Chapter Four

The OSI and TCP/IP Models

Communication and Layer Architecture

- A network is a combination of hardware and software that sends data from one location to another.
- The hardware consists of the physical equipment that carries signals from one point of the network to another.
- The **software** consists of **instruction sets** that make possible the **services** that we expect from a **network**
- For example, the task of **sending** an **e-mail** from one point in the world to another can be broken into several

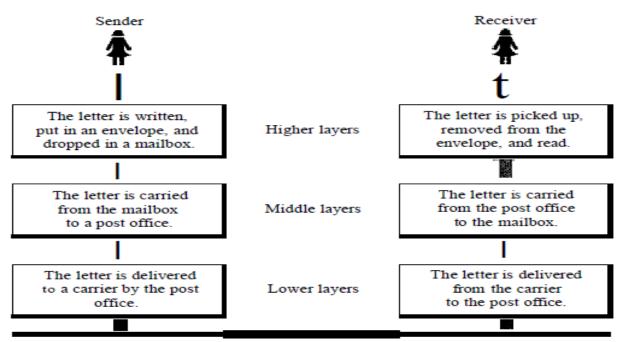
tacks and performed by a congrete coftware package

Architecture----

- Each software package uses the services of another software package.
- At the lowest layer, a signal, or a set of signals, is sent from the source computer to the destination computer.

Layered Tasks

- We use the concept of layers in our daily life.
 - As an **example**, let us consider **two friends** who **communicate** through **postal** mail.
- The process of sending a letter to a friend would be complex if there were no services available from the post office.



The need for Standards

- Without agreed on common standards,
- **devices** made by **different manufacturers** or **vendors** would **not** be able to **communicate**.
- ✓ A protocol is a set of rules that determines how two devices will communicate.
- Networking standards define the rules for data communications that are needed for interoperability of networking technologies and processes.

The need for Standards----

Standards help in

creating and maintaining open markets and

allow different vendors to compete on the basis of the quality of their products while being compatible with

- existing market products.
- ➤ Over the past years many of the **networks** that were **built** used different **hardware** and **software implementations**,

as a result they were **incompatible** and it became **difficult**

for **networks** using **different specifications** to **communicate** with each other.

The need for Standards----

- To address the **problem** of **networks** being **incompatible** and **unable** to **communicate** with **each other**,
- the **International Standardisation** for **Organisation** (ISO) researched various **network schemes**.
- The ISO recognised there was a need to create a NETWORK MODEL that would help vendors create interoperable network implementations.

- A communication architecture is a strategy for connecting host computers and other communicating equipment.
- It defines necessary elements for data communication between devices.
- A communication architecture, therefore, defines a standard for the communicating hosts.
- A programmer formats data in a manner defined by the communication architecture and passes it on to the communication software.

architecture.

- Separating communication functions adds flexibility, for example, we do not need to modify the entire host software to include more communication devices.
- Layer architecture simplifies the network design.✓ It is easy to debug network applications in a layered
- architecture network.
 The network management is easier due to the layered
- Network layers follow a set of rules, called protocol.
- The **protocol** defines the **format** of the **data** being exchanged,

and the **control** and **timing** for the **handshake** between layers

Open Systems Interconnection (OSI)

- International standard organization (ISO) established a committee in 1977 to develop an architecture for computer communication.
- the result of this effort.

Open Systems Interconnection (OSI) reference model is

- In 1984, the **Open Systems Interconnection (OSI)** reference **model** was approved as an **international standard for communications architecture**.
- The term "open" denotes the ability to connect any two systems which conform to the reference model and

associated standards

Open Systems Interconnection (OSI)-----

- The **OSI model** is now considered the **primary Architectural** model for inter-computer communications.
- The **OSI model** describes **how information or data makes** its way from **application programmes**(such as **spreadsheets**) through a **network medium** (such as **wire**) to another

• The OSI reference model divides the problem of moving **information** between computers over a **network medium** into

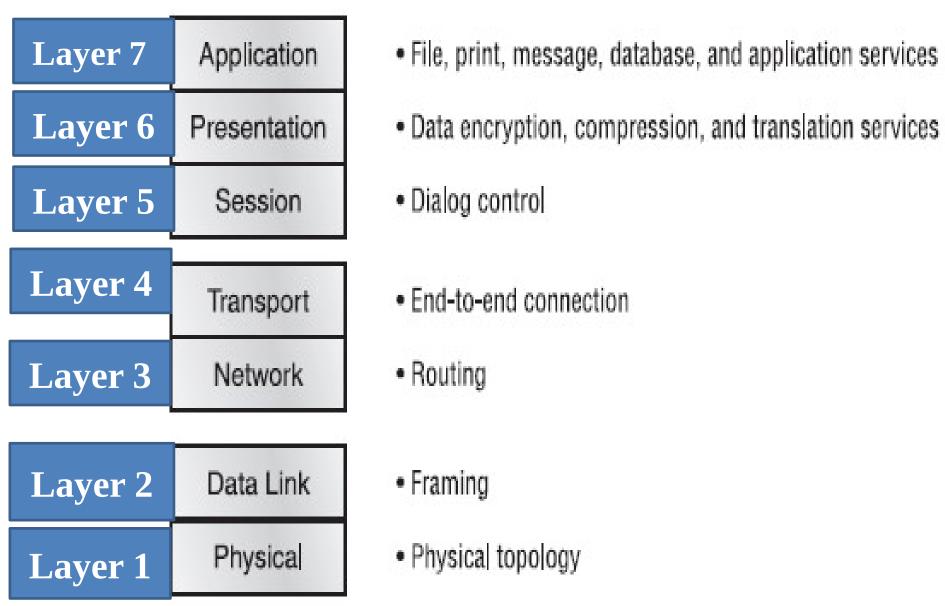
application programme located on another network.

SEVEN smaller and more **manageable problems**.

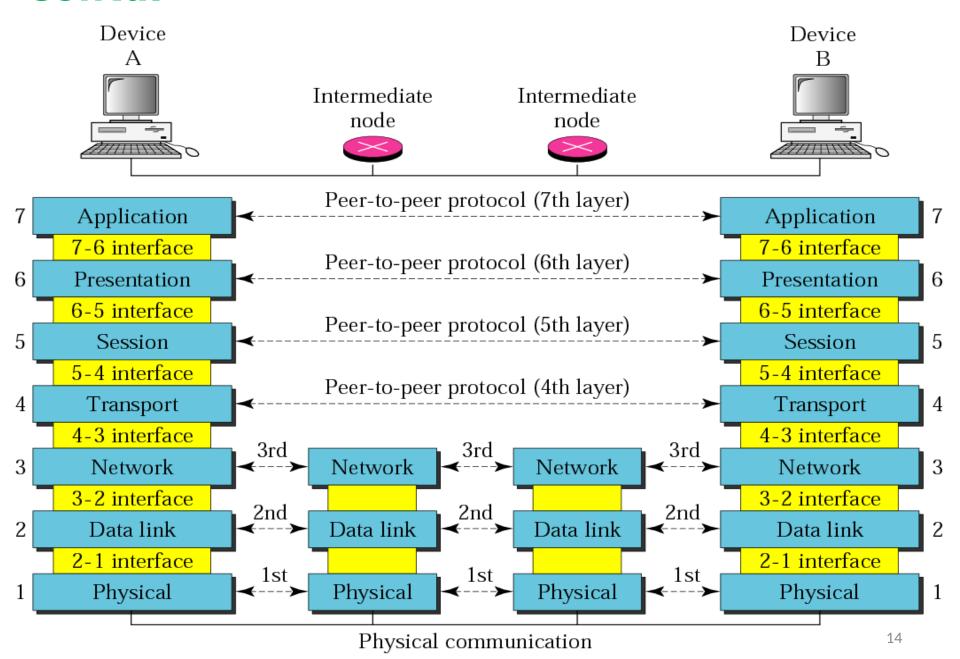
known as layering.

This **separation** into smaller more **manageable functions** is

The Seven Layers



- The figure in the next slide shows the layers involved when a message is sent from **device A** to device B.
- ✓ As the message travels from **A** to **B**, it may pass through many intermediate nodes.
- These intermediate nodes usually involve only the first three bottom layers of the OSI model.



- In developing the model, the designers distilled the process of transmitting data to its most fundamental elements.
- They identified which networking functions had related uses and **collected** those **functions** into **discrete** groups that became the **layers**.
- **Each layer defines** a family of **functions distinct** from those of the other layers.
- By defining and localizing functionality in this fashion, the designers created an **architecture** that is both comprehensive and flexible. 15

- Most importantly, the OSI model allows complete interoperability between otherwise incompatible systems.
- Within a single machine, each layer calls upon the services of the layer just below it.
- Layer 3, for example, uses the services provided by layer
 2 and provides services for layer 4.
- **Between machines, layer** *x on one machine communicates with layer x on* **another machine**.
- This communication is governed by an agreed-upon series of rules and conventions called protocols.
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Cont----

- The processes on each machine that communicate at a given layer are called peer-to-peer processes.
- Communication between machines is therefore a peer-to-peer process using the protocols appropriate to a given layer.

Peer-to-Peer Processes

- At the physical layer, communication is direct:
- ✓ At the higher layers, however, communication must move down through the layers on device A, over to device B, and then back up through the layers.
- Each layer in the sending device adds its own information to the message it receives from the layer just above it and passes the whole package to the layer just below it.
- At layer 1 the entire package is converted to a form that can be transmitted to the receiving device.

Peer-to-Peer Processes-----

- At the receiving machine, the message is unwrapped layer by layer, with each process receiving and removing the data meant for it.
- For **example**, **layer 2 removes** the data meant for it, then passes the **rest to layer 3**.
- ✓ **Layer 3** then **removes** the **data** meant for it and **passes** the **rest** to **layer 4**, and so on.

Interfaces Between Layers

- The passing of the data and network information down through the layers of the
- sending device and back up through the layers of the
- receiving device is made possible by an
 - interface between each pair of adjacent layers.
- Each interface defines the information and services a layer must provide for the layer above it.
- Well-defined interfaces and layer functions provide modularity to a network.

Interfaces Between Layers---

 As long as a layer provides the expected services to the layer above it,

the **specific implementation** of its **functions** can be **modified** or **replaced** without requiring **changes** to the surrounding **layers**.

Organization of the Layers

- The seven layers can be thought of as belonging to three subgroups.
- Layers 1, 2, and 3 physical, data link, and network - are the network support layers; they deal with the physical aspects of moving data from one device to another (such as electrical specifications, physical connections,

physical addressing, and transport timing and

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Cont----

Layers 5, 6, and 7- session, presentation, and application-can be thought of as

the user support layers; they allow interoperability among unrelated software systems.

Layer 4, the transport layer,

links the two subgroups and ensures that what the lower layers have transmitted is in a form that the upper layers can use.

- The upper OSI layers are almost always implemented in
 coftware:
- software;
 lower layers are a combination of hardware and

software, except for the **physical layer**, which is mostly

- hardware
- The figure shown in the next slide, gives an overall view of the OSI layers,
- **D7** means the **data unit** at **layer 7**, **D6** means the **data unit** at
- layer 6, and so on.

