter	IPv4	IPv6
<u>① Address</u>	<ul style="list-style-type: none"> 1) 32-bit Address 2) Address in dotted decimal format. e.g. 192.168.2.1/24 3) IPv6 Address allows CIDR notation e.g. 64.2.1.0/8 which implies that there are 8-bits in Network ID part. 4) 4 octet or 4 Byte Address 5) Specifies binary prefix to classify addresses e.g. Binary Prefix <ul style="list-style-type: none"> 0 Class A Address 10 Class B Address 110 Class C Address 1110 Class D Address 1111 Class E Address 	<ul style="list-style-type: none"> 1) 128 bit address 2) Address in colon Hex format 8000:0000:0000:0000: 0123:4567:89AB:CDEF 3) IPv6 Address extends CIDR notation e.g. 12AB::CD30:0:0:0/60 which specifies the first 60 bits of the address or 12AB00000000CD3 in Hex 4) 16 octets or 16 Bytes Address (32 - Hexadecimal digits Address) 5) Specifies binary prefix to classify addresses e.g. Binary Prefix <ul style="list-style-type: none"> 001 Aggregatable Global Unicast 111111010 Link-Local Unicast Address [FF80::/64] Site-Local Unicast Address 111111011 Unicast Address 11111111 Multicast Address.
<u>Special Addresses</u>	6) Loopback Address 127.0.0.0 to 127.255.255.255	6) Loopback Address 0:0:0:0:0:0:0:1

7) IPv4 Address with all 0 bits in Host ID part is universal/N/W address

A IPv4 Address with all 1 bits in Host ID part is universal Broadcast Address

7) IPv6 Address with all 0's is an unspecified Address which cannot be assigned to any computer or used as a destination. It is only used as a source address during bootstrap by a computer that has not learned its address.

8) 2^{32} address Space

$$2^{32} =$$

9) Address space almost exhausted

10) Limited address space

8) 2^{128} Address Space

$$2^{128} =$$

9) Address space cannot be exhausted in the foreseeable future
10) Extended address space

Header

11) Does not support authentication

12) Does not support flow labeling

13) Supports authentication

14) Supports flow labeling

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- 13) IPv4 datagram
= IPv4 Header
+ ~~option~~ zero or
more options
+ Data
- 13) IPv6 datagram
= IPv6 Header
+ zero or more
extension headers
(options)
+ Data
- 14) 10-octets
IPv6-Header.
- 15) Does not support auto-configuration (requires DHCP)
- 16) Does not support renumbering.
- 17) IPv4 supports defines unicast & multicast addressing
- 17) IPv6 In addition to unicast & multicast addressing IPv6 supports defines Anycast address.

A single anycast address can be assigned to a set of computers ; a datagram sent to the address is delivered to exactly one computer in the set (i.e. the computer closest to the source)

IPV4

IPV6

- | | |
|--|---|
| 18) Does not guarantee delivery of packets | 18) Guaranteed Service |
| 19) TTL field
Type of service field
Total length field | 19) Flow label field |
| 20) Total length field | 20) payload length field |
| 21) Identification, flags & fragment offset field | 21) corresponding fields not present. |
| 22) Time to live field | 22) Hop limit |
| 23) protocol field | 23) No protocol field (Next header field) |
| 24) Header checksum field | 24) No Header checksum field |
| 25) IPV4 needs VLSM, CIDR | 25) IPV6 eliminates the need for VLSM, CIDR |
| 26) IPV4 uses Subnet Mask (n/w prefix) | 26) Subnet Mask is not used for IPV6. |
| 27) Poor QoS | 27) Better QoS. |
| 28) Less real time performance. | 28) Increased real time performance. |

Table 3

Key differences between IPv4 and
IPv6

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IPv4	IPv6
① Source and destination addresses are 32 bits (4 bytes) in length	① Source and destination addresses are 128 bits (16 bytes) in length
② IPsec support is optional	② IPsec support is required
③ No identification of packet flow for QoS handling by routers is present within the IPv4 header.	③ packet flow identification for QoS handling by routers is included in the IPv6 header using the flow label field.
④ fragmentation is done by both routers and sending host	④ fragmentation is not done by routers, only by the sending host.
⑤ Header includes a checksum	⑤ Header includes does not include a checksum.
⑥ Header includes options	⑥ All optional data is moved to IPv6 extension headers.

