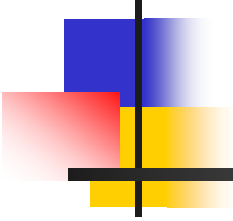


# EE3204/EE3204E- Computer Communication Networks I (part 1)



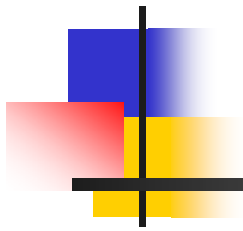
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Lecturer:

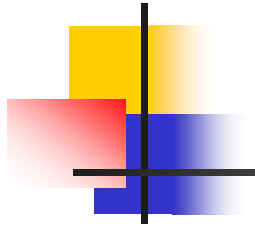
*Mohan Gurusamy*

Associate Professor

Electrical and Computer Engineering



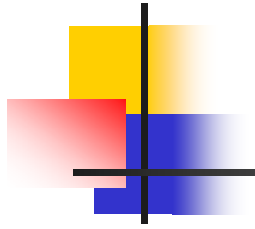
# Network Basics



# Networks: What? Why?

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- A Network is a set of systems interconnected by communication links that is primarily used for information transfer
- System – computer, switch, router
- Different forms of information:
  - Text, voice, audio, video, picture, graphics, animation

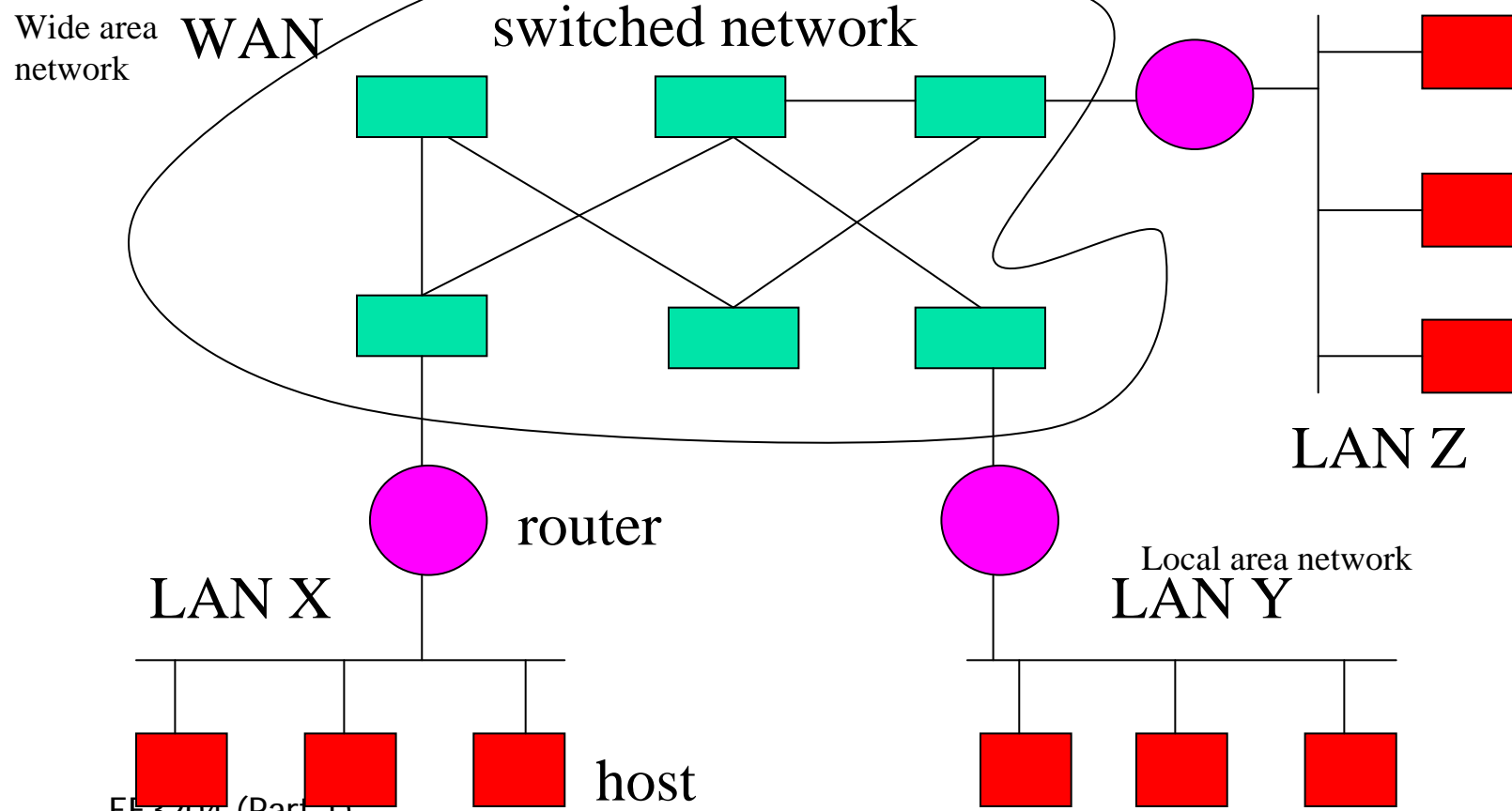


# Network Applications

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1. World Wide Web (www)
2. E-mail, ftp
3. Video and audio streams
4. Voice over Packets
5. Distributed databases (banking, airline Transactions)

# An Example Computer Network





# Broadcast vs Point-to-Point Links

---

- Local area networks (LAN)
  - Host is a device where application runs; source/destination of a traffic; eg: computer, server, phone
  - Multiple hosts (nodes) connected by a broadcast link (also called multiple access or shared access link)
  - One node transmit; all nodes receive
  - Hosts can be connected by wireless LAN (not shown in the example-computer-network figure)
- Switched networks
  - Characterized by point-to-point links
  - Data traverses through one or switches and links
  - Data is forwarded from an input link to an output link within a switch (called switching)



# Telephone Networks

---

- Analog signal is converted to PCM (pulse code modulation) digital signal using CODEC (coder-decoder)
- Voice signal frequencies  $< 4$  kHz
- Sampling theorem: sample at the rate of at least twice the maximum frequency to reconstruct the signal completely
  - $2 \times 4000 = 8000$  samples per second
  - 1 sample per 125 microseconds
  - $8000 \times 8 = 64000$  bits per second
  - data rate required for a voice signal is 64 Kbps, termed as DS0 (Digital Signal Level 0)