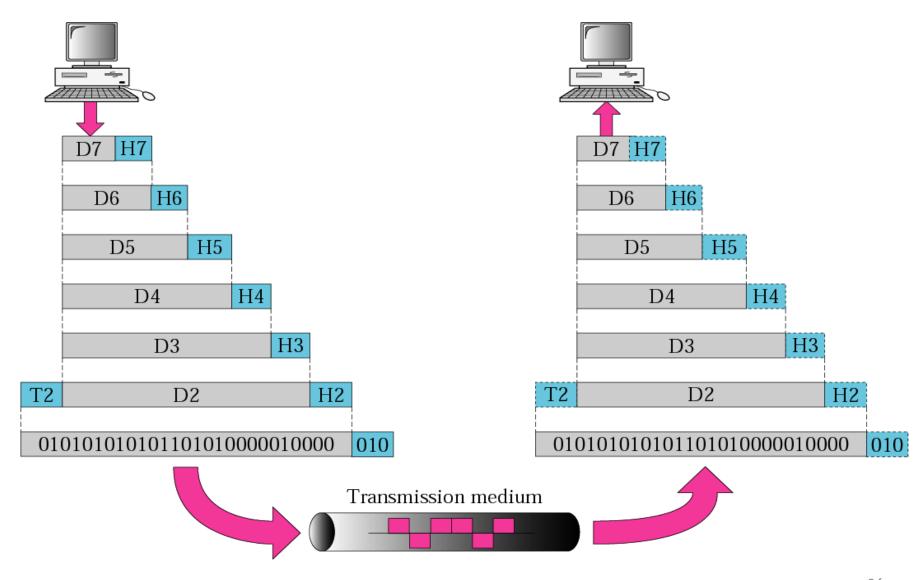
 ✓ The process starts at layer 7 (the application layer), then

morros from larray to larray in descending segmential endex

- At each layer, a header, or possibly a trailer, can be added to the data unit.
- Commonly, the trailer is added only at layer 2.
- When the formatted data unit passes through the physical layer (layer 1),

it is **changed** into an **electromagnetic signal** and **transported** along a **physical link**.

An exchange using the OSI model



- Upon reaching its destination, the signal passes into layer
 1 and is transformed back into digital form.
- ✓ The data units then move back up through the OSI layers.
- As each block of data reaches the **next higher layer**, the **headers** and **trailers attached** to it at the corresponding **sending layer** are **removed**, and actions **appropriate** to that **layer** are taken.
- the **message** is **again** in a form appropriate to the
- application and is made available to the recipient.

By the time it reaches layer 7,

Encapsulation

- A packet (header and data) at level 7 is encapsulated in a packet at level 6.
- ✓ The whole packet at level 6 is encapsulated in a packet at level
 5, and so on.
- In other words, data portion of a packet at level N-1 carries the whole packet (data and header and may be trailer) from level N.
- ✓ The concept is called **encapsulation; level N 1** is **not** aware of which part of the **encapsulated packet is data** and which part is the **header or trailer**.
- For **level N 1**, the whole **packet** coming from **level N** is treated as one **integral unit.**

Benefits of OSI Model

1. Provide a wide variety of choice

- Customers have a wide variety of choice since software/
 hardware from different manufactures work together in harmony.
- OSI model can fit to any compatible software/hardware
 from different users in other parts of the world.

2. It does not rely on a specific operating system

OSI is convenient since errors are dealt with at each level, as different levels operate automatically independent of each other.
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Benefits of OSI Model -----

This makes it easier to troubleshoot problems that may arise at each stage, by separating the networks into small manageable pieces.

3. To understand the common terms used in networking

- OSI model also help the user to understand different networking terms and functional relationship applied on multiple networks.
- In addition, the user also understand how new technologies are developed in the existing networks.

Benefits of OSI Model----

- 4. Interprets product functionality at each stage
- The **OSI model** simply uses **different stages** of **functionality**.
- For instance, each stage has specific functions to ensure all networks operate without technical hitches.
- Also, each layer has it own interface specifications and a well-defined connector.
- 5. Encrypt data for security purposes
- Decryption and encryption services are also available for security purposes.
- **Expansion** and **compression** of messages is simplified to

ensure it travels from one system to another efficiently.

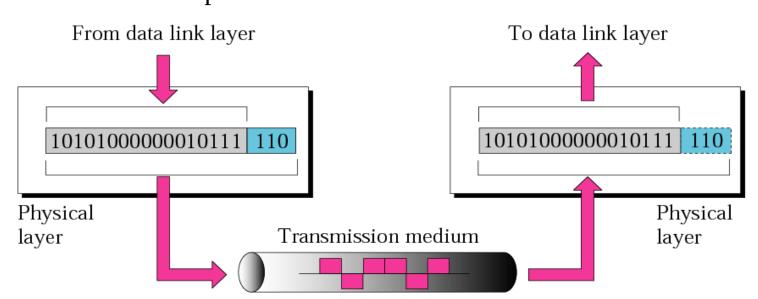
Benefits of OSI Model-----

6. It is easier to add multiple network models

- The **OSI model** is designed in such a way that user further **extend new protocols** within the **process**.
- This means you can use additional layered architecture other than the existing one.
- Due to its complexity, poor performance can be obtained in day to day applications, thereby it requires great technical know-how

1. Physical Layer

- The physical layer coordinates the functions required to carry a bit stream over a physical medium.
- It deals with the mechanical and electrical specifications of the interface and transmission medium.
- It also defines the procedures and functions that physical devices and interfaces have to perform for transmission to occur.



The physical layer is responsible for movements of individual bits from one hop (node) to the next.

1. Physical Layer

The physical layer is also concerned with the following:

- 1. Physical characteristics of interfaces and medium
- The physical layer defines the characteristics of the interface between the devices and the transmission medium.
- It also defines the type of transmission medium.

(sequence of 0s or 1s) with no interpretation

- 2. Representation of bits.
- The physical layer data consists of a stream of bits

- To be transmitted, bits must be encoded into signals electrical or optical.
- The **physical layer** defines the type of **encoding** (how **0s** and **1s** are changed to signals).

3. Data rate.

- The transmission rate the number of bits sent each second - is also defined by the physical layer.
- In other words, the **physical layer** defines duration of a bit, which is how long it lasts.

4. Synchronization of bits.

- The sender and receiver not only must use the same bit rate but also must be synchronized at the bit level.
- In other words, the sender and the receiver clocks must be synchronized.

5. Line configuration.

■ The physical layer is concerned with the connection of devices to the media (point – to – point or multipoint).
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6. Physical topology.

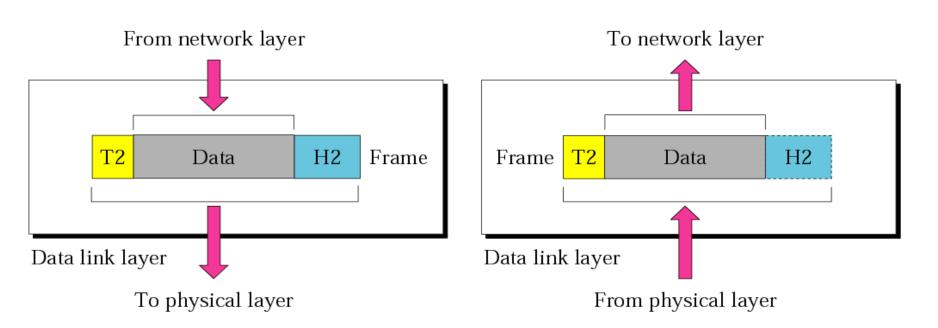
• The physical topology defines how devices are connected to make a network.

7. Transmission mode.

The physical layer also defines the direction of transmission between two devices: simplex, half-duplex, or full-duplex.

2. Data Link Layer

- The data link layer transforms the physical layer, a raw transmission facility, to a reliable link.
- It makes the physical layer appear error-free to the upper layer (network layer).
- ✓ The following figure shows the relationship of the **data link layer** to the **network and physical layers.**



• The data link layer is responsible for moving frames from one hop (node) to the next.

Layer 2 frame structure

Start Frame	He	eader		Trailer	Stop Frame
(Flag)	۸ ما ماريم م م	T /I	Data	FCC	(Flag)
(13.6)	Address	Type/Length		FCS	(13.6)

Data Link Layer Responsibilities

1. Framing

The data link layer divides the packets received from the network layer into manageable data units called frames.

2. Physical addressing

• If frames are to be distributed to different systems on the network, the data link layer adds a header to the frame to define the sender and/or receiver of the frame.

• If the frame is intended for a system outside the sender's network, the receiver address is the address of the device that connects the network to the next one.

3. Flow control

If the rate at which the data are absorbed by the receiver is less than

the **rate** at which **data** are produced in the **sender**,
the **data link layer** imposes a **flow control** mechanism
to avoid **overwhelming** the **receiver**.

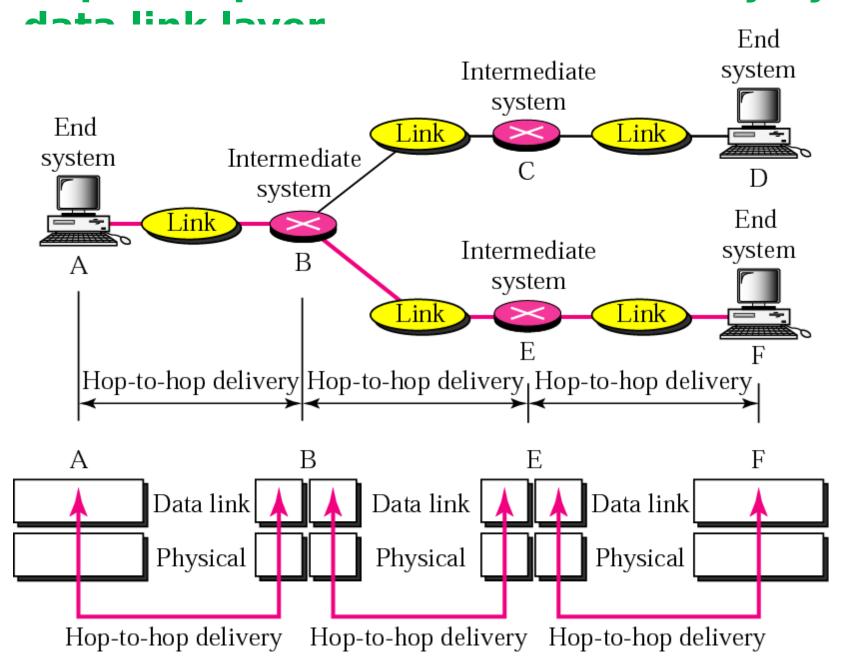
4. Error control

- The data link layer adds reliability to the physical layer by adding mechanisms to detect and retransmit damaged or lost frames.
- It also uses a mechanism to recognize duplicate frames.
- Error control is normally achieved through a **trailer** added to the end of the **frame**.

