OSI Model – Layers 4-7

INFO-6078 – Managing Enterprise Networks



4 – The Transport Layer

Transport

Application

High-level APIs that provide access to network resources

Data translation services including encoding, compression and encryption

Management of communication sessions

Provides segmentation and process-to-process message delivery

Error-free transmission of frames between nodes

Transmission of bits over a medium; includes mechanical and electrical specifications

4 – The Transport Layer

- The transport layer is responsible for segmentation, establishing and maintaining temporary communication channels between application processes on different hosts
- Additionally, the transport layer may provide mechanisms for reliable delivery of messages, error-checking and flow control
- In the TCP/IP suite of protocols, two transport layer protocols are available to meet different reliability requirements:
 - Transmission Control Protocol (TCP)
 - User Datagram Protocol (UDP)



4 – The Transport Layer - Segmentation

- Segmentation allows communication streams from multiple users and applications to share the network by means of multiplexing or interleaving
- Each segment is treated as a separate message and will require addressing and control information for delivery to a destination













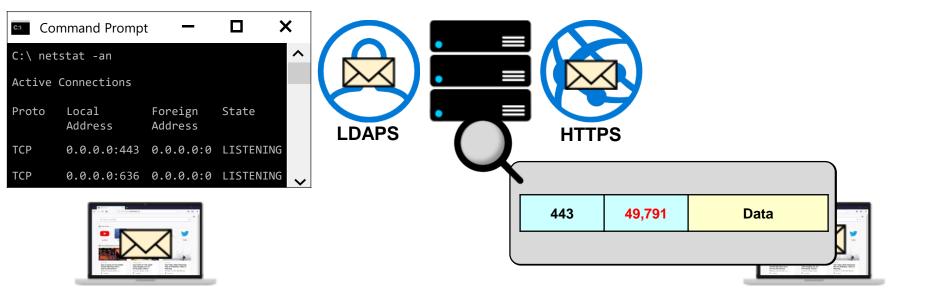
4 – The Transport Layer – Port Numbers

- Both TCP and UDP track connections with the use of source and destination port numbers that range from 0 - 65,535
- The Internet Assigned Numbers Authority (IANA) maintains the registry of assigned port numbers
- Ports are divided into ranges, detailed below:

Port Range	Description
0 to 1,023	Well-Known Ports
1,024 to 49,151	Registered Ports
49,152 to 65,535	Dynamic/Private/Ephemeral Ports

4 – The Transport Layer – Port Numbers

- Services awaiting connections normally use ports in the wellknow or registered range
- Client devices select ports from the Dynamic/Private/Ephemeral range to set as the source port



4 – The Transport Layer – Port Numbers

Port #	ТСР	UDP	Description
22	✓		Secure Shell (SSH)
53	✓	✓	Domain Name System (DNS)
67/68		✓	Dynamic Host Configuration Protocol (DHCP)
80/443	✓		Hypertext Transfer Protocol (HTTP) / HTTP over TLS/SSL (HTTPS)
80/443		✓	Quick UDP Internet Connections (QUIC)
123		✓	Network Time Protocol (NTP)
161/162	✓	✓	Simple Network Management Protocol (SNMP)
1812/1813	✓	✓	RADIUS authentication/accounting protocol
3389	✓	✓	Microsoft Terminal Server (RDP)
5060	✓	✓	Session Initiation Protocol (SIP)

Transmission Control Protocol (TCP)

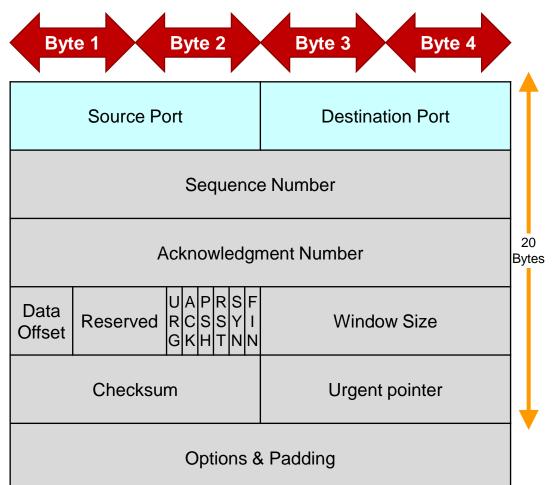
- Provides connection-oriented, stateful delivery of information
- TCP can correct many of the issues related to lower layer delivery of information such as lost, duplicate, or out-of-order packets
- TCP must open a communication session before application data can be transferred
- When the session is no longer needed, TCP must close the connection



