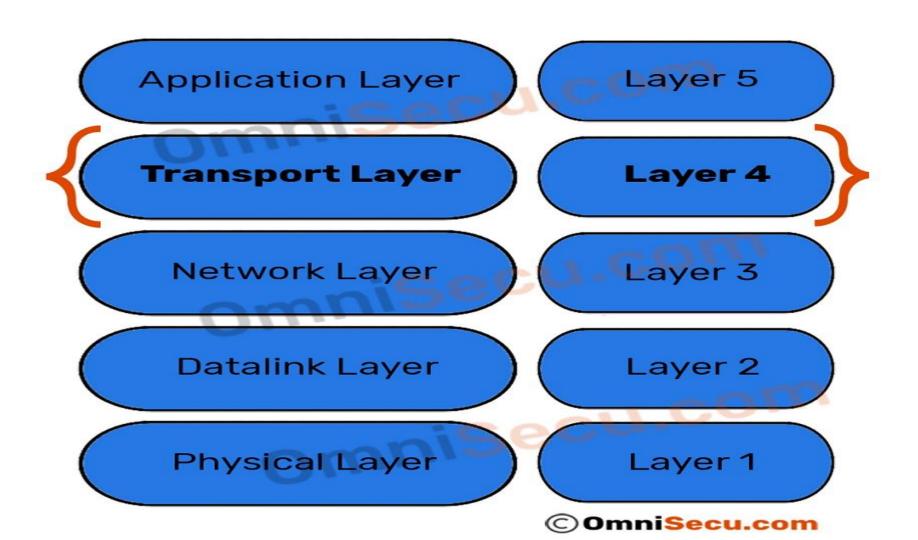
# TCP and UDP Protocols

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## Functions of Transport layer

- To provide an interface for network applications to access the network
- To provide an interface for accepting data from different applications on the source computer and sending that data to the recipient applications on different destination computers (Multiplexing).
- Similarly on the destination computer, the incoming data from different remote computers need to be directed to the correct applications for that data was meant for (De-multiplexing).
- It treats each packet independently for final delivery, because each packet for final delivery belongs to different Applications on different destination computers.
- It has protocol/mechanisms for loss-free delivery of data to destination applications.
- Transport layer provides mechanisms for error checking, flow control, and retransmission of lost data.
- Two major protocols at the Transport layer,
  - Transmission Control Protocol (TCP)
  - User Datagram Protocol (UDP)

# TCP Layered Architecture



## **Transmission Control Protocol (TCP)**

- It is a connection oriented protocol, which means the devices should open a connection before transmitting data and should close the connection gracefully after transmitting the data.
- It assure reliable delivery of data to the destination.
- It provides extensive error checking mechanisms such as flow control and acknowledgment of data.
- Sequencing of data
- Delivery of data is guaranteed
- It is comparatively slow because of these extensive error checking mechanisms
- Multiplexing and Demultiplexing is supporting
- Retransmission of lost packets is supported

# User Datagram Protocol (UDP)

- It is a Datagram oriented protocol with no overhead for opening a connection (using three-way handshake), maintaining a connection, and closing (terminating) a connection.
- It is efficient for broadcast/multicast type of network transmission.
- It has only the basic error checking mechanism using checksums.
- There is no sequencing of data in User Datagram Protocol (UDP).
- The delivery of data cannot be guaranteed in User Datagram Protocol (UDP).
- User Datagram Protocol (UDP) is faster, simpler and more efficient than TCP. It
- It is less robust then TCP
- Multiplexing and Demultiplexing is possible in User Datagram Protocol (UDP) using UDP port numbers.

#### TCP connection.

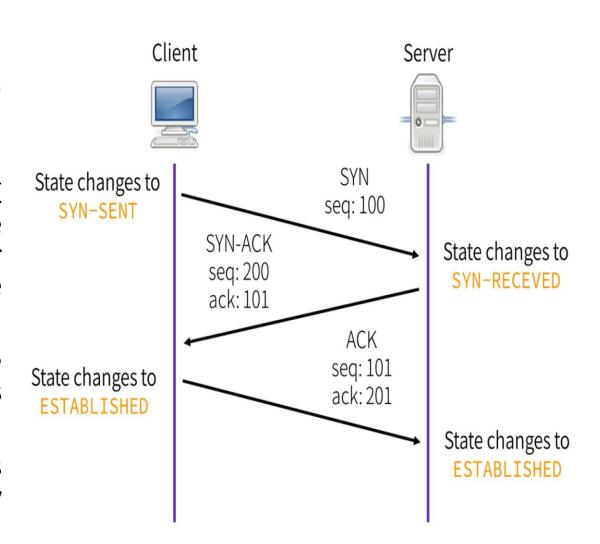
- A connection oriented protocol establishes connection between virtual path between source and destination. All segment belonging to the message are sent over this virtual path.
- TCP connection are virtual, not physical. It operates at higher level.
- It use the services of IP to deliver individual segment to the receiver, but it controls the connection itself.
- If a segment is lost or corrupted, it retransmits.
- If a segment arrives out of order, TCP holds it until the missing segments arrive
- Connection-oriented transmission requires three phases:
  - Connection establishment,
  - data transfer,
  - connection termination.