5. Access control

When two or more devices are connected to the same link, data link layer protocols are necessary to determine which device has control over the link at any given time.

nop-to-nop (node-to-node) delivery by



- As the figure above shows, communication at the data link layer occurs between two adjacent nodes.
- To send data from **A** to **F**, three partial deliveries are made.
- **First**, the **data link layer** at **A sends** a **frame** to the **data** link layer at **B** (a **router**).
- Second, the data link layer at B sends a new frame to the data link layer at E.
- Finally, the data link layer at E sends a new frame to the data link layer at F.

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- Note that the **frames** that are **exchanged** between the **three nodes** have **different values** in the **headers**.
- The **frame** from **A** to **B** has **B** as the **destination address** and **A** as the **source address**.
- The **frame** from **B** to **E** has **E** as the **destination address** and **B** as the **source address**.
- The **frame** from **E** to **F** has **F** as the **destination address** and **E** as the **source address**.
- The values of the trailers can also be different if error checking includes the header of the frame.
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Framing

- The data link layer, needs to pack bits into frames, so that each frame is distinguishable from another.
- Our postal system practices a type of framing.
- The simple act of inserting a letter into an envelope separates one piece of information from another; the envelope serves as the delimiter.
- In addition, each envelope defines the sender and receiver addresses since the postal system is a many-to-many carrier facility.

Cont----

- Framing in the data link layer separates a message from one source to a destination, or from other messages to other destinations, by adding a sender address and a destination address.
- The destination address defines where the packet is to go; the sender address helps the recipient acknowledge the receipt.
- NB: Addressing here is about the next node in the LAN

frame, that is not normally done.

Although the whole message could be packed in one

- One reason is that a frame can be very large, making flow and error control very inefficient.
- When a message is carried in one very large frame,
 even a single-bit error would require the
- retransmission of the whole message.
- When a message is divided into smaller frames, a singlebit error affects only that small frame.
- Frames can be of:
- Fixed size Framing
- Variable size Framing

1. Fixed-Size Framing

- In fixed-size framing, there is no need for defining the boundaries of the frames; the size itself can be used as a delimiter.
- An example of this type of framing is the ATM wide-area network, which uses frames of fixed size called cells.

2. Variable-Size Framing

- Variable-size framing is prevalent in local area networks.
- In variable-size framing, we need a way to define the end of the frame and the beginning of the next.
- Historically, two approaches were used for this purpose: a

character-oriented approach and a hit-oriented approach

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Structure of the Ethernet Frame

Preamble	SFD	Destination MAC Address		Length / Type	Encapsulated Data	FCS
7	1	6	6	2	46 to 1500	4

1. Unaracter-Orienteu Protocois

- In a **character-oriented protocol**, data to be carried are **8-bit characters** from a **coding system** such as **ASCII**.
- The header, which normally carries the source and destination addresses and other control information, and the trailer, which carries error detection or error correction redundant bits, are also multiples of 8 bits.

Character-Oriented Protocols----

- To **separate** one **frame** from the **next**, an **8-bit (1-byte) flag** is **added** at the **beginning** and the **end** of a **frame**.
- The flag, composed of protocol-dependent special characters, signals the start or end of a frame.
- Character-oriented framing was popular when only text was exchanged by the data link layers.
- ✓ The flag could be selected to be any character not used for text communication.
- Now, however, we send other types of information such as graphs, audio, and video.

Character-Oriented Protocols-----

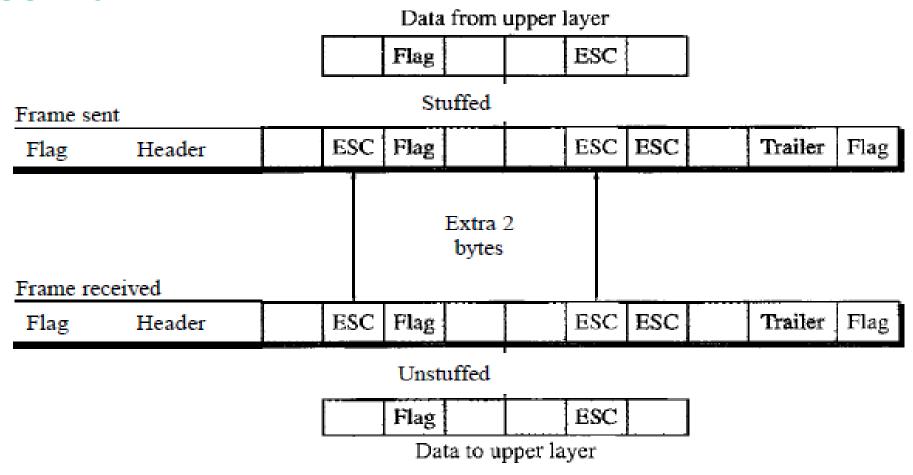
- Any pattern used for the flag could also be part of the information.
- ✓ If this happens, the **receiver**, when it **encounters** this **pattern** in the **middle** of the **data**, thinks it has reached the **end** of the **frame**.
- To fix this problem, a byte-stuffing strategy was added to character-oriented framing.
- In byte stuffing (or character stuffing),a special byte is added to the data section of the frame when
- there is a **character** with the **same pattern** as the **flag**.
- The data section is stuffed with an extra byte.

Character-Oriented Protocols----

- This byte is usually called the escape character (ESC), which has a predefined bit pattern.
- Whenever the receiver encounters the ESC character, it removes it from the data section and treats the next character as data, not a delimiting flag.
- Byte stuffing by the escape character allows the presence of the flag in the data section of the frame, but it creates another problem.
- What happens if the text contains one or more normal escape characters followed by a flag?

Character-Oriented Protocols----

- The receiver removes the escape character, but keeps the flag, which is incorrectly interpreted as the end of the frame.
- To **solve** this **problem**, the **escape characters** that are **part** of the **text** must also be marked by **another escape character**.
- In other words, if the **escape character** is **part** of the **text**, an **extra one** is added to show that the **second one** is part of the **text**.

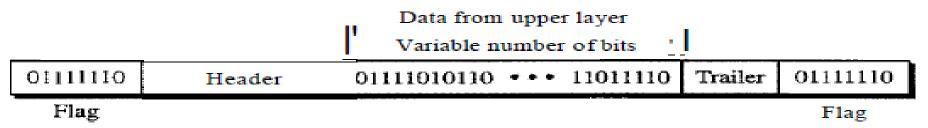


Byte stuffing is the process of adding 1 extra byte whenever there is a flag or escape character in the text.

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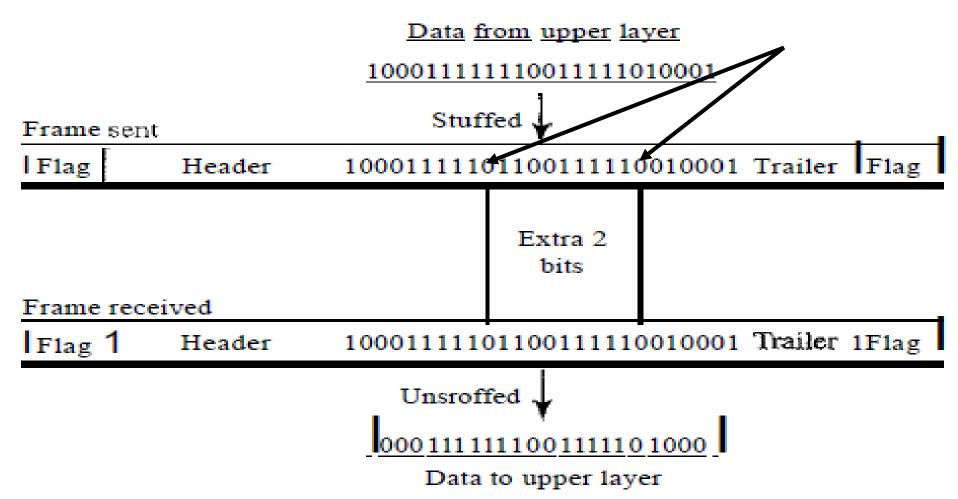
2. Bit-Oriented Protocols

- In a bit-oriented protocol, the data section of a frame is a sequence of bits to be interpreted by the upper layer as text, graphic, audio, video, and so on.
- However, in addition to headers (and possible trailers), we still need a delimiter to separate one frame from the other.
- Most protocols use a special 8-bit pattern flag 011111110 as the delimiter to define the beginning and the end of the frame, as shown in the figure.



- This flag can create the same type of problem we saw in the byte-oriented protocols.
- That is, if the flag pattern appears in the data, we need to somehow inform the receiver that this is not the end of the frame.
- We do this by stuffing 1 single bit (instead of 1 byte) to prevent the pattern from looking like a flag.
- ✓ The **strategy** is called **bit stuffing**.
- In bit stuffing, if a 0 and five consecutive 1 bits are encountered, an extra 0 is added.

- This extra stuffed bit is eventually removed from the data by the receiver.
- Note that the extra bit is added after one 0 followed by five 1s regardless of the value of the next bit.
- This guarantees that the flag field sequence does not inadvertently appear in the frame.



Bit stuffing is the process of adding one extra 0 whenever five consecutive 1s follow a 0 in the data, so that the receiver does not mistaken the pattern 01111110 for a flag.

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- The above figure shows bit stuffing at the sender and bit removal at the receiver.
- Note that even if we have a **0 after five 1s**, we still **stuff** a **0**.
- ✓ The **0** will be removed by the **receiver**.
- This means that if the **flag** like **pattern 011111110** appears in the **data**, it will change to **0111111010** (**stuffed**) and is **not mistaken** as a **flag** by the **receiver**.
- The **real flag 011111110** is **not stuffed** by the **sender** and is recognized by the **receiver** as a **flag**.

2. Flow Control

• Flow control **coordinates** the amount of data that can be sent before **receiving** an acknowledgment and is one of the most

important duties of the data link layer.

which to **store incoming data**.

the **sender** how much data it can **transmit** before it must wait for an **acknowledgment** from the **receiver**.

• In most protocols, flow control is a set of procedures that tells

- The flow of data must not be allowed to overwhelm the receiver.
- Any receiving device has a limited speed at which it can process incoming data and a limited amount of memory in

- The receiving device must be able to inform the sending device before those limits are reached and to request that the transmitting device send fewer frames or stop temporarily.
- Incoming data must be checked and processed before they can be used.
- The rate of such processing is often slower than the rate of transmission.

- For this reason, each receiving device has a block of memory, called a buffer, reserved for storing incoming data until they are processed.
- If the buffer begins to fill up, the receiver must be able to tell the sender to halt transmission until it is once again able to receive.