IPv4

- ⑦ Address resolution protocol (ARP) uses broadcast ARP request frames to resolve an IPv4 address to a link layer address.
- ⑧ Internet Group Management protocol (IGMP) is used to manage local subnet group membership.
- ⑨ ICMP router discovery is used to determine the IPv4 address of the best default gateway and is optional.
- ⑩ Broadcast addresses are used to send traffic to all nodes on a subnet.
- ⑪ Must be configured either manually or through DHCP.

IPv6

- ⑦ ARP request frames are replaced with multicast neighbour solicitation messages.
- ⑧ IGMP is replaced with multicast Listener Discovery (MLD) messages.
- ⑨ ICMP Router Discovery is replaced with ICMPv6 Router solicitation and Router advertisement messages and is needed (required).
- ⑩ There are no IPv6 broadcast addresses. Instead, a link-local scope all-nodes multicast address is used.
- ⑪ Does not require manual configuration or DHCP.

- (12) uses host address(A) resource records in the Domain Name System (DNS) to map host names to IPv4 addresses.
- (13) uses pointer(PTR) resource records in the IN-ADDR.ARPA DNS domain to map IPv4 addresses to host names.
- (14) must support a 576-byte packet size.
- (15) uses host address(AAAA) resource records in the Domain Name System (DNS) to map host names to IPv6 addresses.
- (16) uses pointer(PTR) resource records in the IP6.ARPA DNS domain to map IPv6 addresses to host names.
- (17) must support a 1280-byte packet size.

Table 2

IPv4 Addressing Concepts and Their IPv6 Equivalents

<u>IPv4 Address</u>	<u>IPv6 Address</u>
① Internet address classes	① Not applicable in IPv6.
② Multicast addresses (224.0.0.0/4)	② IPv6 multicast addresses (FF00::/8)
③ Broadcast addresses	③ Not applicable in IPv6.
④ Unspecified address is 0.0.0.0	④ unspecified address is ::
⑤ Loopback address is 127.0.0.1	⑤ Loopback address is ::1
⑥ public IP addresses	⑥ Global unicast addresses.
⑦ private IP addresses (10.0.0.0/8, 172.16.0.0/12, and 192.168.0.0/16)	⑦ site-local addresses (FFCO::/10)
⑧ Autogenerated addresses (169.254.0.0/16)	⑧ link-local addresses (FF80::/64)
⑨ Tent representation: Dotted decimal notation	⑨ Tent representation: Colon hexadecinal format with suppression of leading zeros and zeros completion. IPv4 - compatible addresses are expressed in dotted decimal notation.

- (10) New bits representation:
Subnet mask in dotted decimal notation or prefix length
- (10) New bits representation:
prefix length notation only.
- (11) DNS name resolution:
IPv4 host address(A)
resource record
- (11) DNS name resolution:
IPv6 host address
(AAAA) resource record.
- (12) DNS reverse resolution:
IN-ADDR.ARPA domain
- (12) DNS reverse resolution:
IP.ARPA domain.

Differences between the IPV4 and IPV6 header fields.

IPV4 Header field	IPV6 Header field
① Version	Same field but with different Version numbers.
② Internet Header Length	Removed in IPV6. IPV6 does not include a header length field because the IPV6 header is always a fixed size of 40 bytes. Each extension header is either a fixed size or indicates its own size.
③ Type of Service	Replaced by the IPV6 traffic class field
④ Total Length	Replaced by the IPV6 payload length field, which only indicates the size of the payload.
⑤ Identification fragmentation flags fragment offset	Removed in IPV6. fragmentation information is not included in the IPV6 header. It is contained in a fragment extension header.
⑥ Time to live	Replaced by the IPV6 Hop limit field
⑦ protocol	Replaced by the IPV6 next Header field
⑧ Header checksum	Removed in IPV6. In IPV6, bit level error detection for the entire IPV6 packet is performed by the link layer.
⑨ Source Address	The field is the same except that IPV6 addresses are 128 bits in length.
⑩ Destination Address	The field is the same except that IPV6 addresses are 128 bits in length
⑪ Options	Removed in IPV6. IPV4 options are replaced by IPV6 extension headers.