### Connection Establishment

- TCP transmits data in full duplex mode. When two TCPs in two machines are connected, they are able to send segments to each other simultaneously.
- This implies that each party must initialize communication and get approval from the other party before any data are transferred.
- This connection establishment in TCP is called three way hand shaking

# 3 way Handshaking

- The client sends a SYN (synchronize) packet to the server, which has a random sequence number.
- The server sends back a SYN-ACK packet, containing a random sequence number and an ACK number acknowledging the client's sequence number.
- The client sends an ACK number to the server, acknowledging the server's sequence number.
- The sequence numbers on both ends are synchronized. Both ends can now send and receive data independently

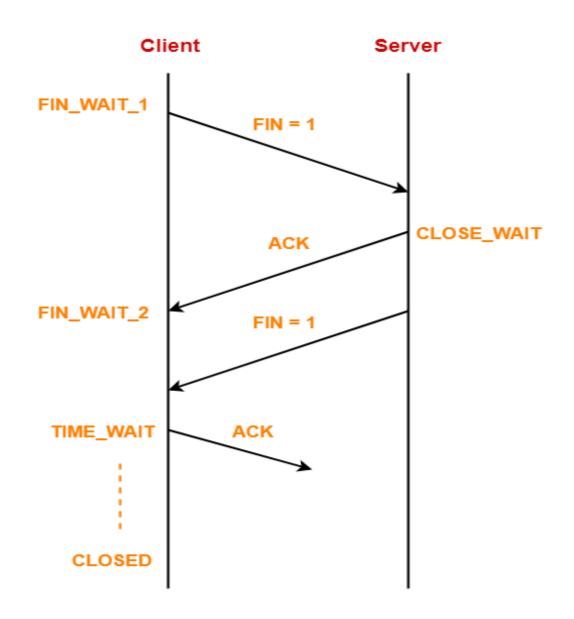


### Data transfer

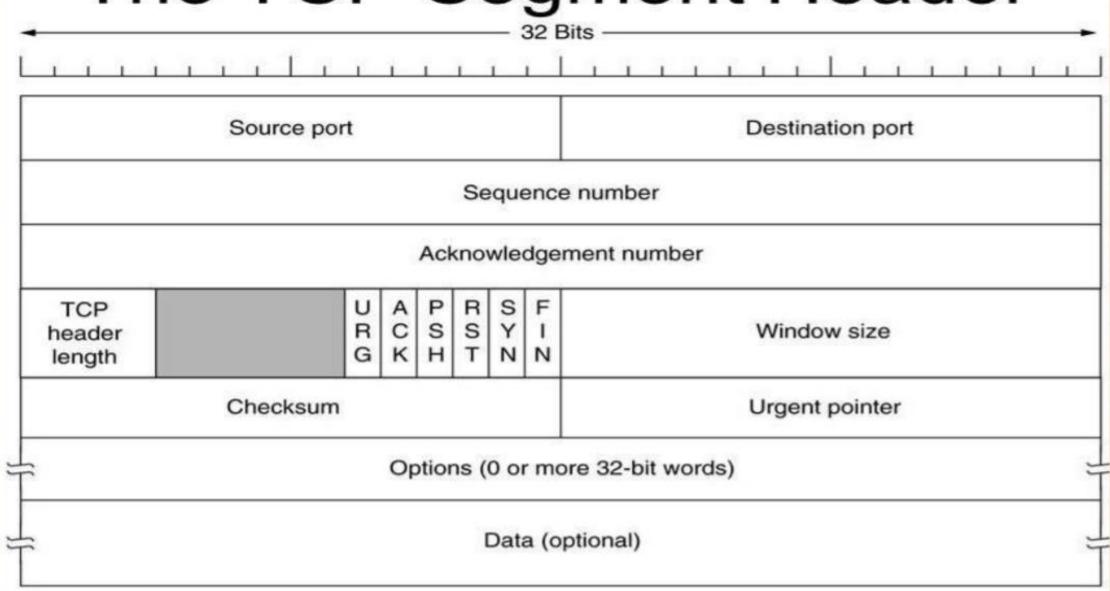
- After connection establishment the data transfer in TCP takes place.
- The data transfer takes place in bidirectional.
- Where the client and server can both send data and acknowledgments.
- The acknowledgments are piggybacked with data
- It uses buffer to store the stream of data coming from the sending application.