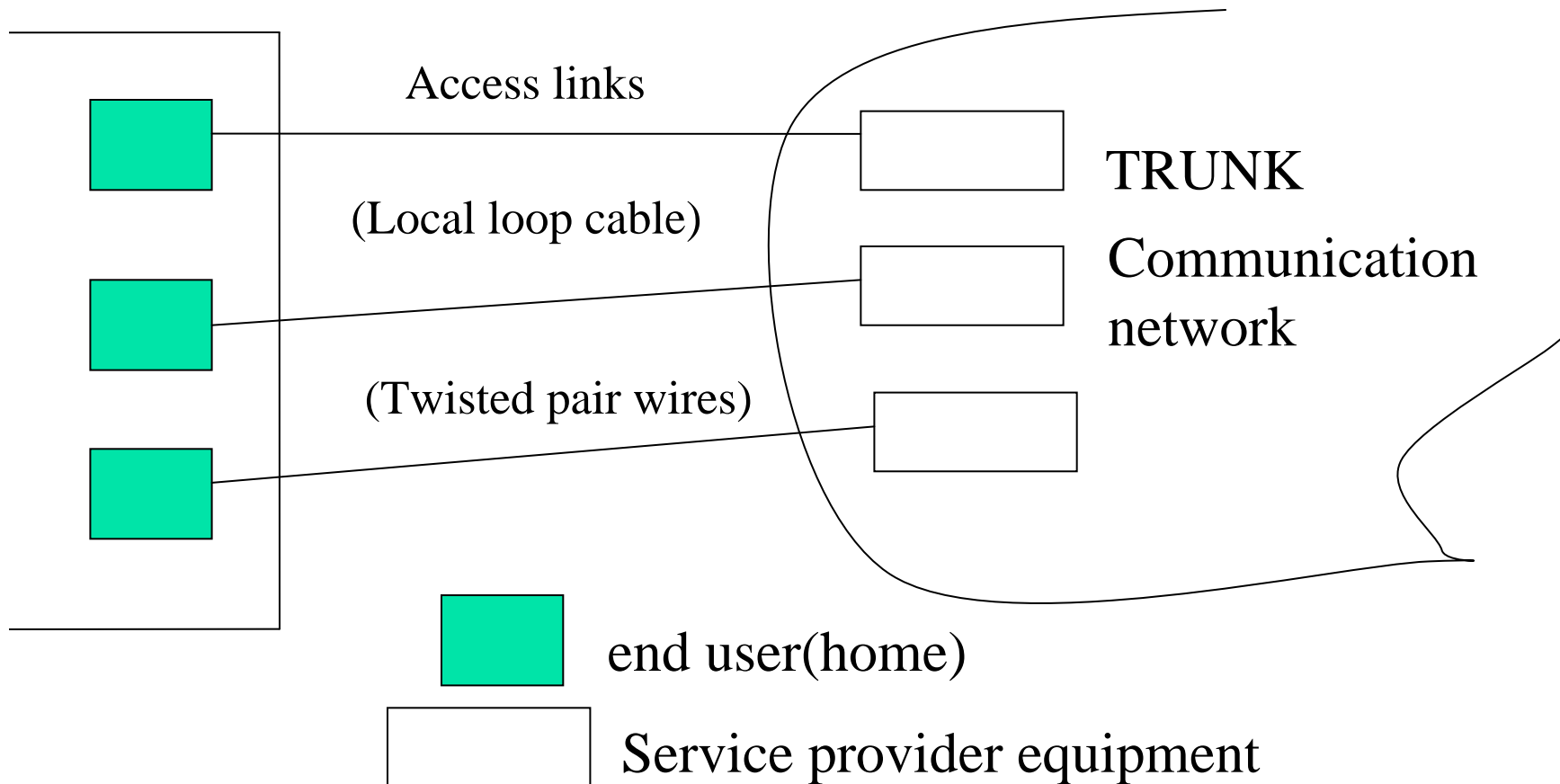
# Some Definitions

---

- Bandwidth
  - Bandwidth of the transmitted signal as constrained by the transmitter and transmission medium, expressed in Hertz
- Data Rate
  - The rate in bits per second (bps) at which data can be communicated
- Channel Capacity
  - The maximum rate at which data can be transmitted over a communication path or channel under certain conditions such as SNR (signal-to-noise ratio)
- In the context of “computer communication networks” the terms bandwidth, data rate, and capacity are sometimes used interchangeably. Students should be able to distinguish them based on “unit” and “context”.



# Telephone Network





# Multiplexing Hierarchy

---

- Multiplex several low rate signals into high rate signal
- Time division multiplexing
- Digital Signal (DS) hierarchy
- DS0, DS1, DS1c, DS2, DS3, and DS4
- DS2 (6.312 Mbps not widely used)
- Local loop: analog : up to 4 kHz
- Digital subscriber line (DSL) (1.544 to 7 Mbps and higher) over twisted pair copper wire

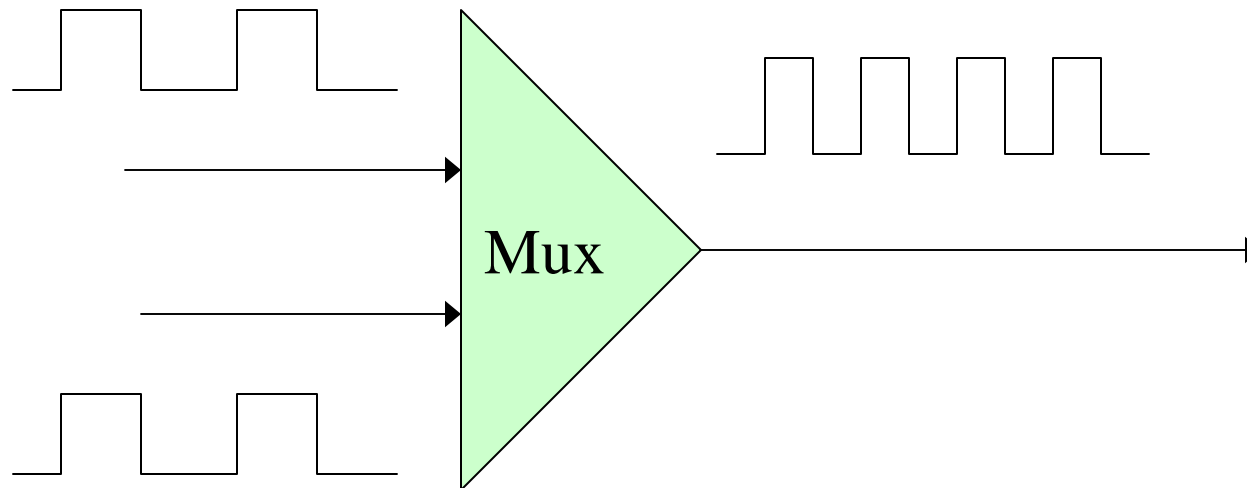


# Time Division Multiplexing

---

- Consider a multiplexer with two input links and one output link
- Data rate of each input link is  $R$  bps requires the output link rate to be at least  $2R$  bps
- If one byte is received at each input link for every  $T$  seconds, 2 bytes would be sent out at the output link for every  $T$  seconds

# Time Division Multiplexing (illustration)





# Digital Signal Hierarchy

Medium	signal	multiplex degree	# of DS0 circuits	Rate in Mbps
T1 pair	DS1	24 DS0	24	1.544
T1c pair	DS1c	2 DS1	48	3.152
T3 coax/ fiber	DS3	7 DS2	672	44.736
T4 coax/ fiber	DS4	6 DS3	4032	274.176

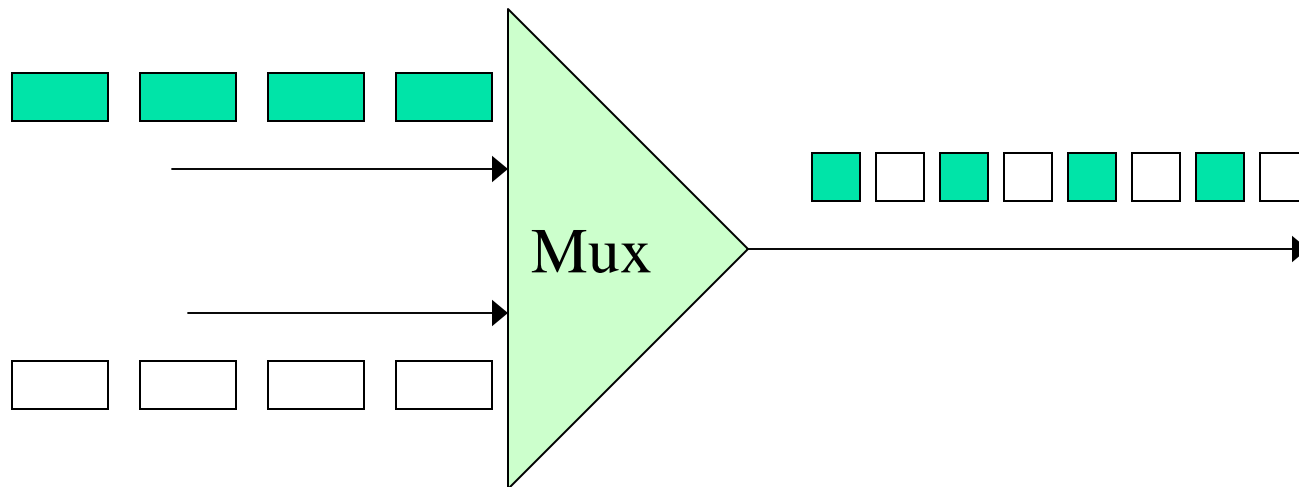


# Time division multiplexing

---

- Fixed multiplexing
  - Fixed and dedicated bandwidth is reserved
  - Guaranteed service
  - Inefficient bandwidth utilization
- Statistical multiplexing
  - On Demand multiplexing
  - No fixed and dedicated bandwidth
  - No service guarantee
  - Efficient bandwidth utilization