3. Media Access control

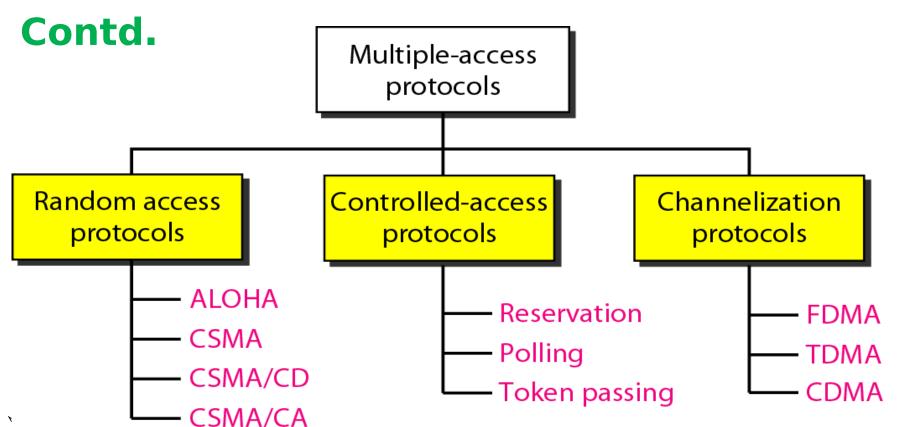
- The **data link layer** can further be divided in to **two layers**:
- ✓ The upper sub-layer that is responsible for flow and error control is called the logical link control (LLC) layer;
- ✓ The lower sub-layer that is mostly responsible for multiple access resolution is called the media access control (MAC) layer
- When nodes or stations are connected and use a common link, called a multipoint or broadcast link,

we need a **multiple-access protocol** to coordinate access to the link.

3. Media Access control-----

- The **problem** of **controlling** the **access** to the medium is similar to the rules of **speaking in an assembly.**
- The procedures guarantee that the right to speak is upheld and ensure that two people do not speak at the same time,

do not interrupt each other, do not monopolize the discussion, and so on.



read about channelization protocois:

Frequency Division Multiple Access (FDMA), Time Division
 Multiple Access (TDMA), and Code Division Multiple Access
 (CDMA)

3.1 Random Access

- In random access or contention methods, no station is superior to another station and none is assigned the control over another.
- No station permits, or does not permit, another station to send.
- At each instance, a station that has data to send uses a procedure defined by the protocol to make a decision on whether or not to send.
- This decision depends on the state of the medium (idle or busy).

questions.

- In other words, each station can **transmit** when it **desires** on the condition that it **follows** the **predefined procedure**, including the **testing** of the **state** of the **medium**.
- In a random access method, each station has the right to the **medium** without being **controlled** by any other **station**.
- However, if more than one station tries to send, there is an access **conflict-collision-**and the **frames** will be either destroyed or modified.
- To avoid access conflict or to resolve it when it happens, each **station** follows a **procedure** that answers the **following**

- When can the station access the medium?
- What can the station do if the medium is busy?
- How can the station determine the success or failure of the transmission?
- What can the station do if there is an access conflict?
- The **random access** methods have evolved from a very interesting **protocol** known as **ALOHA**, which used a very simple procedure called **multiple access** (**MA**).
- The **method** was **improved** with the **addition** of a **procedure** that forces the **station** to **sense** the **medium** before transmitting.

- This was called carrier sense multiple access.
- This method later evolved into two parallel methods:
- ✓ Carrier Sense Multiple Access with Collision

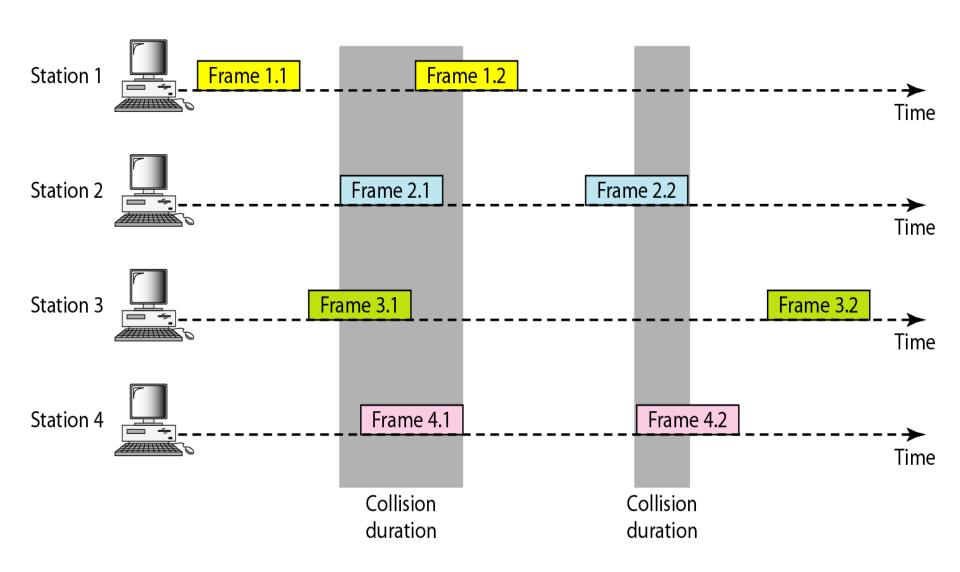
 Detection (CSMA/CD) and
- ✓ Carrier Sense Multiple Access with Collision Avoidance (CSMA/CA).
- CSMA/CD tells the station what to do when a collision is detected.
- **CSMA/CA** tries to **avoid** the **collision**.

Pure ALOHA

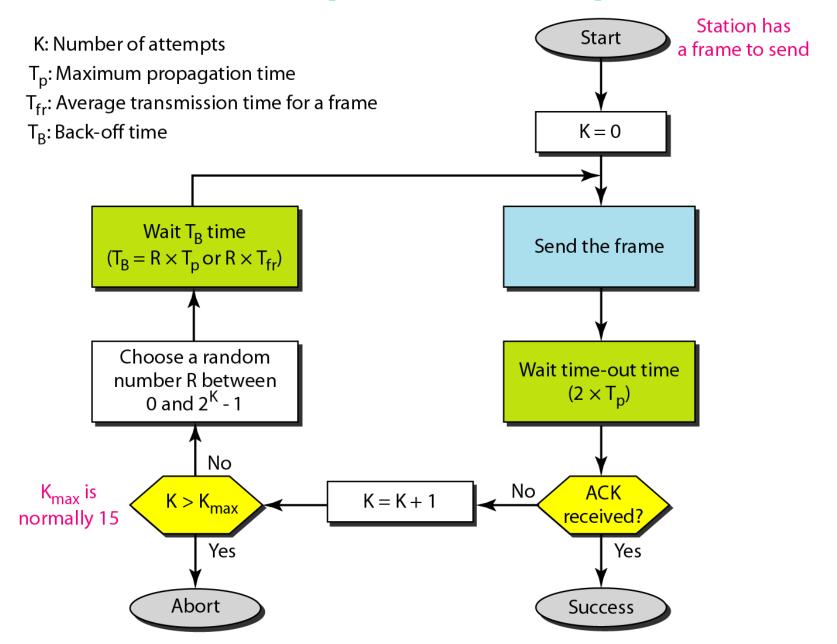
- ALOHA is the simplest technique in multiple accesses.
- Basic idea of this mechanism is a user can transmit the data whenever they want.
- If data is successfully transmitted then there isn't any problem.
- But if collision occurs then the station will transmit again.
- Sender can detect the collision if it doesn't receive the

adrnowladgement from the receiver

Procedure for pure ALOHA protocol



Procedure for pure ALOHA protocol



Carrier Sense Multiple Access (CSMA)

- Protocols that listen for a carrier and act accordingly are called carrier sense protocols.
- Carrier sensing allows the station to detect whether the medium is currently being used.
- Schemes that use a carrier sense circuits are classed together as carrier sense multiple access or CSMA schemes.
- There are two variants of CSMA. CSMA/CD and CSMA/CA
- The simplest CSMA scheme is for a station to sense the medium, sending packets immediately if the medium is idle.
- If the station waits for the medium to become idle it is called persistent otherwise it is called non persistent.

Persistent

- Persistent:-wait if busy and transmit only when the media becomes idle again (not transmission after a triggered timer expire)
- When a station has the data to send, it first listens the channel to **check** if **anyone** else is transmitting data or not.
 - If it senses the channel **idle**, station **starts transmitting** the **data**.
- If it senses the channel busy it waits until the channel is idle, by continuously sensing the channel.

Non-Persistent

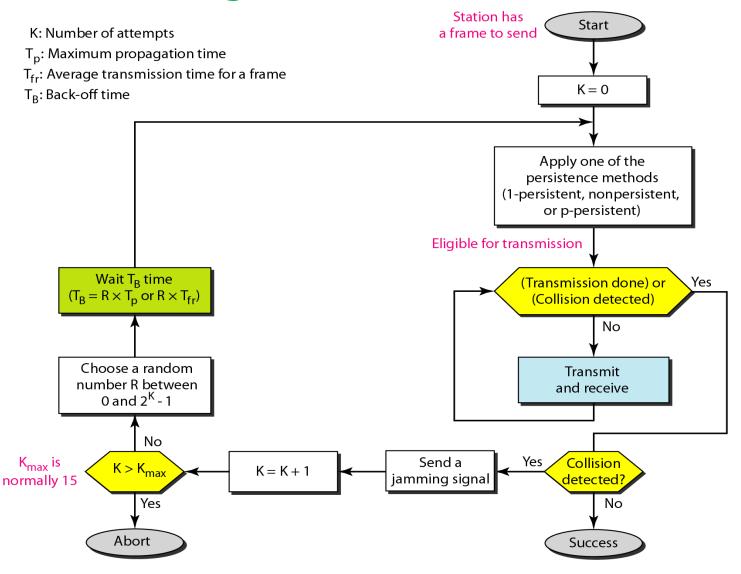
- Non persistent CSMA is less aggressive compared to persistent protocol.
- In this protocol, before sending the data, the station senses the channel and if the channel is idle it starts transmitting the data.
- But if the channel is busy, the station does not continuously sense it but instead of that it waits for random amount of time and repeats the algorithm.
- Here the algorithm leads to better channel utilization but also results in longer delay compared to persistent.
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Carrier Sense Multiple Access/Collision Detection (CSMA/CD)

- CSMA/CD is a technique for multiple access protocols.
- ✓ If **no transmission** is taking place at the time, the particular station can transmit.
- If two stations attempt to transmit simultaneously, this causes a collision, which is detected by all participating stations.
- After a **random time interval**, the **stations** that **collided** attempt to transmit again.
- If another collision occurs, the time intervals from which the random waiting time is selected are increased step by step.

✓ This is known as exponential back off.

Flow diagram for the CSMA/CD



Reading Assignment: Read About CSMA/CA and understand the difference with CSMA/CD

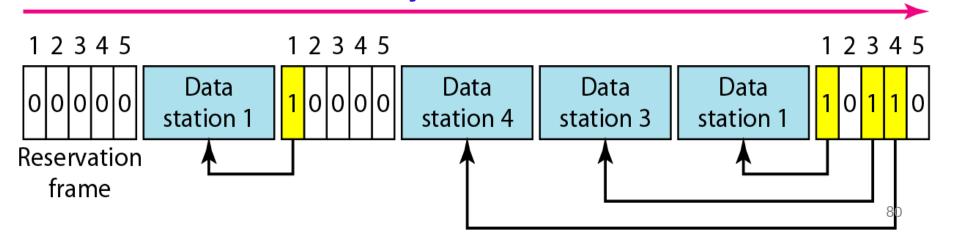
3.2 Controlled access

- In controlled access, the stations consult one another to find which station has the right to send.
- A station cannot send unless it has been authorized by other stations.