	ALOHA	Slotted ALOHA
Vulnerable period	$2X$ seconds	X seconds
Synchronous transmission	No	Yes
Maximum Throughput	18.4% i.e. $(1/2e)$	36.8% i.e. $(1/e)$
Throughput expression	$S = Ge^{-2G}$	$S = Ge^{-G}$
Average packet delay in Slotted ALOHA	$E[T_{ALOHA}]$ $= X + t_{prop} +$ $(e^{2G} - 1) [X + 2t_{prop} + B]$	$E[T_{Slotted ALOHA}]$ $= X + t_{prop} +$ $(e^G - 1) [X + 2t_{prop} + B]$

C) Congestion control

- 1) Congestion control makes sure that the subnet is able to carry the offered load traffic.
- 2) It is a global issue, involving the behaviour of all the hosts, all the routers, the store and forward processing within the routers, and all the other factors that tend to diminish the carrying capacity of the subnet.

Flow control

- 1) Flow control relates to the point-to-point traffic between a given sender and a given receiver.
- 2) Its job is to make sure that a fast sender cannot continually transmit data faster than the receiver is able to absorb it.

(b)

addresses

Circuit Switching

Dedicated transmission path
continuous transmission
of data

Messages not stored

path is established
for entire conversation

call setup delay, negligible
transmission delay

busy signal if called party
busy

Electromechanical or
computerised switching nodes

Fixed BW

No overhead bits after call
setup

No speed or code conversion

Virtual circuit
packet switching

No dedicated path
continuous transmis-
sion of packets

packets stored
until delivered
Route established
for entire
conversation.

call setup delay,
packet transmission
delay

Sender notified of
connection denial

small switching
nodes

Dynamic BW
overhead bits in
each packet

Speed or code conversion

Datagram
packet switching

No dedicated path

Transmission of
packets.
packets stored
until delivered.

Route established
for each packet

packet transmission
delay

Sender may be
notified if packet
not delivered.

Small switching nodes

Dynamic BW
overhead bits in a
packet

Speed or code conversion

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Congestion control

- It ensures that the new carries offered load
- It is a global issue
- It is performed by the routers
- It is done during set up or link establishment

Flow control

- It ensures that the Rx does not get overwhelmed i.e. the rate of processing at receiver & rate of sending processing at Tx remains balanced
- It is a local issue performed at every transmission
- It is usually performed by the sender
- It is done as & when needed.

Circuit Switching

Dedicated transmission path

Continuous transmission of Data

Fast enough for interactive

Messages are not stored

The path is established for entire conversation

Call Set up delay; negligible transmission delay.

Busy Signal if called party is busy

Overload may block call set up, no delay for established calls

Electromechanical or computerized switching

Datagram Packet Switching

No dedicated Path

Transmission of packets

Fast enough for interactive

packets may be stored until delivered

Route established for each packet

Packet transmission delay

Sender may be notified if packet not delivered.

Overload increases packet delay

Small switching nodes

Virtual Circuit packet Switching

No dedicated path (Virtual path)

Transmission of packets

Fast enough for interactive

packets stored until delivered

Route (logical or virtual) established for entire conversation.

Call Set up delay, packet transmission delay.

Sender may be notified if packet not delivered.

Overload may block call set up, increases packet delay.

Small switching nodes.

Circuit Switching

User responsible
for message loss
protection

Usually no speed
or code conversion

Fixed BW

No overhead
bits after
call set up

Datagram Packet Switching

Network may be
responsible for
individual
packets

Speed &
Code
conversion

Dynamic
use of BW

overhead bits
in each
packet

Virtual ckt packet switching

Network may
be responsible
for packet
sequences.

Speed & Code
conversion.

Dynamic use
of BW.

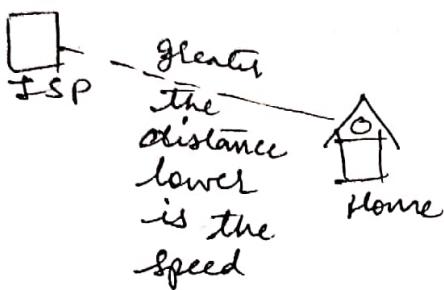
overhead bits
in each
packet.