#### Connection termination.

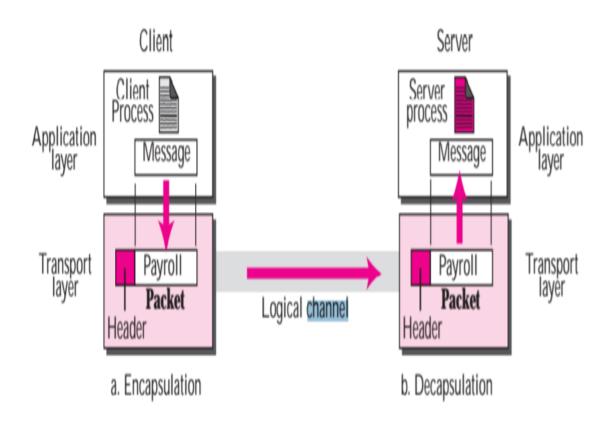
- Any of the two parties involved in exchanging data can close the connection, although it is usually initiated by the client.
- Most of the implementation today allow two options for connection termination they are:
  - 1.way handshaking
  - 2. Four way handshaking with a half -close option.



# The TCP Segment Header



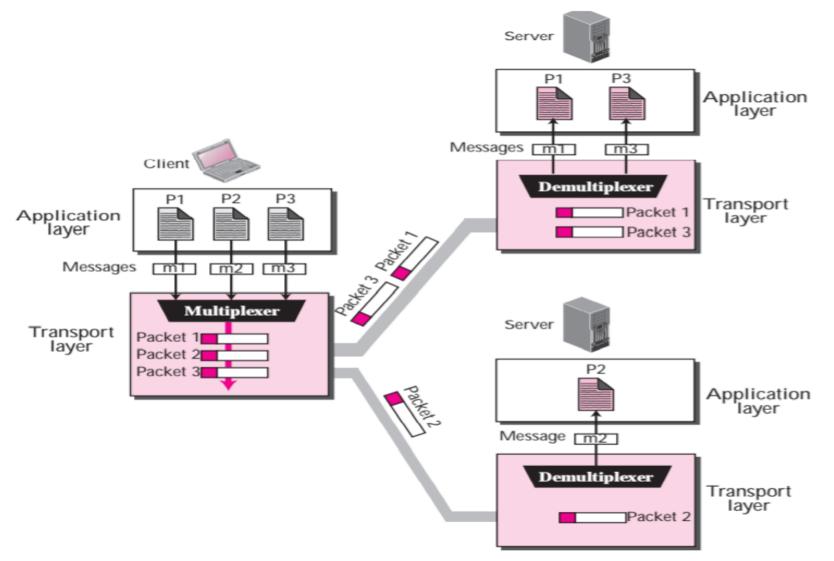
### Encapsulation and Decapsulation



Encapsulation: Encapsulate certain data
Happen at Source node
Adds the header to the payload/ message
and transmit the packet

Decapsulation: opening up encapsulated data back
Done at destination
Removes the header from the packet and creates the original message back

