Computer Network(CSC 503)

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Lecture 32

Transport Layer

- It is the heart of the whole protocol hierarchy
- It is located between the application layer and the network layer.
- It provides services to the application layer and receives services from the network layer.
- Its task is to provide:
 - Reliable, cost-effective data transport from the source machine to the destination machine, independently of the physical network or networks currently in use.

- The ultimate goal of the transport layer is to provide **efficient**, **reliable**, **and cost-effective** service to its users.
- To achieve this goal, the transport layer makes use of the services provided by the network layer.
- The hardware and/or software within the transport layer that does the work is called the **transport entity.**
- There are **two** types of transport service.
 - Connection Oriented Transport Service
 - Connectionless Transport Service
- Connections have three phases: establishment, data transfer, and release.
- Addressing and flow control are also similar in both layers.

Need for Transport Layer

- If the transport layer service is so similar to the network layer service, why are there two distinct layers?
- Why is one layer not adequate?
- The transport code runs entirely on the **users' machines**, but the network layer mostly runs on the routers, which are operated by the carrier.
- What happens if the network layer offers inadequate service? Suppose that it frequently **loses packets**? What happens if **routers crash** from time to time?
- The users have **no real control over the network layer**, so they cannot solve the problem of poor service by using better routers or putting more error handling in the data link layer

In a connection-oriented subnet:

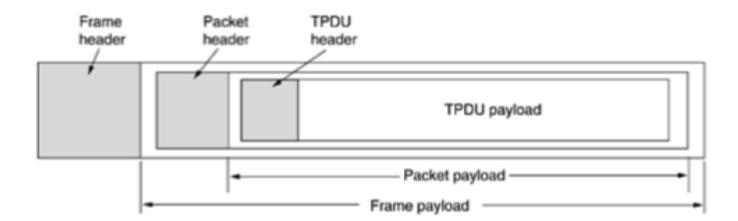
- A transport entity is informed halfway through a long transmission that its network connection has been abruptly terminated,
- What happens to the data currently in transit?
- It can set up a new network connection to the remote transport entity.
- Using this new network connection, it can send a query to its **peer asking which** data arrived and which did not, and then pick up from where it left off.
- In this case the existence of the transport layer makes it possible for the transport service to be **more reliable** than the underlying network service.
- Lost packets and mangled data can be detected and compensated for by the transport layer.

- Application programmers can write code according to a standard **set of primitives** and have these programs work on a wide variety of networks, without having to worry about dealing with **different subnet interfaces and unreliable transmission**
- These transport service primitives can be implemented as calls to **library procedures** in order to make them **independent of the network service primitives**. The network service calls may vary considerably from network to network
- The bottom four layers can be seen as the transport service provider, whereas the upper layer are the transport service user
- Transport layer forms the major boundary between the provider and user of the reliable data transmission service.

Transport Protocol Data Unit (TPDU)

- TPDUs exchanged by the transport layer are contained in packets exchanged by the **network layer**.
- In turn, packets are contained in frames exchanged by the data link layer.
- When a frame arrives, the data link layer processes the frame header and passes the contents of the frame payload field up to the network entity.
- The network entity processes the packet header and passes the contents of the packet payload up to the transport entity.

• Nesting of TPDU, Packet and Frame



Transport Layer Service Primitives

- To allow users to access the transport service, the transport layer must provide some **operations** to application programs, that is known as **transport service** interface.
- Each transport service has its own interface
- Difference between transport service and network service:
 - Transport Service is **reliable** whereas network service is **un-reliable**
 - Network service is **used** only by the **transport entities**.
 - Few users write their own transport entities, and thus few users or programs ever see the bare network service. In contrast, many programs (and thus programmers) see the transport primitives. Consequently, the transport service must be convenient and easy to use.

| Sr. No | Primitive | Meaning |
|-----------|------------|--|
| 1. | LISTEN | When server is ready to accept request of incoming connection, it simply put this primitive into action. Listen primitive simply waiting for incoming connection request. |
| 2. | CONNECT | This primitive is used to connect the server simply by creating or establishing connection with waiting peer. |
| 3. | ACCEPT | It simply accepts incoming connection form peer. |
| 4. | SEND | It is put into action by the client to transmit its request that is followed by putting receive primitive into action to get the reply. Send primitive simply sends or transfer the message to the peer. |
| 5. | RECEIVE | These primitive afterwards block the server. Receive primitive simply waits for incoming message. |
| 6. | DISCONNECT | It is simply used to terminate or end the connection after which no one will be able to send any of the message. |