Difference between DSL & VDSL

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Digital Subscriber Line (DSL)

- If distance between ISP & our home increases, then internet speed becomes slow



- This is because, in DSL the Internet connection is via the telephone lines which are made of copper (twisted pair) & these copper wires are much more susceptible to Electromagnetic interference. So as distance increases Signal becomes weak & speed becomes slow.

Very High Bit rate Digital Subscriber Line (VDSL)

- Works similar to cable internet.

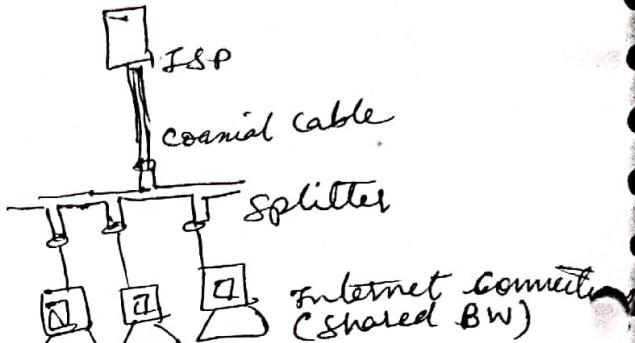


fig. Cable Internet.

VDSL Internet

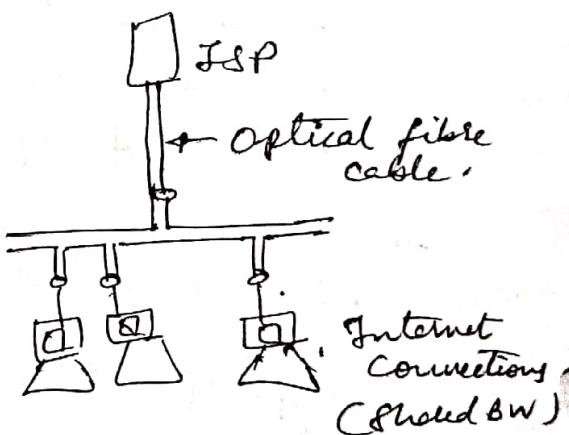


fig: VDSL Internet

- The connection between ISP & home computers is optical fibre cables.
- So even if distance increases, no losses (as no EMF) in OFC, the speed is maintained
- Download speed 50-52 Mbps
- Upload speed 15-16 Mbps
- freq: 12 MHz.

Dial-up Internet Connection

- Internet connection via Normal Telephone lines.
- Whenever Voice call & Data (Internet) connection over a single (same) telephone line.
- So whenever there was a voice call coming in during the Internet connection then that connection would be terminated
- Dial up is now replaced by DSL

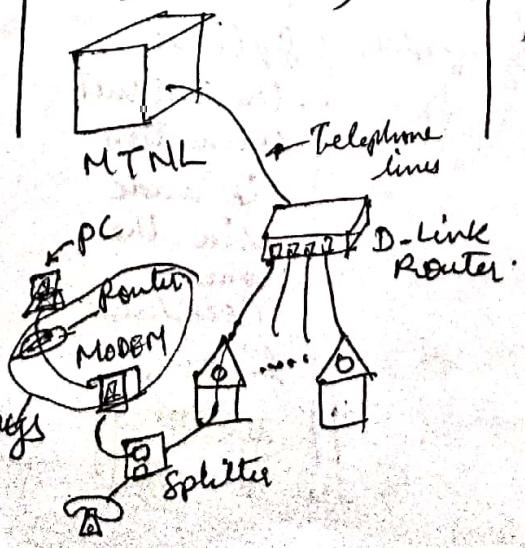
Converts Analog to Digital
nowadays Single device Modem

DSL Internet Connection

- Internet connection is via normal Telephone lines (twisted pairs)
- But 30% of the lines are used for voice connection & 70% of the lines are used for data (Internet) connection.