# Fixed Time Division Multiplexing (FTDM)





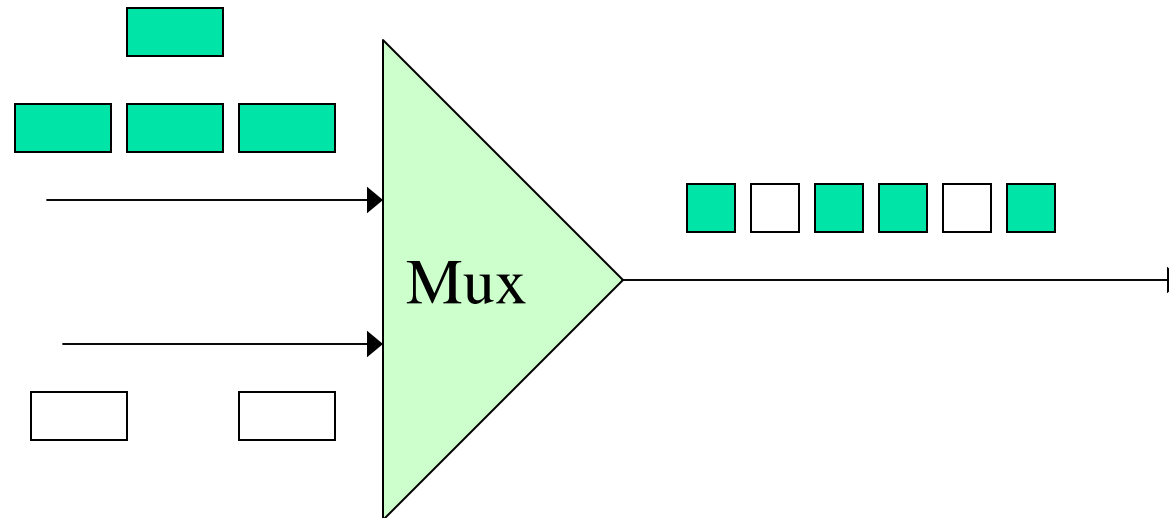
## FTDM (contd.)

---

- For the example above, we can think of repeating frames sent on the output link.
  - Each frame carries two bytes (assuming byte interleaved multiplexing)
  - First byte comes from the top input link and second byte comes from the bottom input link

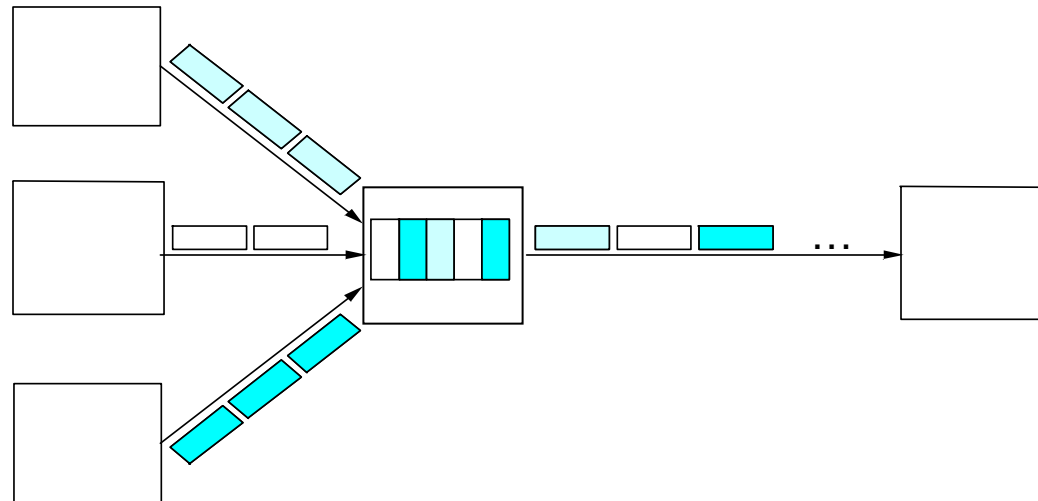


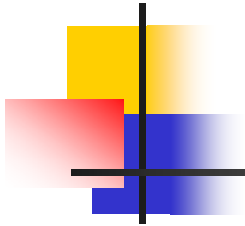
# Statistical Time Division Multiplexing



# Statistical Multiplexing in Networks

- On-demand time-division, better resource sharing
- Schedule link on a per-packet basis
- Packets from different sources interleaved on link
- Buffer packets that are *contending* for the link
- Buffer (queue) overflow is called *congestion*

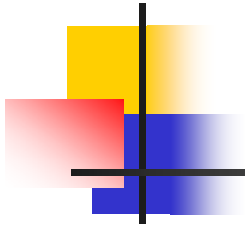




# Computer Networks - Classification

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- A computer network comprises nodes and links
- Nodes (computers, switches, routers, gateways)
- Links
  - Wired: twisted pair, coax cable, optical fiber
  - Wireless: microwave, spread spectrum radio, infrared
- Classification
  - Local area networks (LANs)
  - Metropolitan area networks (MANs)
  - Wide area networks (WANs)



# Local Area Networks

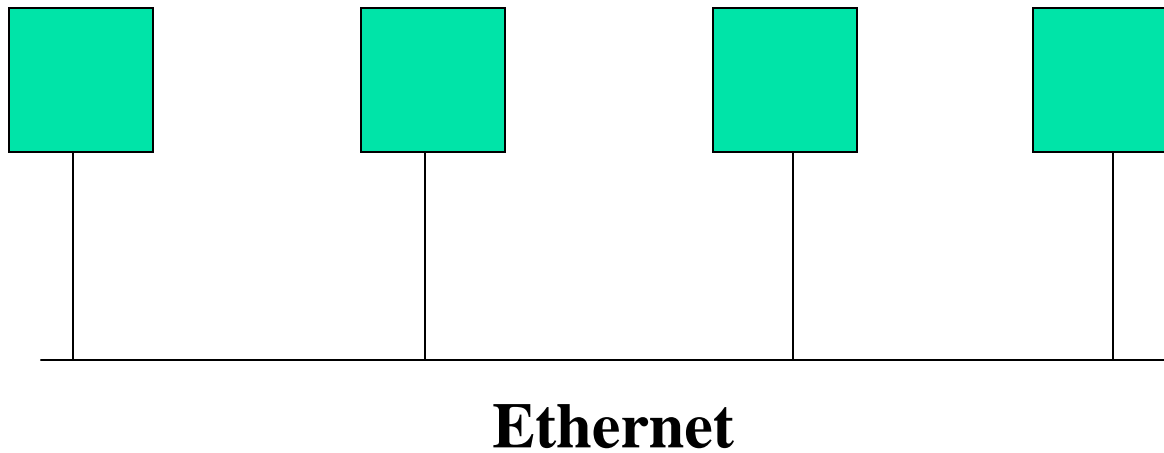
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- Small geographical area, e.g., campus
- Multiple access link, shared medium
- Medium access control
  - Ethernet (IEEE 802.3)
    - Broadcast medium
    - 10 Mbps, 100 Mbps
  - Token Ring (IEEE 802.5)
    - Token passing ring
    - 4 Mbps, 16 Mbps
  - Wireless (IEEE 802.11)