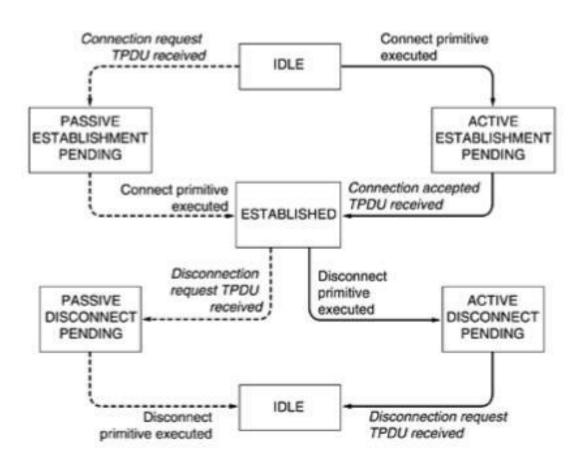
Example of Transport Layer Service Primitive

- Consider an application with a **server** and a number of **remote clients**.
- Step-1: The server executes a LISTEN primitive, by calling a library procedure that makes a system call to block the server until a client turns up.
- Step-2: When a client wants to talk to the server, it executes a CONNECT primitive. The transport entity carries out this primitive by blocking the caller and sending a packet to the server. The packet is encapsulated in the payload contains a transport layer message for the server's transport entity.

- Step-3: The client's CONNECT call causes a CONNECTION REQUEST TPDU to be sent to the server. When it arrives, the transport entity checks to see that the server is blocked on a LISTEN It then unblocks the server and sends a CONNECTION ACCEPTED TPDU back to the client. When this TPDU arrives, the client is unblocked and the connection is established.
- Step-4: Data can now be exchanged using the SEND and RECEIVE primitives. Either party can do a (blocking) RECEIVE to wait for the other party to do a SEND. When the TPDU arrives, the receiver is unblocked. It can then process the TPDU and send a reply

- Step-5: When a connection is no longer needed, it must be released to free up table space within the two transport entities. **Disconnection** has two variants: asymmetric and symmetric.
- In the asymmetric variant, either transport user can issue a DISCONNECT primitive, which results in a DISCONNECT TPDU being sent to the remote transport entity. Upon arrival, the connection is released.
- In the symmetric variant, each direction is closed separately, independently of the other one. When one side does a **DISCONNECT**, that means it has no more data to send but it is still willing to accept data from its partner.
- In this model, a connection is released when both sides have done a DISCONNECT

Example of Transport Layer Service Primitive



Berkeley Sockets

- It is another set of transport primitives known as the socket primitives
- These primitives are widely used for Internet programming or Socket Programming
- They follow the same model of but offer more features and flexibility

| Primitive | Meaning | | | |
|-----------|---|--|--|--|
| SOCKET | Create a new communication end point | | | |
| BIND | Attach a local address to a socket | | | |
| LISTEN | Announce willingness to accept connections; give queue size | | | |
| ACCEPT | Block the caller until a connection attempt arrives | | | |
| CONNECT | Actively attempt to establish a connection | | | |
| SEND | Send some data over the connection | | | |
| RECEIVE | Receive some data from the connection | | | |
| CLOSE | Release the connection | | | |

Transmission Control Protocol (TCP)

- TCP (Transmission Control Protocol) was specifically designed to provide a reliable end-to end byte stream over an unreliable internetwork.
- TCP service is obtained by both the sender and receiver creating end points, called sockets
- Each socket has a socket number (address) consisting of the **IP address** of the host and a 16-bit number local to that host, called a **port**.
- A port is the TCP name. For TCP service to be obtained, a connection must be explicitly established between a socket on the sending machine and a socket on the receiving machine

TCP Port numbers

| Port Protocol | | Use | | |
|---------------|--------|-------------------------------------|--|--|
| 21 | FTP | File transfer | | |
| 23 | Telnet | Remote login | | |
| 25 | SMTP | E-mail | | |
| 69 | TFTP | TFTP Trivial file transfer protocol | | |
| 79 | Finger | Lookup information about a user | | |
| 80 | HTTP | World Wide Web | | |
| 110 | POP-3 | Remote e-mail access | | |
| 119 | NNTP | USENET news | | |

TCP

- Every byte on a TCP connection has its own 32-bit sequence number
- The basic protocol used by TCP entities is the **sliding window protocol**.
- The sending and receiving TCP entities exchange data in the form of segments.
- A TCP segment consists of a fixed **20-byte header** (plus an optional part) followed by zero or more data bytes.
- The TCP software decides **how big segments** should be.
- It can accumulate data from several writes into one segment or can split data from one write over multiple segments.