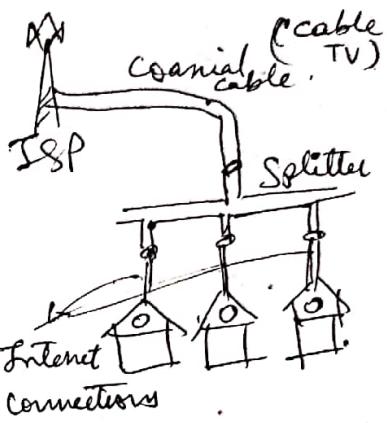


- Therefore Voice calls & Internet connections can be simultaneously done. (which was not possible with Dialup connection)



Cable Internet Connection

- Internet Connection via cable Television cables (coaxial cables)



- Shared Bandwidth
- Slower Than DSL

Comparison Between Dial up & DSL

Parameter	Dial up	DSL
Speed	Dial-up access offers speeds upto a maximum of 56 kbps	DSL offers guaranteed speeds (symmetrical up to <u>1M bits/s</u> or 35 times faster than 28.8 kbits/s analog modem) and assymetrical up to 7 M bits/s
Flexibility	Dial-up access is capable of providing internet access to only one PC (end-user), thereby charging extra for each additional (PC or enduser) access. Dial up access is not a scalable service due to its Bandwidth limitations	DSL provides Internet access to multiple PC's (end users) on a single connection, thereby not charging extra for all additional PC (end user). DSL is fully scalable service possessing a wide range of potential speeds.
Reliability	Has tedious process of dialing in for the Internet access	No time consuming dialing in for Internet access. Connection is dedicated, so no disruptions.

DSL Vs Cable Modem

Shubhangi K

Parameters	DSL	Cable Modem
Speed	<ul style="list-style-type: none">• DSL offers a wide range of guaranteed speeds as high as 1 Mbps (symmetrical)	<ul style="list-style-type: none">• Cable Modem exists on a shared network thereby making speed performance unpredictable; it is entirely contingent on network traffic volume.
Security	<ul style="list-style-type: none">• DSL is on a closed dedicated circuit making it less susceptible to outside hackers	<ul style="list-style-type: none">• Cable Modem is on a shared network making it more vulnerable to hackers.
Reliability	<ul style="list-style-type: none">• DSL is on a closed dedicated circuit thereby offering guaranteed speeds	<ul style="list-style-type: none">• Cable Modem exists on a shared network thereby making speed performance unpredictable.
Accessibility	<ul style="list-style-type: none">• DSL utilizes ubiquitous, 100-year old telephone infrastructure (RJ-11 jacks, copper phone wire, data backbones etc), which makes up nearly 100% market accessibility	<ul style="list-style-type: none">• Sporadic & inconsistent service availability.• Cable Modem has a slower rate of market infiltration because of its

	WiMAX	WiFi
Coverage	→ Best, Broader Area (upto 40 miles)	Small (approx 100 feet)
Spectrum	Licensed	Unlicensed
Mobility	Anywhere, Anytime	Limited to Hotspot Locations
Speed	Higher Speed (70Mbps or more)	Limited Speed (54Mbps)
Quality of Service	Support Multimedia Applications (Guaranteed QoS)	Does not support Multimedia applications, Unreliable. (QoS is not guaranteed)
Standard	Based on IEEE 802.16 Standard	Based on IEEE 802.11 Standard
Waves	Uses microwaves (unidirectional)	uses radio waves (bidirectional)
Launched in Year	2004	1997
Network	MAN	LAN
Channel Bandwidth	20 MHz, Variable, 1.25 - 20 MHz	20 MHz
Subscribers	Unlimited	1-10 approx
Radio Technology	OFDM	Direct access Spread Spectrum
Access protocol	CSMA/CA	Grant or Request.

Wi-MAX (MAN)

Technical Specifications

Specifications	802.16 d (fixed access)	802.16 e (mobile access)
1) BW (MHz)	1.75 to 20 MHz	1.25 to 20 MHz
2) Operating frequency	2-11 GHz	2-6 GHz
3) Mobility	Fixed / (Roaming)	Mobile
4) Tx Technology (access method)	Multicarrier OFDMA	OFDMA
5) Spectrum assignment	Subchannelization	Subchannelization
6) Mode	TDD, FDD	TDD, FDD
7) Peak rate	75 Mbps	5, 30 Mbps
8) Modulation	BPSK, 16 QAM	64 QAM
9) Enhanced Techniques	Intelligent Antenna Systems MIMO, HARQ.	MIMO, HARQ.
10) Energy Saving		Idle & sleep mode.