# Local Area Networks - Ethernet

---





# Ethernet –MAC -Principle

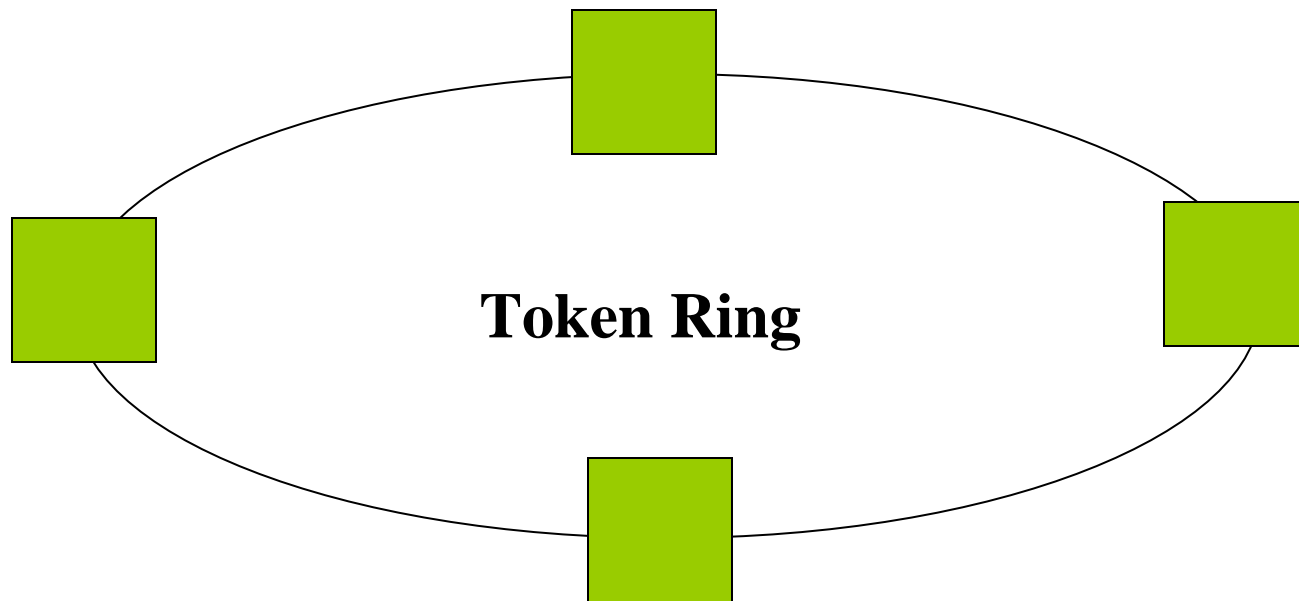
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- Broadcast medium
  - Traffic transmitted by a host can be seen by every host
- Possibility of more than one host transmitting data frames at the same time leading to collisions
- Mechanisms for detecting a collision and retransmitting data in the event of a collision are used [Details are available in Lecture Notes-4 on Ethernet]



# Local Area Networks – Token Ring

---





# Token Ring – MAC - Principle

---

- Broadcast Medium
- No collision
- A special packet called Token is circulating around the ring
- A host waits for its turn; captures token; transmits traffic; passes token to the next host
- Packet makes one full round and is drained at the source

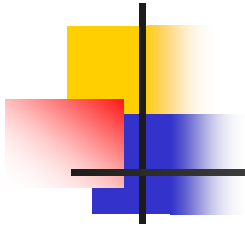




# Metropolitan Area Networks

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- Medium geographical area: city
- Fiber Distributed Data Interface
  - Token ring, 100 Mbps
- Synchronous Optical Network (SONET)
  - Fixed bandwidth allocation,  $> 51.84$  Mbps
  - Fixed TDM
- Asynchronous Transfer Mode (ATM)
  - Flexible bandwidth allocation, statistical TDM



# Wide Area Networks

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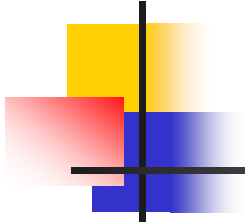
- Large geographical area (country, continent, world)
- Irregular topologies, point-to-point links
- Interconnecting several LANs and MANs
- IP, SONET, ATM, WDM technologies
- IP is carried on SONET, ATM, WDM networks
  - IP – Internet Protocol
  - SONET – Synchronous Optical Networks
  - ATM – Asynchronous Transfer Mode
  - WDM - Wavelength Division Multiplexing (simultaneous transmission of messages on different wavelengths on a fiber each operating at the rate of a few Gbps)
- Moving from Gigabit networks to Terabit networks, e.g. WDM (wavelength division multiplexing) networks



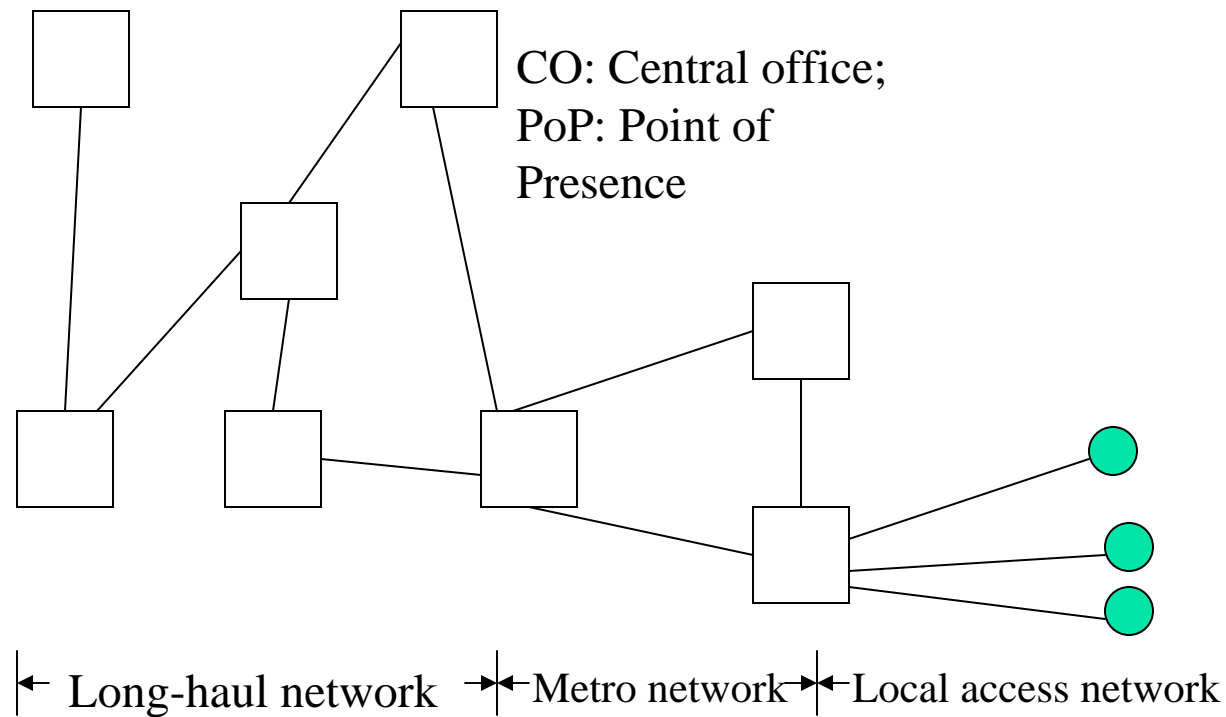
# Access Networks

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- Last Leg of telecom networks providing services to homes and small businesses
- Telephone network-based DSLs and Community Antenna/cable TV (CATV)-based Cable modems
- Offer downstream bandwidth of a few Mbps only
- Can be up to a few tens of Mbps with technology advances
  - have limitations on distance (from CO to subscriber) of up to 4km
- In CATV network channel bandwidth is shared, and at high loads the performance degrades
- Passive Optical Networks (PONs) are becoming popular solutions for access networks
- ATM PON (APON), Gigabit PON (GPON), Ethernet PON (EPON), WDM PON