TCP segment

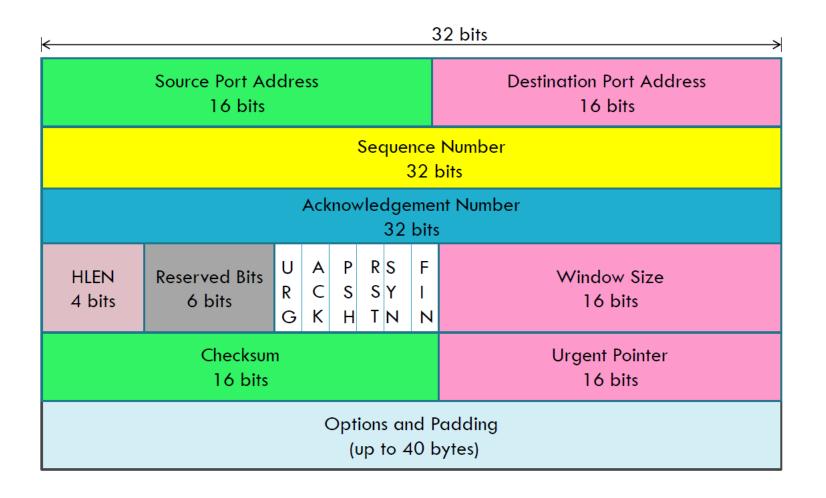
- A TCP segment **encapsulates the data** received from the application layer. The TCP segment itself is encapsulated in an IP datagram, which in turn is encapsulated in a frame at the data link layer.
- Two limits restrict the segment size:
 - Each segment, including the **TCP header**, must fit in the **65,515-byte** IP payload.
 - Each network has a maximum transfer unit, or MTU, and each segment must fit in the MTU. In practice, the MTU is generally 1500 bytes (the Ethernet payload size) and thus defines the upper bound on segment size.

TCP Header Format

- The segment consists of a header of 20 to 60 bytes, followed by data from the application program.
- The header is 20 bytes if there are no options and up to 60 bytes if it contains options.
- TCP segment



TCP Header



TCP Header Format

- Source port address- This is a 16-bit field that defines the port number of the application program in the host that is sending the segment.
- Destination port address- This is a 16-bit field that defines the port number of the application program in the host that is receiving the segment.
- Sequence number- This 32-bit field defines the number assigned to the first byte of data contained in this segment.
 - TCP is a **stream** transport protocol.
 - To ensure connectivity, each byte to be transmitted is numbered.
 - The sequence number tells the destination which byte in this sequence is the **first byte** in the segment.
 - During connection establishment each party uses a **random number generator** to create an **initial sequence number (ISN)**, which is usually different in each direction

- Acknowledgment number- This 32-bit field defines the byte number that the receiver of the segment is expecting to receive from the other party.
- **Header length** This 4-bit field indicates the number of 4-byte words in the TCP header.
 - The length of the header can be between 20 and 60 bytes. Therefore, the value of this field is always between 5 (5 \times 4 = 20) and 15 (15 \times 4 = 60).
- Reserved Bits:- This 6-bit field is reserved for future use. The value is set to 0

• Control- This field defines 6 different control bits or flags

• One or more of these bits can be set at a time.

• These bits enable flow control, connection establishment and termination, connection abortion, and the mode of data transfer in TCP

URG: Urgent pointer is valid

ACK: Acknowledgment is valid

PSH: Request for push

RST: Reset the connection

SYN: Synchronize sequence numbers

FIN: Terminate the connection

| URG ACK | PSH | RST | SYN | FIN |
|---------|-----|-----|-----|-----|
|---------|-----|-----|-----|-----|

- Window size- This field defines the window size of the sending TCP in bytes.
 - Note that the length of this field is 16 bits, which means that the maximum size of the window is 65,535 bytes.
 - This value is normally referred to as the **receiving window** (**rwnd**) and is determined by the receiver. The sender must obey the dictation of the receiver in this case
- Checksum. This 16-bit field contains the checksum.
 - Use of the checksum for TCP is mandatory.

- Urgent pointer- This 16-bit field, is valid only if the urgent flag is set
 - It is used when the segment contains urgent data.
 - It defines a value that must be added to the sequence number to obtain the number of the last urgent byte in the data section of the segment.

• Options- There can be up to 40 bytes of optional information in the TCP header.

Example

- Ex. 1: The following is a dump of a TCP header in hexadecimal format:
- 05320017 00000001 00000000 500207FF 00000000
- (i) What is the source port number?
- (ii)What is the destination port number
- (iii) What is the length of the header?
- (iv) What is the type of segment?
- (v) What is the window size?

Soln.:

The dump of a TCP header in hexadecimal format: 05320017 00000001 00000000 500207FF 00000000

- 1) Source port number:- (2 byte) ->) 0532(in hex) or 1330 (in decimal)
- 2) Destination port number:- (2 byte) -> 0017(in hex) or 23(in decimal))
- 3) Length of the header (4 bits) 5 (in hex and decimal) Means 5 * 4 = 20 bytes header
- 4) Type of the segment 0X02:-is control field and this indicates a SYN packet.
- 5) Window size 07FF ((in hex) or 2047 (in decimal)