Reservation

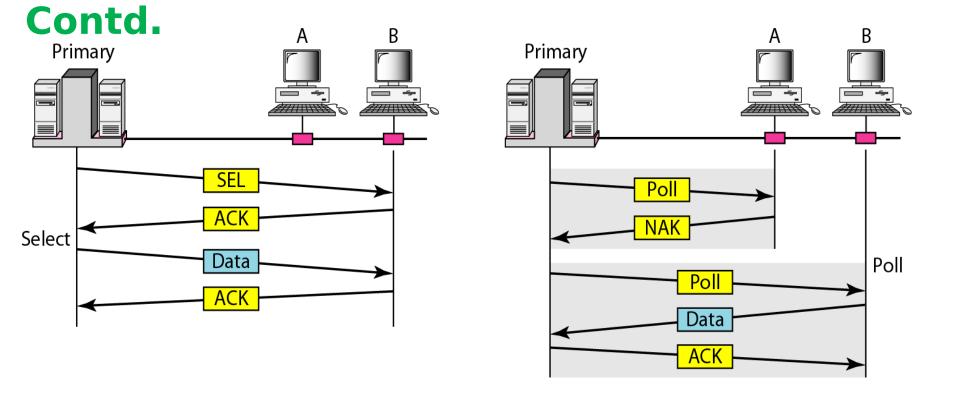
- In the **reservation method**, a **station** needs to make a **reservation** before **sending data**.
- **Time** is divided into **intervals**.
- In each interval, a reservation frame precedes the data frames sent in that interval.
- If there are *N* stations in the system, there are exactly *N* reservation minis lots in the reservation frame.
- Each mini slot belongs to a station.
- When a station needs to send a data frame, it makes a reservation in its own minis lot.

- The stations that have made reservations can send their data frames after the reservation frame.
- The following figure shows a situation with five stations and a five-minis lot reservation frame.
- In the first interval, only stations 1, 3, and 4 have made reservations.
- In the second interval, only station 1 has made a reservation.



Polling

- Polling works with topologies in which one device is designated as a primary station and the other devices are secondary stations.
- All data exchanges must be made through the primary device even when the ultimate destination is a secondary device.
- The primary device controls the link; the secondary devices follow its instructions.
- It is up to the **primary device** to determine which **device** is allowed to use the **channel** at a given **time**.
- The primary device, therefore, is always the initiator of a session.
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- If the primary wants to receive data, it asks the secondary devices if they have anything to send; this is called poll function.
- If the **primary** wants to **send data**, it tells the **secondary** to get **ready** to **receive**; this is called **select function**.

Token Passing

- In the token-passing method, the stations in a network are organized in a logical ring.
- In other words, for each station, there is a predecessor and a *successor*.
- The predecessor is the station which is logically before the **station** in the ring;
- the **successor** is the **station** which is **after** the **station** in
- the **ring**.
- The current station is the one that is accessing the **channel** now.

- The right to this access has been passed from the predecessor to the current station.
- The right will be passed to the successor when the current station has no more data to send.
- But how is the right to access the channel passed from one station to another?
- In this method, a special packet called a token circulates through the ring.
- The possession of the token gives the station the right to access the channel and send its data.

- When a station has some data to send, it waits until it receives the token from its predecessor.
- It then holds the token and sends its data.
- When the station has no more data to send, it releases the token, passing it to the next logical station in the ring.
- The station cannot send data until it receives the token again in the next round.

3. Network Layer (3rd OSI Layer)

- Concerned with getting packets from source to destination.
- The network layer must know the topology of the subnet and choose appropriate paths through it.
- When source and destination are in different networks, the network layer must deal with these differences.
- The network layer is responsible for the source-to-destination delivery of a packet, possibly across multiple networks.

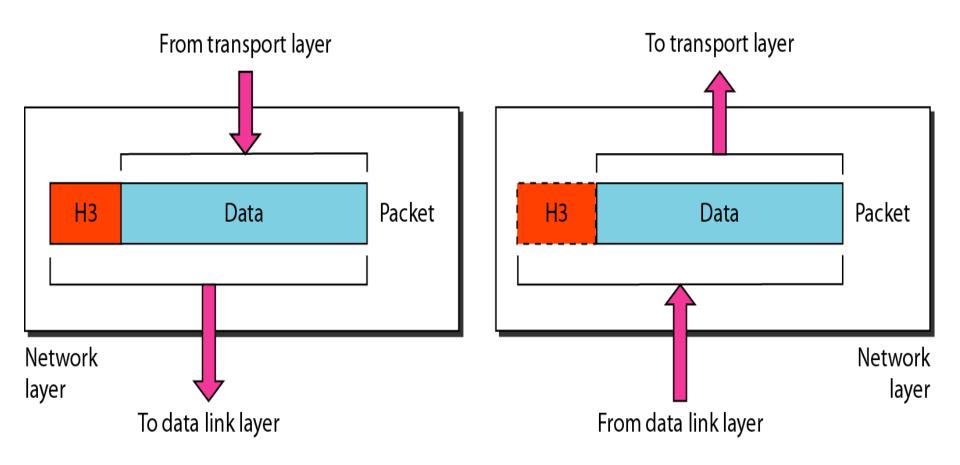
3. Network Layer (3rd OSI Layer)-----

 Whereas the data link layer oversees the delivery of the packet between two

systems on the same network, the network layer ensures that

- point of origin to its final destination.
- If two systems are connected to the same local network, there is usually no need for a network layer.
- However, if the two systems are attached to different networks with connecting devices between the networks, there is often a need for the network layer to accomplish

each **packet** gets from its



• The network layer is responsible for the delivery of individual packets from the source host to the destination host.

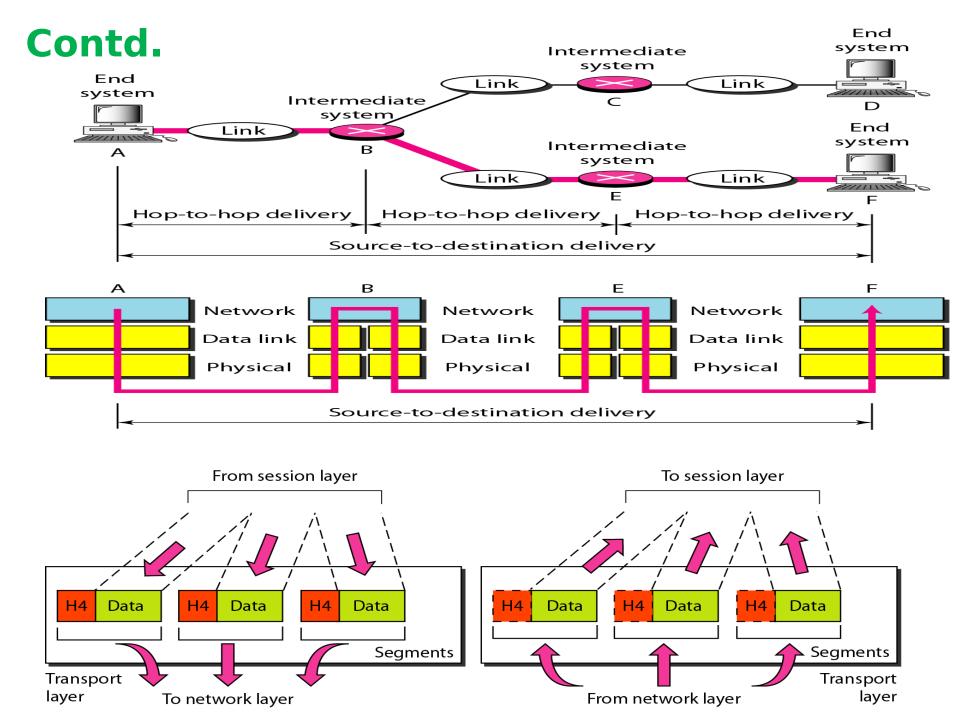
Other responsibilities of the network layer include the following:

1. Logical addressing:-

- The physical addressing implemented by the data link layer handles the addressing problem locally.
- If a packet passes the network boundary, we need another addressing system to help distinguish the source and destination systems.
- The network layer adds a header to the packet coming from the upper layer that, among other things, includes the logical addresses of the sender and receiver.

2. Routing:-

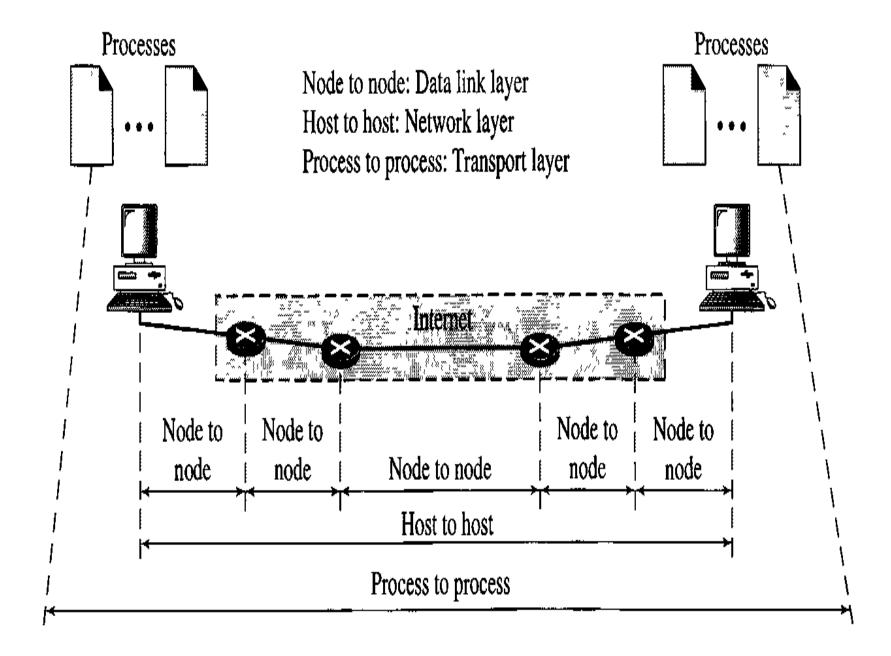
- When independent networks or links are connected to create internetworks (network of networks) or a large network, the connecting devices (called routers or switches) route or switch the packets to their final destination.
- ✓ One of the **functions** of the **network layer** is to provide this mechanism.



4. Transport layer (4th OSI layer)

- The **transport layer** is responsible for **process-to-process delivery** of the entire message.
- A process is an application program running on a host.
- Whereas the network layer oversees source-to-destination delivery of individual packets, it does not recognize any **relationship** between those **packets**.
- The **network layer** treats each **packet independently**, as though each piece belonged to a **separate message**, whether or **not** it does.
- The **transport layer**, on the other hand, **ensures** that the whole message arrives intact and in order, overseeing both error **control** and **flow control** at the source-to-destination level.