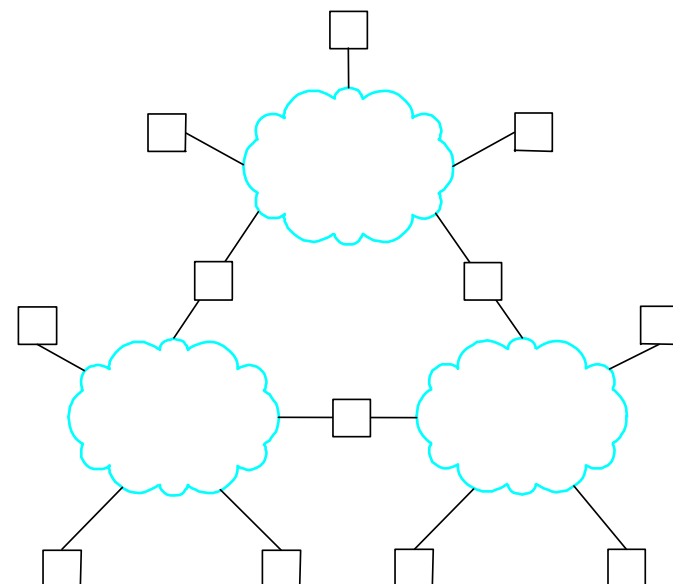
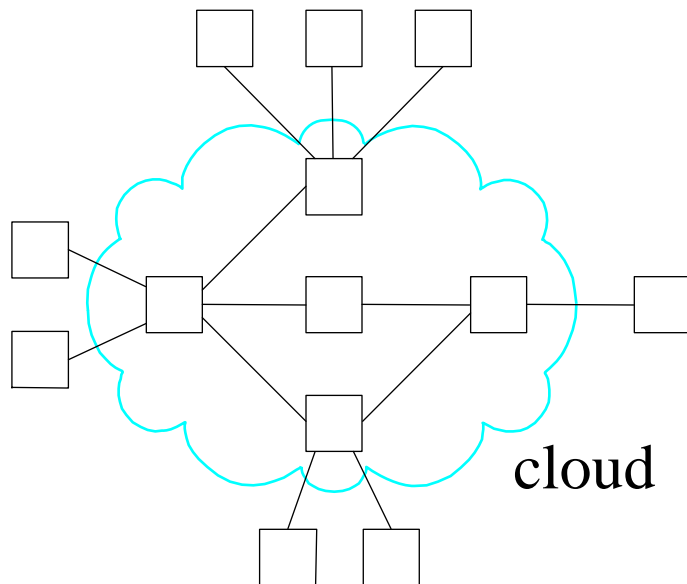


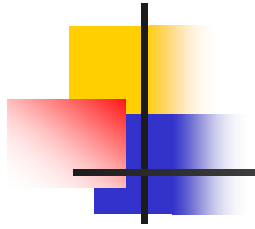
# Telecommunication Networks



# Switched Networks

- A network can be defined recursively as...
  - two or more nodes connected by a link, or
  - two or more networks connected by two or more nodes





# Network Services

---

- Connection-oriented service (eg. ATM, SONET, WDM)
  - Three phases: connection set up, message transmission and connection release
  - All packets in a message follow the same route
  - Packets are received in the order they were sent
  - Every node maintains connection state (id) information
  - Small header
  - Better support for Quality-of-Service (QoS)
    - QoS: delay, jitter, bandwidth, packet/cell loss rate
  - Connection setup/release overhead
  - Bandwidth utilization may not be efficient

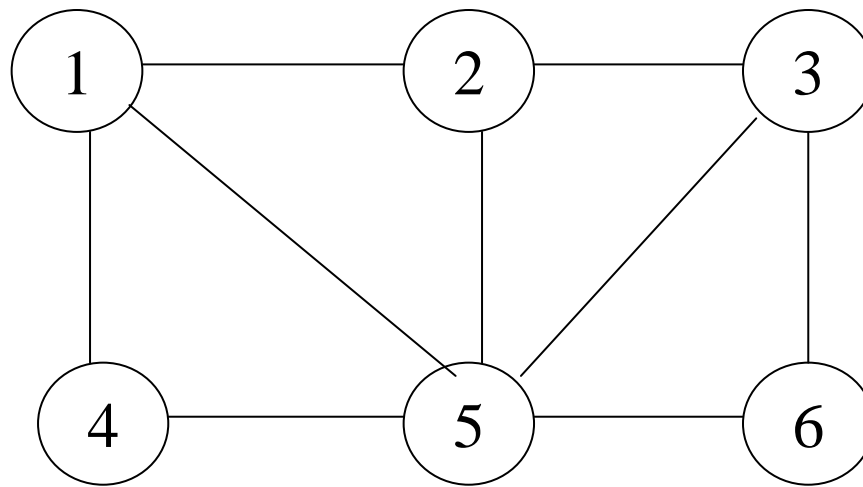


# Network Services (contd.)

---

- Connectionless service (Datagram service) (eg. IP)
  - packets in a message need not follow the same route
  - Packets may be received out of order
  - No connection state (id) information is maintained
  - large header (every packet carries source, destination information...)
  - No QoS guarantee (poor QoS support)
  - No Connection setup/release overhead
  - Bandwidth utilization is efficient

# Connection-oriented and Connectionless Routing



Source: 1 destination: 6

Connection-oriented: 1-5-3-6 (all packets)

Connectionless: 1-4-5-6 (pkt1) 1-5-3-6 (pkt2) 1-5-6 (pkt3)





# Switching Techniques

---

- Circuit switching

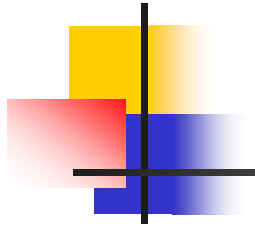
- Link bandwidth (capacity) is divided into fixed-size circuits
- eg: telephone network, SONET, WDM
- 3 distinct phases: circuit setup, data transmission, and circuit release phases
- Connection-oriented service
- An application uses a fixed path to be used by all data
- data arrive in sequence
- fixed bandwidth (BW) circuits, low BW utilization
- guaranteed service



# Switching Techniques (contd.)

## Packet Switching

- store and forward technique employed
- switching entity is a packet. Statistical multiplexing is used.
- High bandwidth utilization
- Connectionless packet switching (eg. IP)
  - Different packets of the same message may traverse different routes
  - May result in out-of-order sequence of packet delivery
  - Difficult to provide guaranteed service, usually best-effort service
- Connection-oriented virtual circuit switching, cell switching (eg. ATM)
- Congestion can occur at a node when incoming traffic speed exceeds the outgoing link capacity and sufficient buffers are not available to hold the excess data



# Addressing

---

- Address: byte-string that identifies a node
  - usually unique
- Routing: process of forwarding messages to the destination node based on its address
- Types of addresses
  - unicast: node-specific
  - broadcast: all nodes on the network
  - multicast: some subset of nodes on the network



# Packet Transfer – A simplified example (Refer slide 5)

---

- A host (or router) has 4-byte IP address (format: x.y.z.w)
  - Hierarchical address; a part of address denotes network id (say, LAN), all hosts in a network has same network id
  - Host IP address from its ASCII name is obtained using DNS (Domain Name Server)
- A host (or router) has network interface card (nic) connecting to the cable
- A host has 6-byte Ethernet (or hardware or MAC address) hardwired at nic (format: a:b:c:d:e:f)
  - Non-hierarchical address; When a host sees a packet carrying its own h/w address, it copies the packet onto itself
  - Hardware address of IP address of each host on a LAN is known to host on the same LAN using ARP (Address Resolution Protocol)



# Example (contd.) – Case 1

---

- Host A at LAN X wants to transfer a packet to Host B at LAN Z
- At host A: obtain IP address of host B using DNS and attach to the packet
- Compare Network id part of host A's IP address with B's. Since they differ, hardware address of router interface at LAN X is attached to the packet
- Broadcast the packet on LAN X
- Router at LAN X grabs the packet; uses IP address for routing the packet
- Packet reaches the router attached at LAN Z. Network id part of packet's IP address matches that of this router
  - This router knows the corresponding hardware address (using ARP)
  - Router attaches the hardware address of host B with the packet
  - Router broadcasts the packet on LAN Z
  - Host B finds a match with its hardware address
  - Host B copies the packet



## Example (contd.) – Case 2

---

- Host A at LAN X wants to transfer a packet to Host C at the same LAN X
- At host A, obtain IP address of host C using DNS and attach to the packet
- Compare Network id part of host A's IP address with C's. Since they match, Hardware address of host C at LAN X is attached to the packet
  - Hardware address of host C is known to host A using ARP
- Broadcast the packet on LAN X
- Host C at LAN X takes the packet;



# Bandwidth and Throughput

---

- Bandwidth, data rate, capacity
  - These terms might be used interchangeably as stated earlier
  - Number of bits that could be transmitted per time unit
  - link versus end-to-end
  - Bandwidth of a link (or end-to-end connection)
  - Notation: 1 Kbps bandwidth =  $10^3$  bits per second
  