### Multiplexing and Demultiplexing

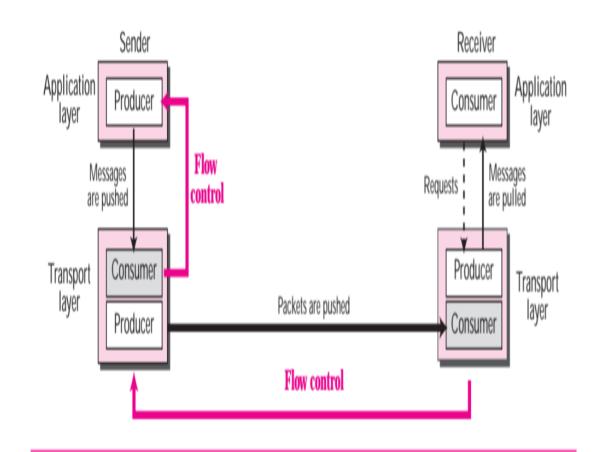


Multiplexing: Many to one Happens at Source Node

De-Multiplexing: one to Many Happens at Destination Node

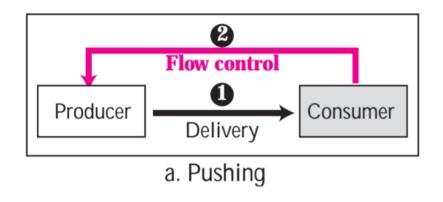
### Flow Control

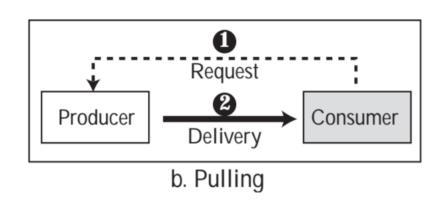
- Used to maintain the balance between production and consumption rates.
- If the items are produced faster than they can be consumed, the consumer can be overwhelmed and needs to discard some items.
- If the items are produced slower than they can be consumed, the consumer should wait; the system becomes less efficient.
- Flow control is used to prevent losing the data items at the consumer site



# Pushing or Pulling

- Delivery of items from a producer to a consumer can occur in one of the two ways: pushing or pulling.
- Pushing: If the sender delivers items whenever they are produced without the prior request from the consumer
- pulling: If the producer delivers the items after the consumer has requested them





### TCP Error Control Methods

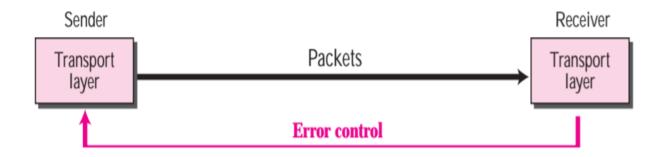
Checksum

Acknowledgment

Time out

#### Error control and its Functions

- Detect and discard corrupted packets.
- Keep track of lost and discarded packets and resend them.
- Recognize duplicate packets and discard them.
- Buffer out-of-order packets until the missing packets arrive.



### Sequence Numbers

- Error control requires that the sending transport layer knows which packet is to be resent
- Receiving transport layer knows which packet is a duplicate, or which packet has arrived out of order.
- This can be done if the packets are numbered.
- Sequence Number field is added to packet by the transport layer
- When a packet is corrupted or lost, the receiving transport layer can inform the sending transport layer to resend that packet using the sequence number.
- The receiving transport layer can also detect duplicate packets if two received packets have the same sequence number.
- The out-of-order packets can be recognized by observing gaps in the sequence numbers.

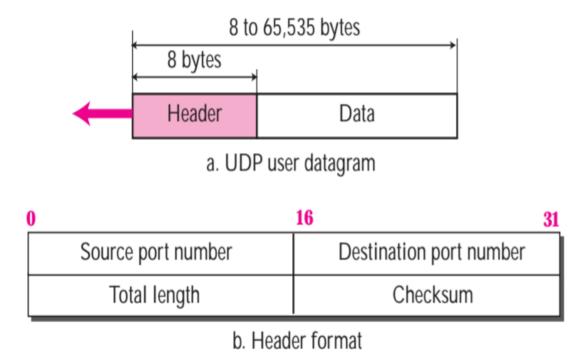
# Acknowledgment

- The receiver side can send an acknowledgement (ACK) for each or a collection of packets that have arrived safe
- The receiver can simply discard the corrupted packets.
- The sender can detect lost packets if it uses a timer.
- When a packet is sent, the sender starts a timer; when the timer expires, if an ACK does not arrive before the timer expires, the sender resends the packet.
- Duplicate packets can be silently discarded by the receiver. Out-of- order packets can be either discarded or stored until the missing ones arrives.

#### UDP

- UDP is a transport protocol that creates a process-to-process communication.
- UDP is a (mostly) unreliable and connectionless protocol that requires little overhead and offers fast delivery.
- The UDP packet is called a user datagram.
- UDP's only attempt at error control is the checksum.
- UDP has no flow-control mechanism.
- A user datagram is encapsulated in the data field of an IP datagram.
- UDP uses multiplexing to handle outgoing user datagrams from multiple processes on one host.
- UDP uses demultiplexing to handle incoming user datagrams that go to different processes on the same host.