- The **Transport layer** encompasses these **functions**:
- 1. Enables multiple applications to communicate over the network at the same time on a single device
- 2. Ensures that, if required, all the **data** is **received reliably** and in order by the **correct application**
- 3. Employs error handling mechanisms



1. Service-point addressing

- **Computers** often **run** several **programs** at the **same time**.
- For this reason, source-to-destination delivery means delivery not only from one computer to the next but also from a specific process

(running program) on one computer to a specific

process on the **other**.

The transport layer header must therefore include a type of address called a service-point address (or port address).

Responsibilities Transport Layer-----

The network layer gets each packet to the correct computer; the transport layer gets the entire message to the correct process on that computer.

2. Segmentation and reassembly

- A message is divided into transmittable segments, with each segment containing a sequence number.
- These numbers enable the transport layer to reassemble the message correctly upon arriving at the destination and to identify and replace packets that were lost in transmission.

3. Flow control

- Like the data link layer, the transport layer is responsible for flow control.
- However, flow control at this layer is performed end to end rather than across a single link.

4. Error control

- Like the data link layer, the transport layer is responsible for error control.
- However, error control at this layer is performed process-toprocess rather than across a single link.

- The sending transport layer makes sure that the entire message arrives at the receiving transport layer without error (damage, loss, or duplication).
- Error correction is usually achieved through retransmission.

5. Connection control

The transport layer can be either connectionless or connection-oriented.

A. Connectionless

Transport layer treats each **segment** as an **independent packet** and **delivers** it to the **transport layer** at the destination machine.

A. Connection-oriented

- **Transport layer** makes a **connection** with the **transport** layer at the destination machine first before delivering the **packets**.
- After all the data are transferred, the **connection** is terminated.

1. Process-to-Process Delivery

- As a revision, the data link layer is responsible for delivery of frames between two neighboring nodes over a link.
- ✓ This is called **node-to-node delivery.**
- The network layer is responsible for delivery of packets between two hosts.
- ✓ This is called **host-to-host delivery.**
- Communication on the Internet is not defined as the exchange of data between two nodes or between two hosts.
- Real communication takes place between two processes
 (application programs).

1. Process-to-Process Delivery---

- We need process-to-process delivery.
- However, at any moment, several processes may be running on the source host and several on the destination host.
- To complete the delivery, we need a mechanism to deliver data from one of these processes running on the

process running on the destination host.

source host to the corresponding

- ✓ The transport layer is responsible for process-to-process delivery-the delivery of a packet, part of a message, from one process to another.
- **Two processes** communicate in a **client/server relationship.**

2. Transport layer addressing

- Whenever we need to deliver something to one specific destination among many, we need an address.
- At the data link layer, we need a MAC address to choose one node among several nodes if the connection is not point-topoint.
- A frame in the data link layer needs a destination MAC address for delivery and a source address for the next node's reply.
- At the network layer, we need an IP address to choose one host among millions.

- A datagram in the network layer needs a destination IP address for delivery and a source IP address for the destination's reply.
- At the transport layer, we need a transport layer address, called a port number, to choose among multiple processes running on the destination host.
- **The destination port number** is needed for **delivery**; the **source port number** is needed for the **reply**.
- In the **Internet model**, the **port numbers** are **16-bit** integers between 0 and 65,535.

- The client program defines itself with a port number, chosen randomly by the transport layer software running on the client host.
- **✓** This is the **ephemeral (temporal) port number**.

3. Identifying Applications (Processes)

- In order to pass data streams to the proper applications, the
 Transport layer must identify the target application.
- To accomplish this, the **Transport layer assigns** an **application identifier** called a **port number**.
- Each software process that needs to access the network is assigned a port number unique in that host
- This **port number** is used in the **transport layer header** to indicate to which **application** that **piece** of **data** is associated.
- The **server process** must also define itself with a **port number**.
- This **port number**, however, **cannot be chosen randomly**.

If the computer at the server site runs a server process and assigns a random number as the port number,

the **process** at the **client site** that wants to **access** that **server** and use its **services will not know the port number.**

- Of course, one solution would be to send a special packet and request the port number of a specific server, but this requires more overhead.
- The **Internet** has decided to use **universal port numbers** for **servers**; these are called **well-known port numbers**.
- Example of well known port numbers: 21 for FTP, 23 telnet, 25
 SMTP, 80 HTTP, etc

Internet Assigned Number Authority (IANA) Ranges

- There are some **exceptions** to this **rule**; for **example**, there are **clients** that are **assigned well-known port numbers.**
- Every **client process** knows the well-known **port number** of the corresponding **server process**.

1. Well-known ports.

✓ The **ports** ranging from **0 to 1023** are assigned and controlled by **IANA**. These are the **well-known ports**.

2. Registered ports.

✓ The **ports** ranging from **1024 to 49,151** are **not assigned** or controlled by **IANA**.

Internet Assigned Number Authority (IANA) Ranges

They can only be **registered** with **lANA** to **prevent duplication**.

3. Dynamic ports.

- The ports ranging from 49,152 to 65,535 are neither controlled nor registered.
- ✓ They can be used by **any process**.
- ✓ These are the **ephemeral ports** (temporary ports).

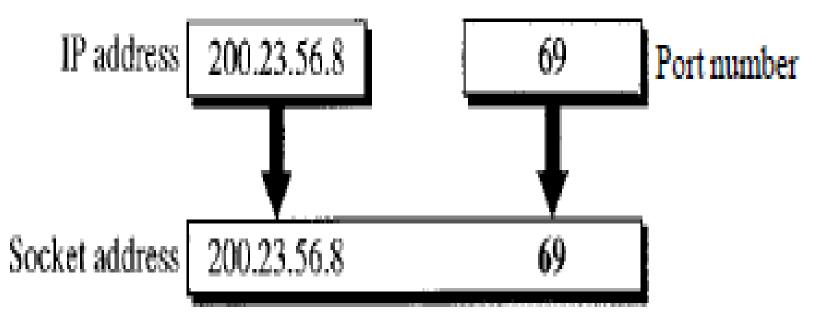
Socket Addresses

- Process-to-process delivery needs two identifiers, IP address and the port number, at each end to make a connection.
- The combination of an IP address and a port number is called a socket address.
- ✓ The client socket address defines the client process uniquely just as the server socket address defines the server process uniquely.
- A transport layer protocol needs a pair of socket addresses: the client socket address and the server socket address.

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Socket Addresses

- These four pieces of information are part of the IP header and the transport layer protocol header.
- The **IP header** contains the **IP addresses**; the **UDP** or **TCP header** contains the **port numbers**.



Transport Layer Protocols

A transport layer protocol can either be connectionless or connection-oriented.

1. Connectionless Service

- In a connectionless service, the packets are sent from one party to another with **no need for connection establishment** or **connection release.**
- The packets are not numbered; they may be delayed or lost or may arrive out of sequence.
- ✓ There is **no acknowledgment** either.
- **UDP** is connectionless.

2. Connection-Oriented Service

- a. In a connection-oriented service, a connection is first established between the sender and the receiver.
- **b.** Data are transferred. At the end, the connection is released.
- c. TCP and SCTP are connection-oriented protocols.

Reliable Versus Unreliable

- The **transport layer** service can be **reliable** or **unreliable**.
- If the application layer program needs reliability, we use a reliable transport layer protocol by implementing flow and error control at the transport layer.
- ✓ This means a slower and more complex service.
- On the other hand, if the application program does not need
 reliability because

it uses its own flow and error control mechanism or

it needs fast service or the nature of the service does not

demand flow and error control (real-time applications),

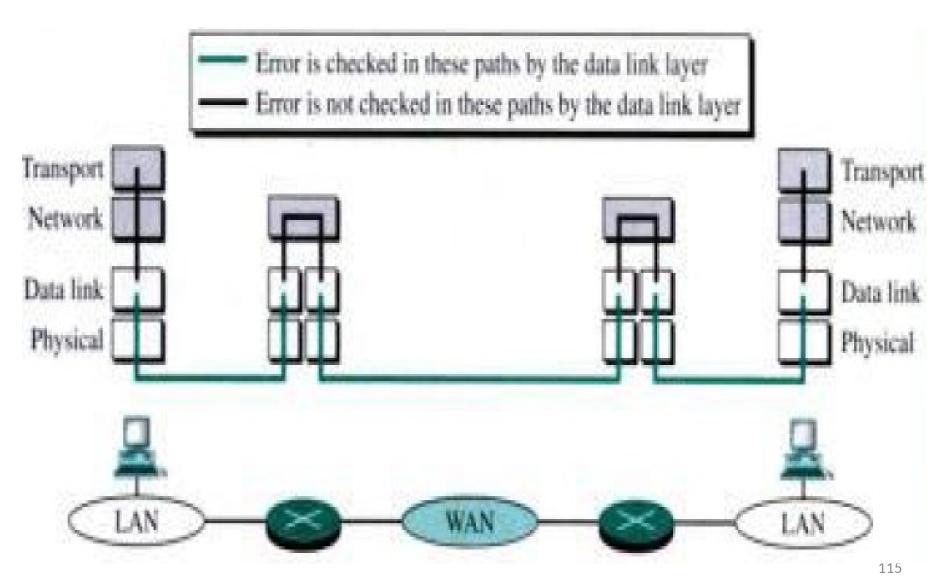
then an unreliable protocol can be used

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Reliable Versus Unreliable

- In the **Internet**, there are **two common** different **transport layer protocols**, as we have already mentioned.
- UDP is connectionless and unreliable;
 TCP and SCTP are connection-oriented and reliable protocols.
- ✓ These three can respond to the demands of the **application layer programs**.

Why flow control and error control at the transport layer as we have it at the data link layer???



USER DATAGRAM PROTOCOL (UDP)

- The **User Datagram Protocol (UDP)** is called a **connectionless, unreliable transport protocol.**
- It does **not add** anything to the **services** of **IP** except to provide **process-to-process communication**.
- Also, it performs very limited error checking. If UDP is so powerless, why would a process want to use it?
- ➤ With the disadvantages come some advantages. **UDP** is a **very simple protocol** using a **minimum** of **overhead**.
- If a process wants to send a small message and does not care much about reliability, it can use UDP.
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- Sending a small message by using UDP takes much less interaction between the sender and receiver than using TCP or SCTP.
- UDP packets, called user datagrams, have a fixed-size header of 8 bytes

1. Source port number.

✓ This is the port number used by the **process running** on the **source host**.