- Throughput or Effective throughput
  - Number of bits transferred per unit time (measured quantity)
  - Ratio between message size and message transfer time
  - Notation: message size 1 KB =  $2^{10}$  bytes



# Message Transfer Time

---

- Transmission time ( $T_t$ )
  - Time to transmit (or pump) message bits on to the link
  - i.e time elapsed between the beginning of transmission of the first bit and the end of transmission of the last bit
  - Message size / Bandwidth
- Propagation time ( $T_p$ )
  - Time to traverse from node A to node B
  - Distance / Propagation speed
- Queuing time ( $T_q$ )
  - Time for which a message is waiting in the queue at a node before the start of transmission
- Message Latency  $T_t + T_p + T_q$  for one-way unacknowledged message transfer
  - Time required to transfer a message from A to B





# Acknowledged Message Transfer

---

- Round-Trip Time (RTT)
  - Time taken by a message to traverse from A to B then back to A
  - Usually the transmission time is excluded
- RTT is usually used in acknowledged services wherein data is transmitted in one direction and ACK is transmitted in the reverse direction
- Message Latency  $T_t + 2T_p + T_q$  for one-way acknowledged message transfer
  - Time required to transfer a message from A to B + Time required to transfer ACK from B to A

# Delay x Bandwidth Product

- Here, delay refers to the propagation time on the link (pipe)
- Amount of data “in flight” or “in the pipe”
- Example:  $100\text{ms} \times 45\text{Mbps} = 4500 \text{ Kbits}$

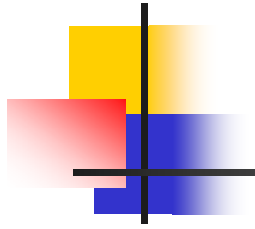




# Significance of $D \times B$ Product

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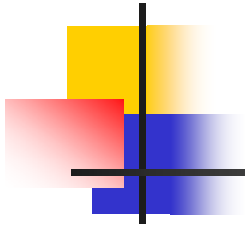
- Upon receiving a bit, if the receiver sends a signal to the sender to stop transmission, then the sender would have transmitted  $D \times B$  bits (D: Delay, B: Bandwidth)
- If the sender expects some signal (say acknowledgement) from the receiver before filling the pipe, then the bandwidth is underutilized



# (Effective) Throughput

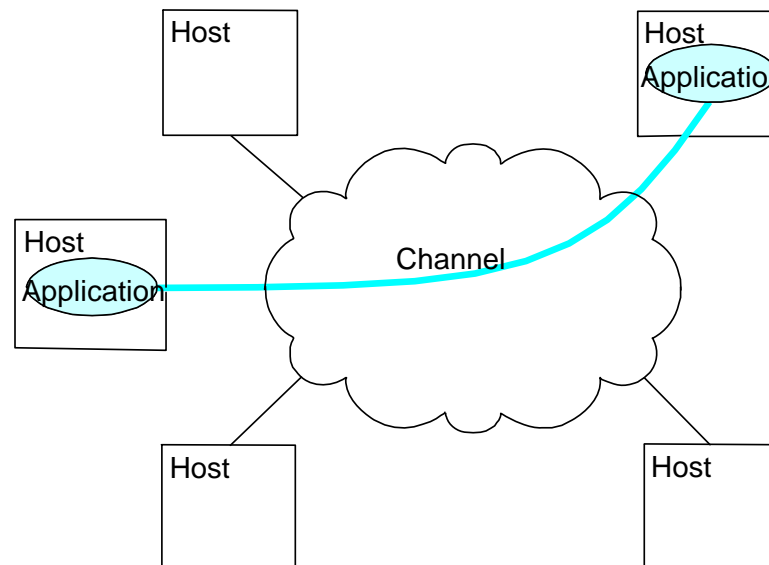
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- Depends on RTT and transmission time
  - For unacknowledged one-way message transfer, one-way propagation time is used instead of RTT
- It is desirable that the throughput is close to the bandwidth
- Case 1: RTT=10 ms, message size=1MB, b/w=10 Mbps
  - Transfer time=  $10 + (1\text{MB}/10\text{Mbps}) = 849\text{ ms}$
  - Eff. Throughput=  $1\text{MB}/849\text{ms} = 9.88\text{ Mbps}$
- Case 2: RTT=10 ms, message size=1MB, b/w=1 Gbps
  - Transfer time=  $10 + (1\text{MB}/1\text{ Gbps}) = 18.4\text{ms}$
  - Eff. Throughput=  $1\text{MB}/18.4\text{ms} = 456\text{Mbps}$



# Inter-Process Communication

- Turn host-to-host connectivity into process-to-process communication.
- Fill gap between what applications expect and what the underlying technology provides.





# Different Kinds of Channels

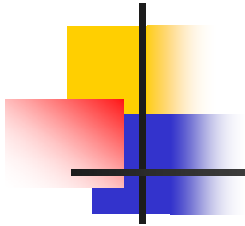
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- Request/Reply

- distributed file systems
- digital libraries (web)

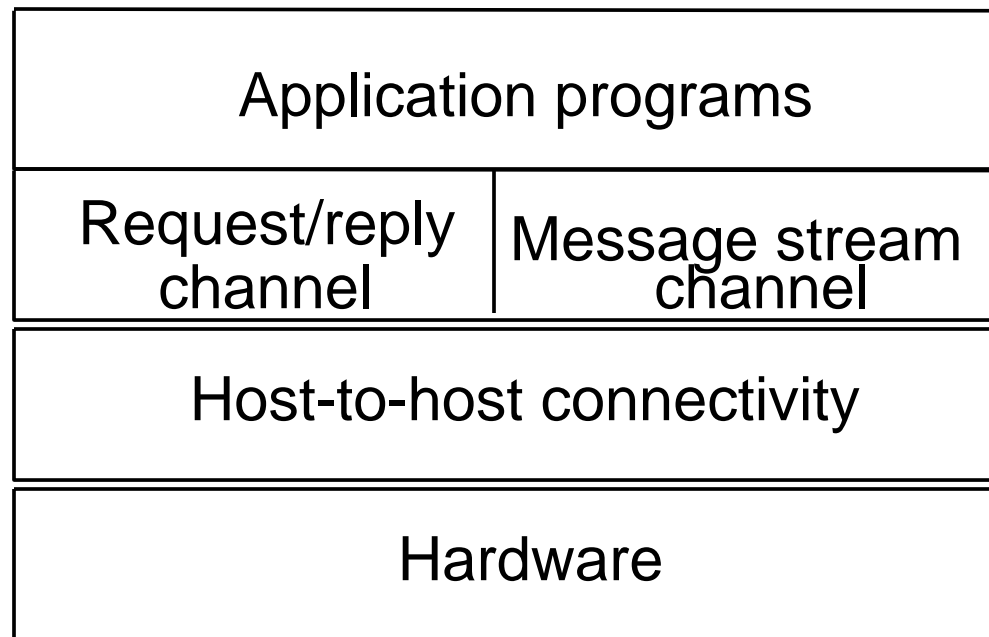
- Stream-Based

- video: sequence of frames
  - 1/4 NTSC = 352x240 pixels
  - $(352 \times 240 \times 24)/8 = 247.5\text{KB}$
  - 30 fps = 60Mbps (approx)
- video applications
  - on-demand video
  - video conferencing



# Network Architecture - Layering

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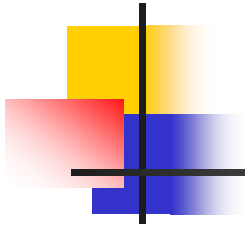


# Advantages of Layering

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- Use abstractions to hide complexity
- Abstraction naturally lead to layering
- Alternative abstractions at each layer
- Network design becomes easier as layering allows decomposing of the design problem into manageable components
- Layering aids modular design. New services can be added to a layer without modifying the functionality of other layers