FDM

Definition

Type of signal

Signal

Circuity

Crosstalk

Fading

- 1) Signals are transmitted at different frequencies but at the same time.
(Multiplexing in frequency domain)
- 2) FDM is usually preferred for analog signals
- 3) Synchronization is not required
- 4) FDM requires a complex circuitry at Tx & Rx
- 5) FDM suffers from the problem of crosstalk due to imperfect BPF
- 6) Due to Bandwidth fading in the Tx medium, all the FDM channels are affected

TDM

- 1) Signals are transmitted at different times but at the same frequency

(Multiplexing in Time domain)

- 2) TDM is preferred for the digital signals
- 3) Synchronization is required
- 4) TDM requires simple circuitry
- 5) In TDM the problem of crosstalk is not severe
- 6) Due to fading only a few TDM channels will be affected.

Advantages & Disadvantages of Multiplexing Techniques

Multiplexing Technique	Advantages	Disadvantages
Frequency Division Multiplexing	<ul style="list-style-type: none"> Simple Popular with radio, TV, Cable TV All receivers such as cellular telephones, do not need to be at the same location 	<ul style="list-style-type: none"> Noise problems due to analog signals wastes Bandwidth Limited by frequency ranges.
Synchronous Time Division Multiplexing	<ul style="list-style-type: none"> Digital signals relatively simple commonly used with T-1 T-1, ISDN 	wastes Bandwidth
Asynchronous (Statistical) Time division Multiplexing	<ul style="list-style-type: none"> More efficient use of BW. frame can contain control & error information packets can be of varying size 	more complex than Synchronous Time division Multiplexing

* T-1 lines used in North America
E-1 lines are used in Europe.

Disadvantages

Multiplexing Technique

Advantages

Wavelength Division Multiplexing

- Very high capacities over fibre
- signals can have varying speeds
- scalable

cost complexity

Code Division Multiplexing

- Large Capacities
- Scalable

- Complexity
- primarily a wireless technology

* Difference Between Hub & a Switch

[Shubhangi Khatore]

	Hub	Switch
Layer in OSI model	Physical layer (Layer 1 Device)	Data Link Layer (Layer 2 Devices)
Transmission Type	only Broadcast	At Initial level Broadcast then unicast & multicast
Table	There is no MAC Table in Hub, Hub can't learn MAC address	Store MAC address in lookup table, Switch can learn MAC address
Usage	LAN	LAN
Ports	4 ports	24/48 ports
Collision	In Hub Collision occurs	In full duplex mode no collision occurs
Transmission Mode	Half Duplex	Full duplex
Collision domain	Hub has one collision domain	In switch, every port has its own collision domain
Cost	Cheaper than Switches	3-4 times costlier than Hub
Broadcast Domain	Hub has one Broadcast Domain	Switch has one Broadcast Domain

* Difference Between Switch & Router (networking devices)

Switch	Router
1) It is the LAN device	1) It is the WAN device
2) Switch maintains MAC tables	2) Router maintains IP tables (routing) tables
3) Switch understands MAC address (Hardware physical / link) address	3) Router understands IP address (logical address)
4) Switch connects devices in a LAN or Switch connects devices belonging to two similar LAN's together	4) Router connects more than two dissimilar networks together. Router connects LAN to a WAN. Router can connect Ethernet LAN with token ring LAN (dissimilar nw's)
5) Switch is an Intranetworking device (LAN)	5) Router is an Internetworking device (connects diff nw's in WAN or Internet)
6) Switch is Layer 2 networking device (used at Layer 2 of OSI model)	6) Router is Layer 3 networking device (used at Layer 3 of OSI model)
7) In Full Duplex, no collision occurs	7) No collisions
8) Full Duplex Transmission	8) Full Duplex transmission
9) Speed: 1-10 Mbps (full duplex) 100 Mbps (switch)	Speed 10/100 Mbps 1 Gbps

	<u>Switch</u>	<u>Router</u>
9)	Speed 10/100 Mb/s 1 Gb/s	Speed 1-10 Mb/s (wireless) 100 Mb/s (wired)
10)	Switch has one broadcast domain	Every port has its own Broadcast domain
11)	Takes more time for complicated switching decision	Takes faster routing decision