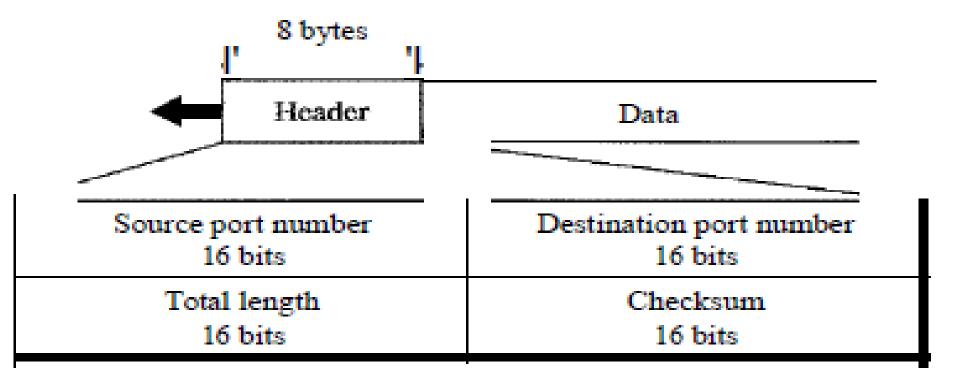
2. Destination port number.

✓ Used by the **process running** on the **destination host**. 117

Cont----

3. Length

• This is a 16-bit field that defines the total length of the user datagram, header plus data.



Contd.

- **UDP** does **not** perform:
- Flow control, Error control and connection control
- ✓ All these **functions** are done by the **processes** (**application layer programs**) using **UDP**
- UDP is not capable of segmenting and reassembling frames and does not implement sequence numbers
- ► But **UDP**, like **TCP**, performs:
 - Service point addressing
- UDP can transmit only small portions of data at a time because it is not capable of segmenting and reassembling

frames and does not implement sequence numbers

Use of UDP

- 1. UDP is suitable for a process that requires simple request-response communication with little concern for flow and error control.
- It is not usually used for a process such as FTP that needs to send bulk data
- 2. **UDP** is suitable for a process **with internal flow and error control** mechanisms.
- For example, the Trivial File Transfer Protocol (TFTP) process includes flow and error control.
- It can easily use UDP.

Use of UDP-----

- 3. **UDP** is a suitable **transport protocol** for **multicasting**.
- Multicasting capability is embedded in the UDP software but not in the TCP software.
- 4. **UDP** is used for **management processes** such as **SNMP**.
- 5. **UDP** is used for some **route updating protocols** such as **Routing Information Protocol (RIP).**

Transmission Control Protocol (TCP)

- **TCP**, like **UDP**, is a **process-to-process** (program-toprogram) protocol.
- **TCP**, therefore, like **UDP**, uses **port numbers**.

services of IP

- Unlike UDP, TCP is a connection-oriented protocol; it creates a **virtual connection** between **two TCPs** to **send data**.
- In addition, TCP uses flow and error control mechanisms at the **transport level**.
- In brief, **TCP** is called a **connection-oriented**, **reliable** transport protocol.
- ✓ It **adds connection-oriented** and **reliability features** to the

TCP Services

- 1. Process-to-Process Communication:-
- Like UDP, TCP provides process-to-process communication using port numbers.
- 2. Stream Delivery Service
- 3. Full-Duplex Communication
- 4. Connection-Oriented Service
- 5. Reliable Service

5. Session Layer

■ The lowest four layers of the OSI model (Physical, Data link,

- Network, and Transport) provide the means for the reliable exchange of data and provide a fast data service.
 For example, a remote terminal access application might
- A transaction-processing application might require checkpoints in the data-transfer stream to permit backup and recovery.

require a half-duplex dialogue.

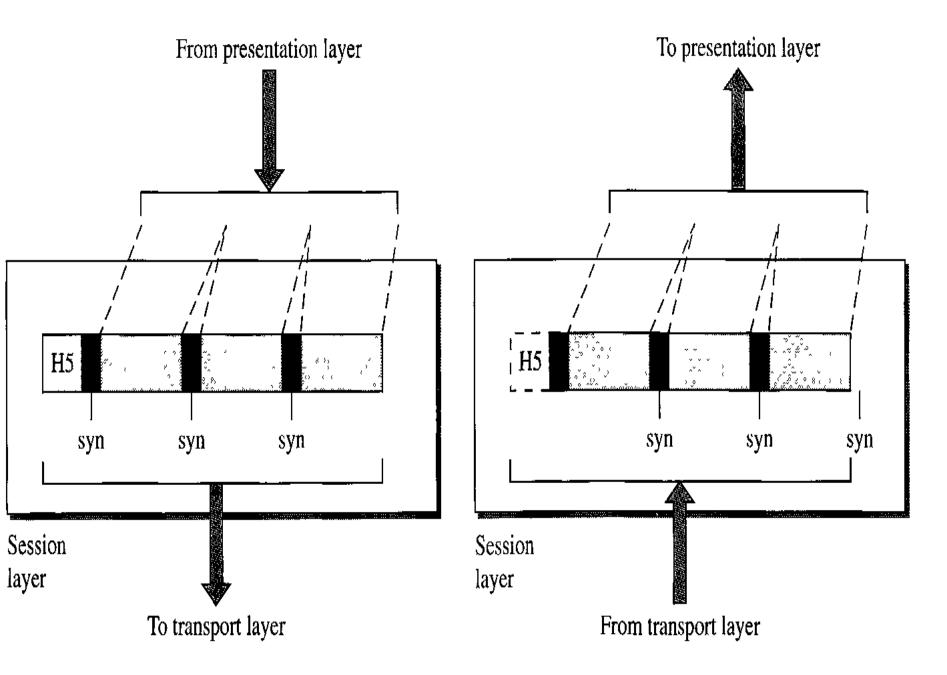
A message processing application might require the ability to interrupt a dialogue in order to prepare a new portion of a message and later to resume the dialogue where it was left off.

Cont----

- All these capabilities could be embedded in specific applications at layer 7.
- However, because these **types** of **dialogue-structuring** tools have **widespread applicability**, it makes sense to organize them into a **separate layer**: the **session layer**.
- The session layer provides the mechanism for controlling the dialogue between applications in end systems.
- ✓ In many cases, there will be **little** or **no need** for **session-layer services**,

but for some **applications**, such **services** are used.

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Key Services provided by the Session Layer

1. Dialogue discipline.

the next **denartment**

■ This can be **two-way simultaneous (full duplex)** or two way alternate **(half duplex) communication** between **processes**.

2. Grouping.

- The flow of data can be **marked** to **define** groups of **data**.
- For example, if a **retail store** is **transmitting sales data** to a **regional office**, the **data** can be **marked** to **indicate** the **end** of the **sales data** for **each department**;

this would signal the host computer to finalize running

totals for that department and start new running counts for

Key Services provided by the Session Layer

3. Recovery.

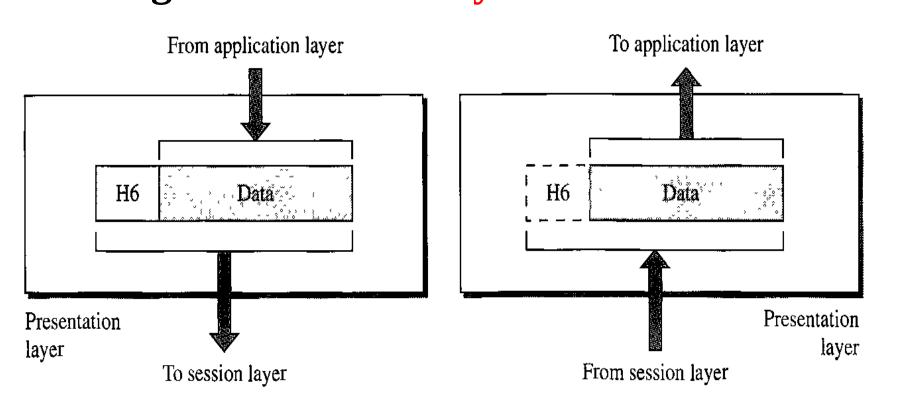
The **session layer** can provide a **check pointing** mechanism,

so that if a **failure** of some **sort occurs** between **checkpoints**, the **session entity** can **retransmit** all **data** since the last checkpoint.

• For example, if a **system** is **sending** a **file** of **2000 pages**, it is **advisable** to **insert checkpoints** after every **100** pages to ensure that each 100-page unit is received and acknowledged independently.

6. Presentation Layer

The presentation layer is concerned with the syntax and semantics of the information exchanged between two systems.



Specific responsibilities of the presentation layer:

1. Translation

- The processes (running programs) in two systems are usually exchanging information in the form of character strings, numbers, and so on.
- The information must be changed to bit streams before being transmitted.
- ✓ Because different computers use different encoding systems, the presentation layer is responsible for interoperability between these different encoding methods.
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Specific responsibilities of the presentation layer: The presentation layer at the sender changes the

common format.

ensure privacy.

message out over the network.

The presentation layer at the receiving machine changes the common format into its receiver-dependent format.
 Encryption

■ To carry sensitive information, a system must be able to

information from its **sender-dependent format** into a

Encryption means that the **sender transforms** the **original information** to **another form** and **sends the resulting**

Specific responsibilities of the presentation layer:

Decryption reverses the original process to transform the message back to its original form.

3. Compression

- **Data compression** reduces the **number** of **bits** contained in the **information**.
- Data compression becomes particularly important in the transmission of multimedia such as text, audio, and video.

7. Application Layer

- The application layer enables the user, whether human or software, to access the network.
- It provides user interfaces and support for services such as electronic mail, remote file access and transfer, shared database management, and other types of distributed information services.
- Application layer is where users actually communicate to the computer.
 - Take the case of **Internet Explorer (IE).**
- It is also responsible for identifying and establishing the

availability of the intended communication partner

- > Typical **application** layer **protocols**
 - Hyper Text Transfer Protocol (HTTP)
 - File Transfer Protocol (FTP)
 - E-mail (SMTP,POP,IMAP)
 - Domain Name System (DNS)

1. Domain Name System (DNS)

- Thousands of servers, installed in many different locations, provide the services we use over the Internet.
- Each of these servers is assigned a unique IP address
- It would be **impossible** to **remember** all of the **IP**

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DNS continued...

- DNS provides a way for hosts to use this name to request the IP address of a specific server.
 - ✓ DNS names are registered and organized on the Internet within specific high level groups, or domains.
 - ✓ Some of the **most** common **high level domains** on the **Internet** are **.com**, **.edu**, and **.net**
- A DNS server contains a table that associates hostnames in a domain with corresponding IP addresses

- ► Web client and web server
- A web client first receives the IP address of a web server from DNS server
- Then the client browser uses that IP address and port 80 to request
 web services
- This request is sent to the server using the Hypertext TransferProtocol (HTTP)
- The information content of a web page is encoded using specialized 'mark-up' languages.
 - ✓ E.g. HTML (Hypertext Mark-up Language)
- Many different web servers and web clients from many different manufactures work together seamlessly because of HTTP₁₃₆and

2. File Transfer Protocol (FTP)

- FTP is another common service used across the Internet that allows users to transfer files
- A host running FTP client software can access an FTP server to perform
 various file management functions including file uploads and downloads
- FTP service uses two different ports to communicate between client and server
 - ✓ Requests to begin an FTP session are sent to the server using destination port 21.
 - ✓ Once the **session** is **opened**, the **server** will change to port **20** to **transfer** the **data files**
- FTP client software is built into computer operating systems and into most web browsers

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3. E-mail

- Each mail server receives and stores mail for users who have mailboxes configured on the mail server
- Each user with a mailbox must then use an email client to access the mail server and read these messages
 - ✓ Mailboxes are identified by the format: user@company.domain
- Three application protocols used in processing email include:

- Simple Mail Transfer Protocol (SMTP):
- ✓ To **send** mail from **client** to **server** or **server** to **server**
- Post Office Protocol (POP3):
- ✓ To **download email** from **server** to **client**, and the **server deletes** the **mail**
- Internet Message Access Protocol (IMAP4):
- ✓ To **download email** from **server** to **client**, and the **server** does **not delete** (**keeps**) the **mail**

Activity

- 1. Which one of the following **is not correct** about POP3 and SMTP?
 - A. A sender can send email using SMTP
 - B. Both are application layer protocols
 - C. Both are Network Layer protocols
 - D. A sender can receive email using POP3
- 2. Which type of network is the largest as compared to the rest?
 - A. Metropolitan Area Network
 - B. The Internet
 - C. Local Area Network
 - D. Wide Area Network

Activity-----

3. A university owns a number of private local area and wide area networks, which are designed for an access by its academic staff, students and administrative workers. Which term best describes the University's network?