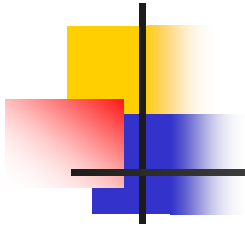




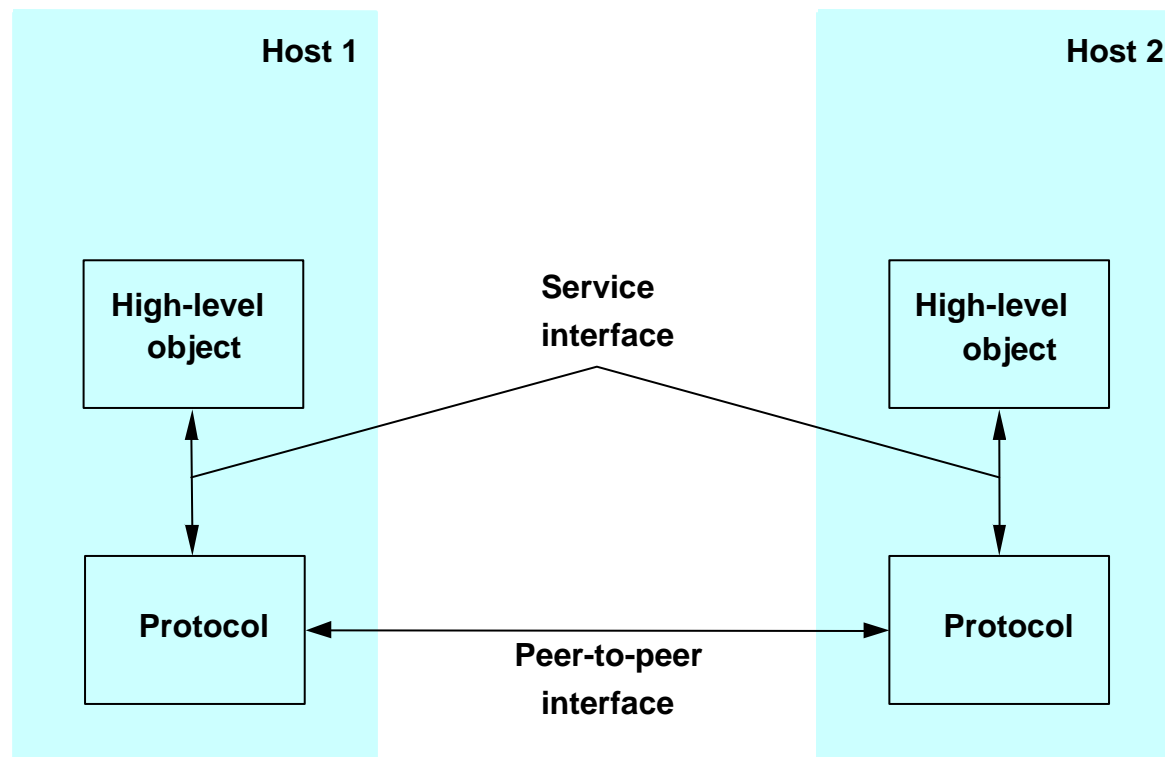
# Protocols

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- Building blocks of a network architecture
- Each protocol object has two different interfaces
  - *service interface*: operations on this protocol
  - *peer-to-peer interface*: messages exchanged with peer
- Term “protocol” is overloaded
  - specification of peer-to-peer interface
    - pseudocode /state transition diagrams
  - module that implements this interface
    - may exist more than one implementation for the same specification (module interoperability)



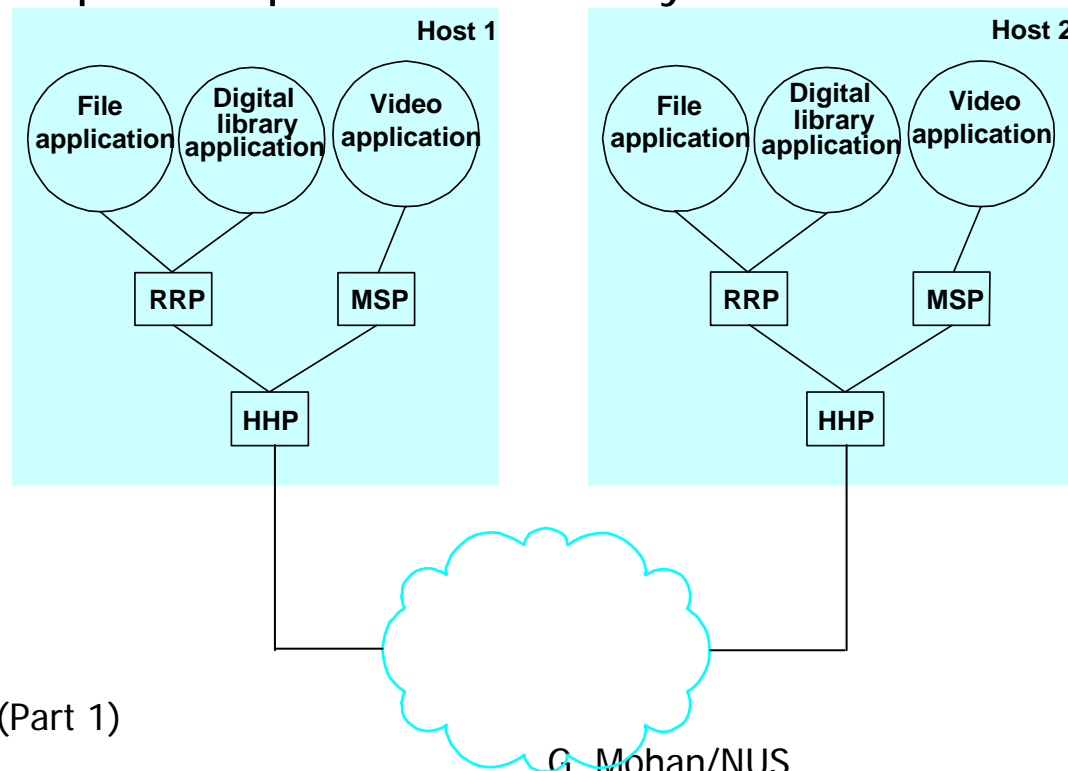
# Interfaces



# Protocol Machinery

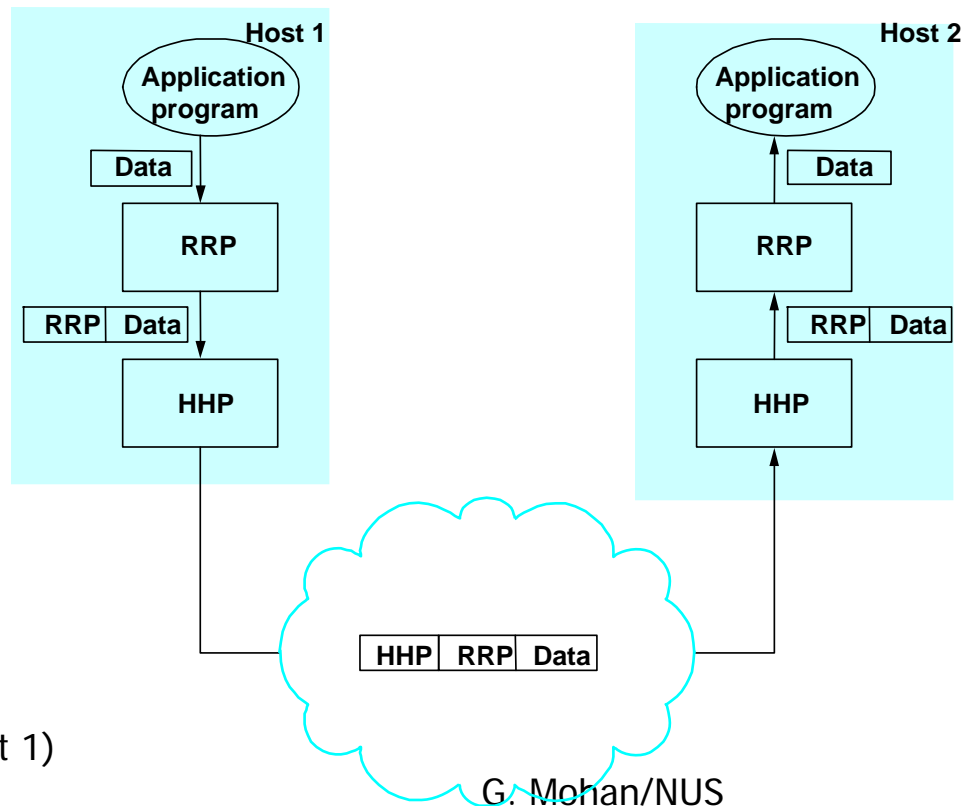
- Protocol Graph

- most peer-to-peer communication is indirect
- peer-to-peer is direct only at hardware level