Shubhangi Kharche

Repeater	Hub	Switch	Bridge	Router
1) Layer 1 (Physical) of OSI Model	Layer 1 (Physical) of OSI model	Layer 2 (Data link) of OSI model	Layer 2 (Data link) of OSI model	Layer 3 (Network layer of OSI model)
2) Regenerates the signal (Extends LAN segments)	connects computers in LAN or connects two similar LAN segments Together (Ethernet to Ethernet) or (Token Ring to Token Ring)	connects two dissimilar LAN's Together to Token Ring or (Ethernet to FDDI)	connects two dissimilar LAN's Together (Ethernet to FDDI LANs to WAN)	
3) Does not have filtering capability	10BaseT to 10BaseF or 10BaseT to 10BaseF	filters (Has filtering capability)	filters (Has filtering capability)	
4) Dumb device	Dumb device	Intelligent device	Intelligent device	Intelligent device

Repeater	Hub	Switch	Bridge	Router
It has Two ports	usually has 4 ports 5th port connects to another hub	usually has even number of ports 4, 8, 12, 16, 24 maximum 24 ports	less no of ports than switch (2-3 ports)	minimum two interfaces (maximum can be upto 100)
Shubhamji Kharche				
⑥ No memory	No memory	Memory to save MAC Tables	Router have Five ports 1) console port (connects to PC) 2) auxiliary port (connects to internet) 3) LAN port (RJ-45). (connects to Ethernet or Fast Ethernet or Gigabit Ethernet)	
		Memory to save MAC Tables Also has processing capability	4) RS232 Interface, RJ-11 \Rightarrow Telephone line (Serial connects \Rightarrow Router)	
			5) USB port .	
		RAM	RAM	
		ROM	ROM	
		Flash	Flash	
		NVRAM	NVRAM	
			-Set-up configuration	
			Running OS	
			OS	
			System files	
			• Bootloader • etc.	
			• Running configuration	
			• file • Routing Tables • ARP tables • Packet buffer	

Repeater	Hub	Switch	Bridge	Router
	<ul style="list-style-type: none"> • Single collision domain • All port (all mac's) lie in single collision domain 	<ul style="list-style-type: none"> • Separate collision domain • (No collision) • (Each port of a switch lies in separate collision domain) 	<ul style="list-style-type: none"> • Separate Broadcasting & collision domains 	
	<ul style="list-style-type: none"> • Implementation is always broadcasted • Later on unicasting or multicasting 		<ul style="list-style-type: none"> • Broadcasting when some initialised 	
			<p>Chaitanya Khurana</p> <ul style="list-style-type: none"> • Removes congestion • Does Traffic Management 	

Physical Layer of Standard Ethernet

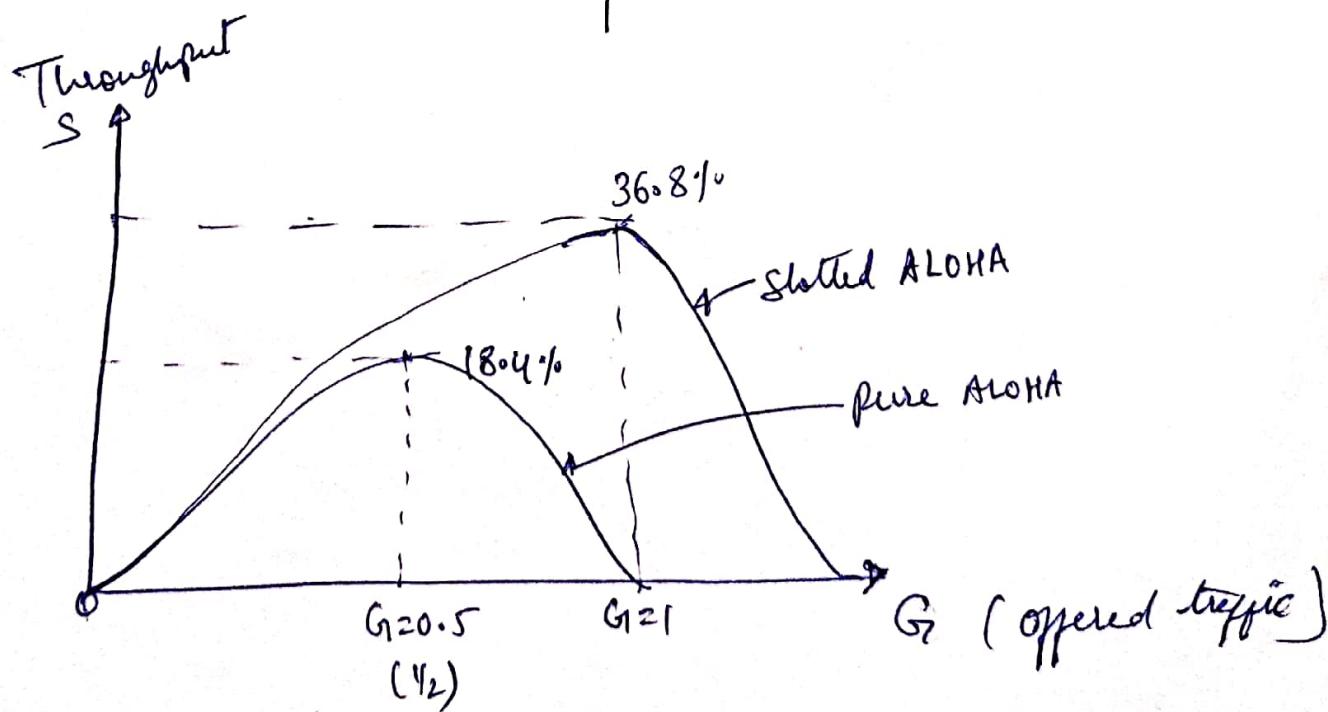
Standard Ethernet Common Implementations				
10 Base 5	10 Base 2	10 Base-T	10 Base-F	
10 Mbps Baseband signalling	10 Mbps Baseband	10 Mbps Twisted baseband pair	10 Mbps Baseband	
500m	200m			
Bus	Bus	Star	Star	
Thick coaxial	Thin coaxial	UTP (2 pairs)	Fiber (in number)	
500 m	185 m	100 m	2000 m	
Manchester	Manchester	Manchester	Manchester	
Thick Ethernet or Thicknet	Thin Ethernet or cheapernet	Twisted pair	Fiber	
		Ethernet	Ethernet	