- A. Wide Area Network
- B. Metropolitan Area Network
- C. The Internet
- D. Local Area Network
- 4. Which layer uses port number to identify applications?
 - A. Application layer

C. Network Layer

B. Physical layer

D. Transport Layer

Activity-----

- 5. At which layer of the OSI model devices such as bridges, switches and Network Interface Cards (NICs) are used?
 - A. Application layer
 - B. Physical layer
 - C. Network layer
 - D. Data link layer
- 6. Write the difference and similarities between UDP and TCP transport Layer Protocols?

2. TCP/IP Reference Model

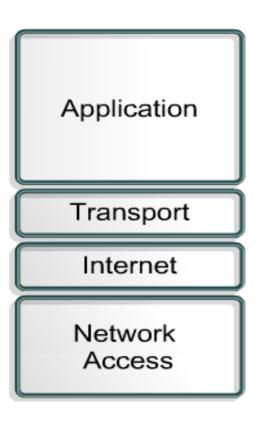
- TCP/IP Reference Model is a four-layered suite of communication protocols.
- It is named after the two main protocols that are used in the model, namely, TCP TCP/IP stands for Transmission Control

Protocol / Internet Protocol

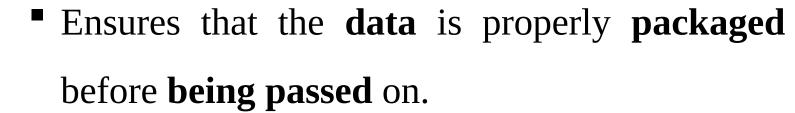
- It was developed by the **DoD** (**Department of Defence**) in the **1960s** as a **model** able to **withstand** intense **military attack** and **not fail**.
- **Data transmission** was possible to any **destination** on the **network** under any circumstances.
- Standardized in 1981, the TCP/IP model is now the standard on which the Internet is based.

2. TCP/IP Reference Model-----

- > It has **four layers**
- There are similarities and differences between the TCP/IP model and the seven layer OSI model.



1. TCP/IP Application Layer



- Handles high-level protocols, representation, encoding, and dialog control.
- There are **lots** of **protocols** defined at this layer:
 - ✓ **Simple Network Management Protocol (SNMP)** allows network managers to manage configurations, statistics, performance, and security. 145
 - **Domain Name System (DNS)** used to **translate domain names** into **IP addresses**. 145



1. TCP/IP Application Layer-----

• Has protocols to support file transfer, e-mail, and remote login.



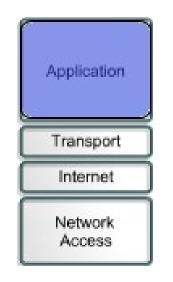
- 1. Trivial File Transfer Protocol (TFTP)
- Unreliable, connectionless User Datagram Protocol (UDP) service used to transfer configuration files, Cisco IOS images, and to transfer files in a LAN.

2. File Transfer Protocol (FTP)

• Reliable, connection-oriented service that uses TCP to transfer files between systems

3. Network File System (NFS)

Allows file access to a remote storage device such as a hard disk₁₄₆



1. TCP/IP Application Layer-----

►E-mail:

- 1. Simple Mail Transfer Protocol (SMTP)
- Administers the transmission of plain text e-mail between email servers.

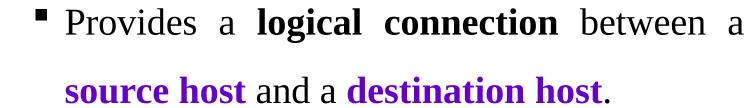
2. Post Office Protocol (POP3) and IMAP4

• Handles email transfer between a mail server and client machine

2. Remote Access:

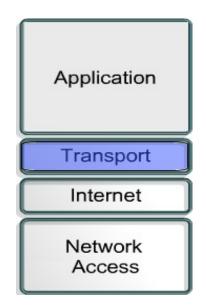
- Telnet
- ✓ A protocol that remotely access a computer, enabling a user to log into an Internet host and execute commands.
- ✓ A **Telnet client** is called a **local host**. A **Telnet server** is called a **remote host**.





- Transport Layer protocols segment and reassemble data sent by applications, into the same data stream, between end points.
- Provides end-to-end control and reliability as data travels through the cloud, accomplished through:

✓ sequence numbers, acknowledgments and sliding windows.



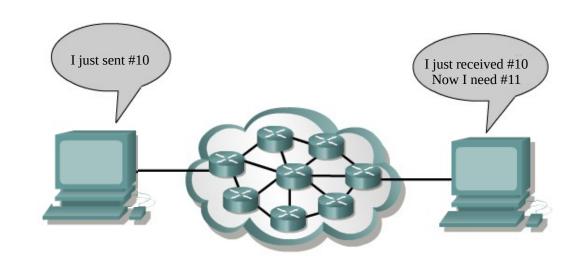
Transport

Application

Transport

Internet

Network Access



This shows sequence numbers and acknowledgements.

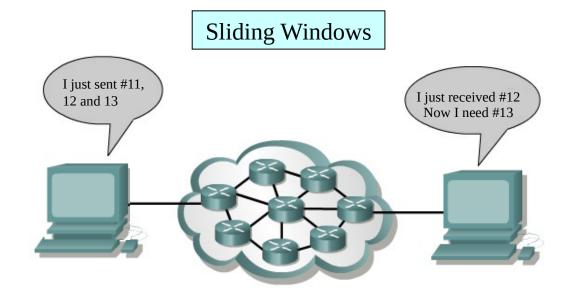
Transport

Application

Transport

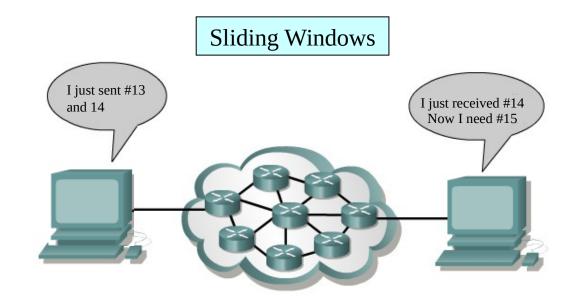
Internet

Network Access



This indicates that packet 13 either did not arrive, or arrived with errors, and needs retransmission.

Application Transport Internet Network Access



The sliding window has worked as the last packet sent has arrived.

The only **Transport** layer **protocols** are **TCP** and **UDP**.

- 1. Transmission Control Protocol (TCP)
 - Connection-oriented protocol
 - End-to-end operation
 - Flow control sliding windows
 - Reliability—sequence acknowledgments

numbers

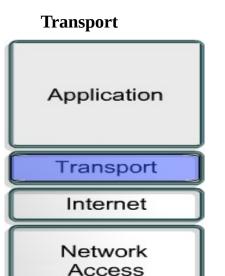
and

2. User Datagram Protocol (UDP)

Connectionless

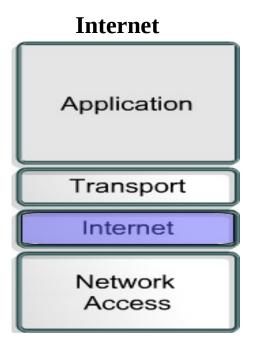
152

Unreliable (no acknowledgments or error checking)



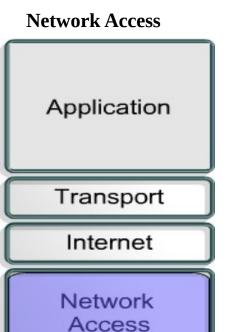
3. TCP/IP Internet Layer

- Two purposes are determining the best path and packet-switching.
- No error checking or correction
- > Protocols:
 - Internet Protocol (IP) connectionless, best-effort delivery routing of packets; determines best path to destination
 - Internet Control Message Protocol (ICMP) control and messaging
 - Address Resolution Protocol (ARP) determines the MAC address, for a known IP address.
 - Reverse Address Resolution Protocol (RARP) determines the IP address for a known MAC address.



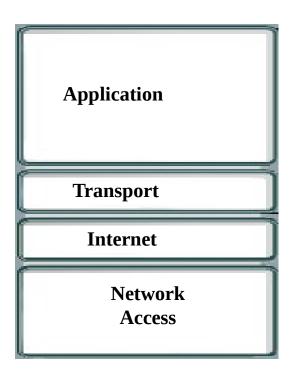
4. TCP/IP Network Access Layer

- Allows an IP packet to make a physical link to the network media
- Maps IP addresses to MAC addresses
- Encapsulates IP packets into frames
- Drivers for modem cards, and other devices operate at the network access layer.
- Serial Line Internet Protocol (SLIP) and Point-to-Point Protocol (PPP) provide network access.
- ARP and RARP also work at this layer.
 - **✓** Note: This layer is **not** defined by **TCP/IP**.

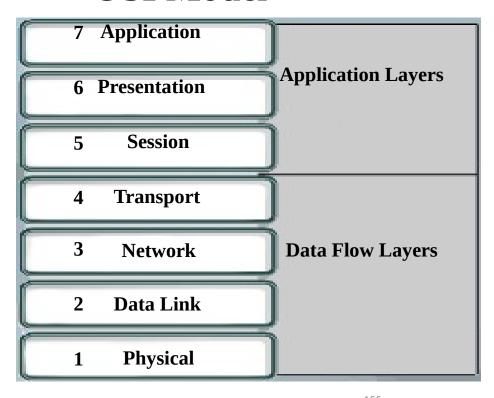


Comparing TCP/IP and OSI

TCP/IP Model



OSI Model



Activity

- 1. Write the advantages and disadvantages of OSI Model
- 2. Write the advantages and disadvantages of TCP/IP Model
- 3. Write the difference and similarities between OSI and
- TCP/IP models
- 4. Are OSI and TCP concepts outdated?
- 5. Can TCP work without OSI?
- 6 Which layors of OSI are relevant to TCD2
- 6. Which layers of OSI are relevant to TCP?7. Write the difference between Circuit-switching and Packet-
- switching?8. Explain the term IP packet?