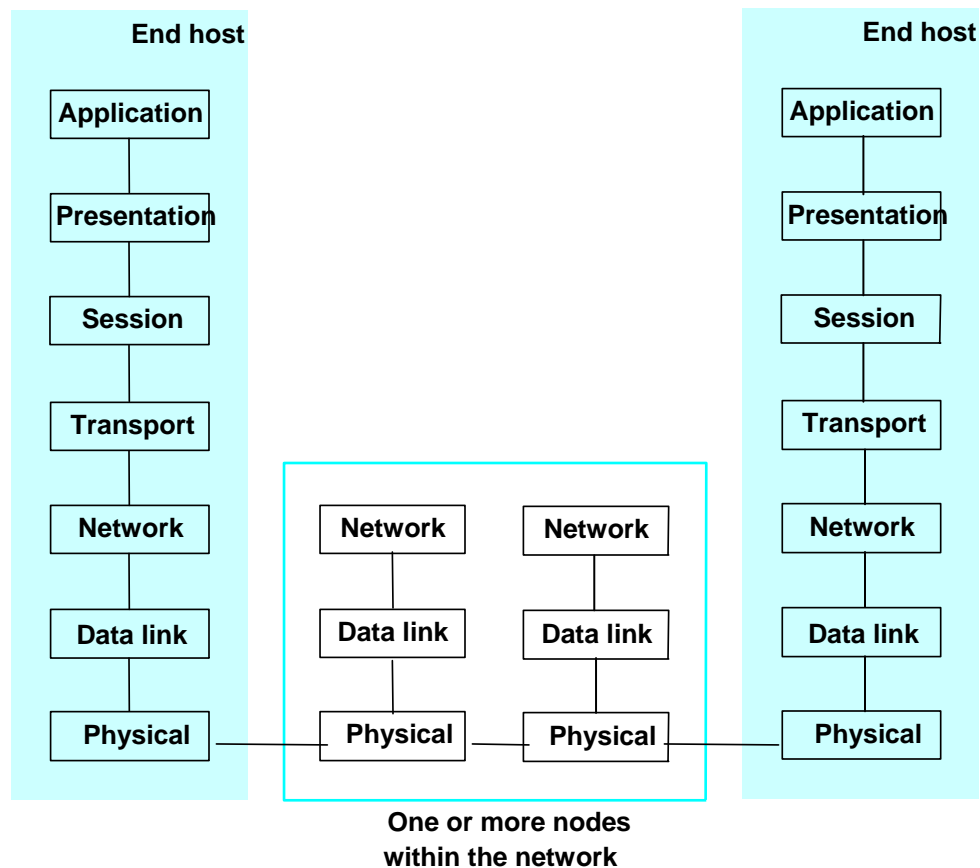
# Protocol Machinery (cont)

- Multiplexing and Demultiplexing (demux key)
- Encapsulation (header/body)



# ISO-OSI Architecture

Intl. Standards Organization-Open Systems  
Interconnection





# Functions of Layers

---

- Physical layer (Layer 1, in short L1)
  - Transmission of raw bits over a communication link
  - Electrical characteristics: voltage levels, signal durations
  - Mechanical characteristics: pluggable connector specifications
- Data link layer (Layer 2)
  - Aggregation of bits into frames, error and flow control (optional)
  - Medium access control (MAC), e.g. Ethernet
- Network layer (Layer 3)
  - Routing packets between nodes, may do congestion control (IP routers do not do congestion control)
- Transport layer (Layer 4)
  - Handles messages Process-to-process channel, end-to-end layer, end-to-end message transfer, reliable message delivery, congestion control, flow control



# Functions of Layers (contd.)

---

- Session layer
  - provide name space to collectively refer to different transport streams that are a part of a single application
  - eg. Video conferencing (audio and video streams)
- Presentation layer
  - data representation (integer, string)
  - how data is transmitted MSB first or LSB first.
  - Compression, encryption
- Application layer
  - Provides protocols to support different applications such as FTP, email/SMTP (Simple Mail Transfer Protocol)



# Internet Architecture

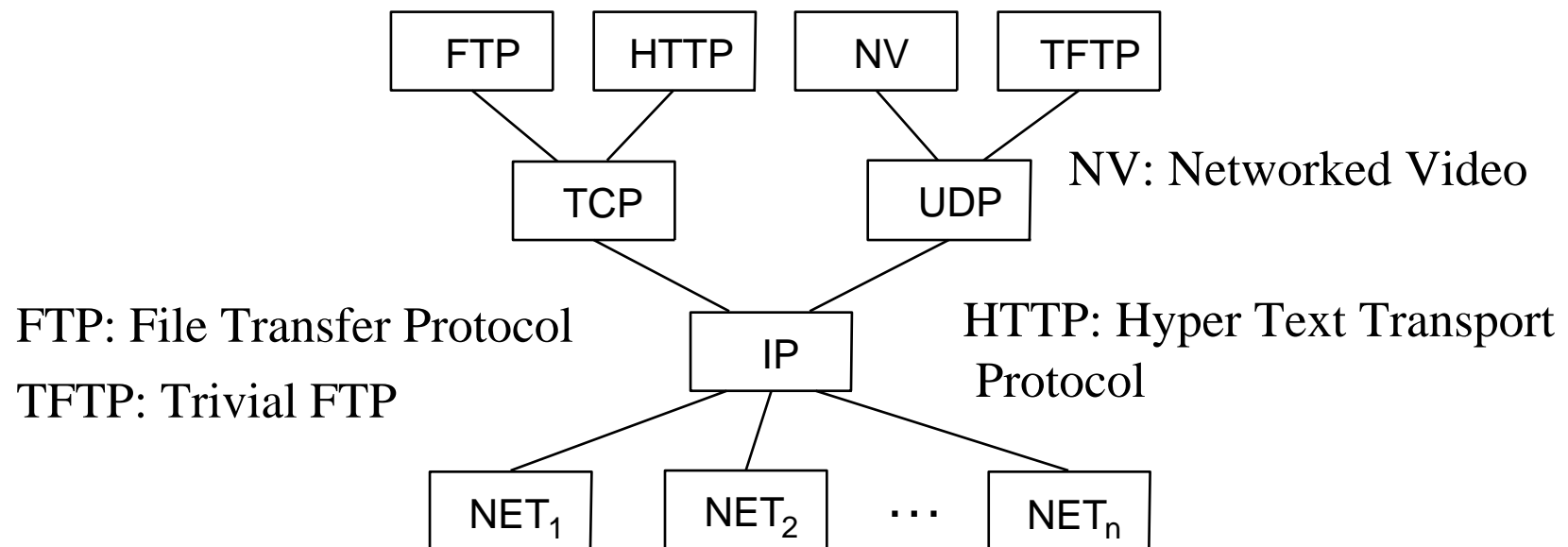
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- Five layer architecture
- Application Layer (L5, in ISO-OSI architecture, it is L7)
  - Present at the end systems (hosts)
- TCP or UDP layer (L4)
  - Present at the end systems (hosts); lowest end to end layer
  - TCP: Transmission Control Protocol
    - Reliable connection oriented transport service, guarantee for no errors, in-order delivery;
    - Flow control, error control, congestion control
  - UDP: User Datagram Protocol
    - Unreliable connectionless message transport service
- IP layer (L3)
  - Routing; does not do congestion
- Physical Network (Link layer (L2) and physical layer (L1))
  - Flexible to support any underlying wired or wireless networks
    - E.g. Ethernet, wireless LAN, SONET, ATM



# Internet Architecture

- Defined by Internet Engineering Task Force (IETF)
- Hourglass Design
- Application vs. Application Protocol (FTP, HTTP)



# Protocol Data Units in the TCP/IP Architecture