Pure ALOHA

- 1) No Synchronised Transmission
- 2) Vulnerable period
 $= 2 \times T_{fr}$
- 3) Maximum throughput
 $S_{max} = 0.184$
 $= 18.4\% \text{ at } G_1 = \frac{1}{2}$
 Throughput $S = G_1 e^{-2G_1}$

Slotted ALOHA

- 1) Synchronised transmission
 (At the beginning of every time slot)
- 2) Vulnerable period
 $= T_{fr}$
- 3) Maximum throughput
 $S_{max} = 36.8 \text{ or } 0.368$
 $= 36.8\% \text{ at } G_1 = 1$
 Throughput $S = G_1 e^{-G_1}$



Distance Vector Vs Link State Routing protocols

<u>Distance Vector Routing Protocol</u>	<u>Link State Routing Protocol</u>
• Entire routing table is sent as an update	• updates are incremental and entire routing table is not sent as update.
• Send periodic update (Every fixed interval of time) at every 30 or 90 second.	• Updates are triggered not periodic.
• Updates are broadcasted (not in RIPV2) Due to this large BW is used	• updates are multicasted • No unnecessary BW consumption.
• updates are sent to directly connected neighbor only.	• update are sent to entire network
• Routers don't have end to end visibility of entire network	• Routers have visibility of entire network of that area only
• Prone to routing loops	• No routing loops

	RIP	OSPF	EIGRP
Nature	Distance vector	Link State	Hybrid
Scale	Small networks	Enterprise networks	Medium
Routing	Classful routing, loop counter mechanism	Classless	Classless 100% loop free
Metrics	Number of hops	The inverse of BW of links	Available BW, delay, load MTU & the link reliability
updates	Periodic (Broadcasts)	Incremental	... Incremental updates (multicasts)
Failure Recovery	Slow convergence	Generally faster than RIP	DUAL Algorithm
Load Balancing	only supported on equal-cost paths	Support six equal cost paths, but difficult to implement	Supports six equal cost paths but commonly ignored due to its complexity & instability

RIP

OSPF

Features	Version 1	Version 2	
Algorithm	Bellman Ford	Dijkstra	
Path selection	Hop based		shortest path
Routing	Classful	classless	classless
Transmission	Broadcast	Multicast	Multicast
Administrative Distance	120	110	
Hop count limitation	15	MD5	No Limitation
Authentication	No	MD5	MD5
Protocol	UDP		TCP
Convergence Time	More	(Seconds)	Less