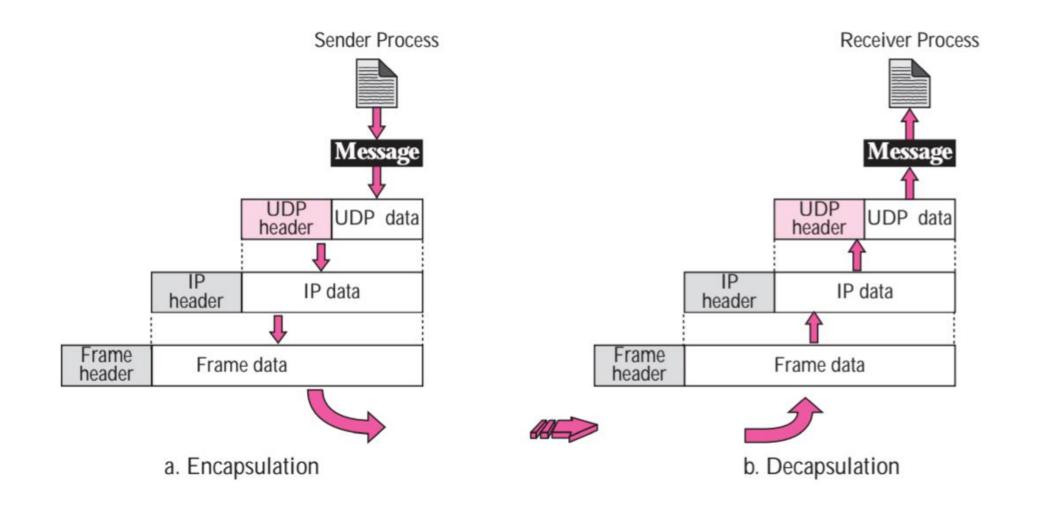
#### **UDP** Header



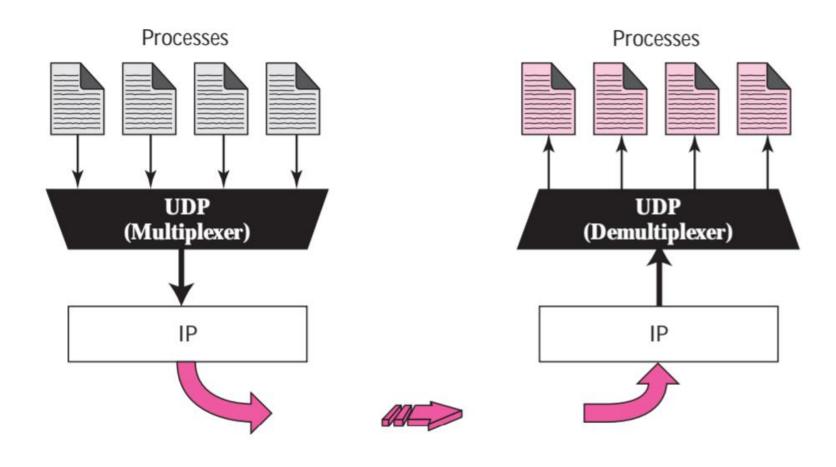
**Source port number:** This is the port number used by the process running on the source host. It is 16 bits long.

**Destination port number:** This is the port number used by the process running on the destination host. It is also 16 bits long.

# Encapsulation and Decapsulation



# Multiplexing and demultiplexing



# Differences are-

Properties	TCP	UDP
Header	Dynamic header ( 20 – 60 B)	Static header of 8 Bytes
Max segment	any size or 2^30 B	short message 65536 Bytes
Flow Control	Yes, Window and seq. no.	NO
Checksum	Compulsory	Optional
Connection nature	TCP+ IP = connection oriented	UDP+ IP= connection less
Error control	Own mechanism	Depends on ICMP (No self feature)
Support multicast	NO	YES
Support broadcast	NO	Yes
Examples service	HTTP,SMTP,FTP,TELNET	TFTP,DNS,SNMP

#### TCP vs UDP

#### **Transport Layer Protocols**

#### UDP

- Fast when compared with TCP
- \* Not a reliable Protocol
- \* Less Overhead
- \* No Connection Establishment
- \* No Acknowledgment
- \* Streaming

#### **Application Layer**

Transport Layer
TCP or UDP

#### **Network Layer**

**Datalink layer** 

Physical layer

#### TCP

- \* Slow when compared to UDP
- \* Reliable Protocol
- \* More Overhead
- Connection Establishment Required
- \* Acknowledgment
- \* Resend lost data



### References

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- file:///C:/Users/S%20R%20N%20REDDY/Downloads/tcptransmissioncontrolprotocol-copy-150311124141-conversiongate01.pdf
- B.A. Forouzan, "TCP/ IP Protocol Suit", 4<sup>th</sup> Edition [ Online Edition]

Q&A

**Thanks**