Transmission Control Protocol (TCP)

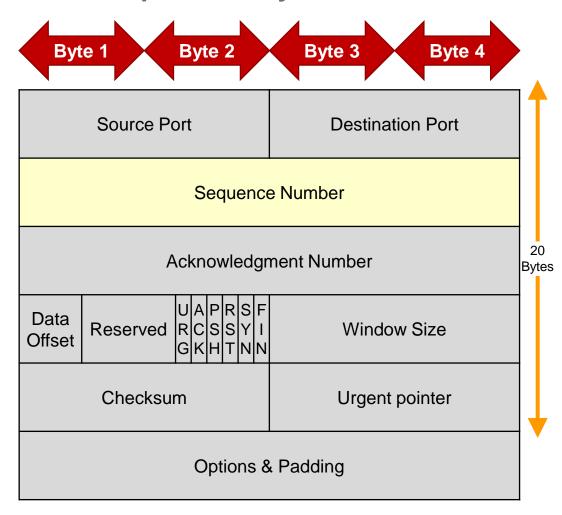
TCP is used as the transport layer protocol when the communication requires one or more of the following:

- Guaranteed data delivery
 - Checksums and timers ensure corrupt or lost segments are retransmitted
- Ordered data delivery
 - Ensures segments are delivered up the OSI model in the same sequence in which they were transmitted
- Session flow control
 - Flow control manages individual communication channels to ensure the receiving device does not get overwhelmed with data



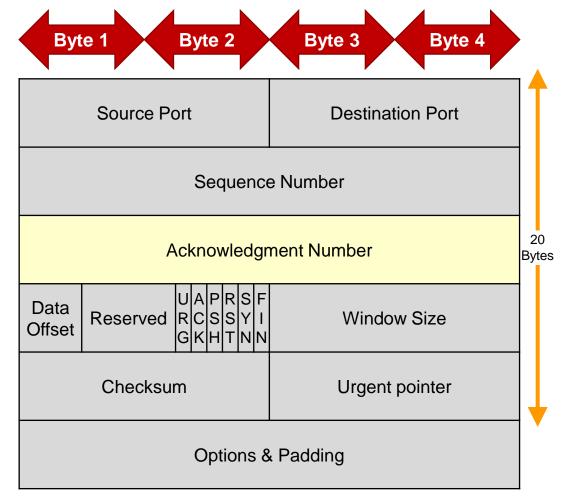
Source/Destination Port 16 bits:

 Identifies the sending and receiving processes



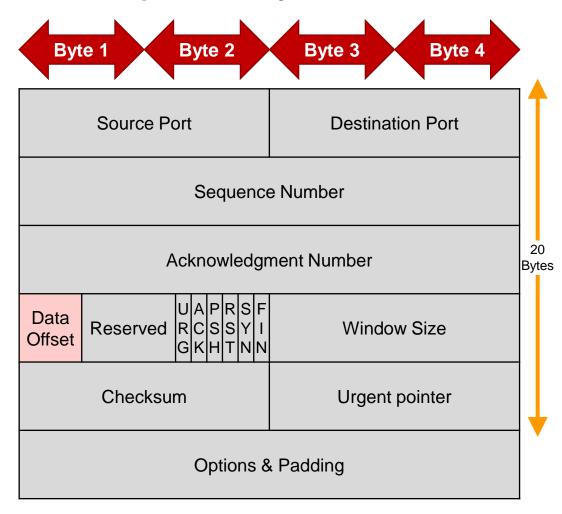
Sequence Number 32 bits:

- The sequence number identifies the first byte in the segment
- If the SYN flag is set, the sequence is the Initial Sequence Number (ISN)



Acknowledgment Number 32 bits:

- If the ACK flag is set, the acknowledgement contains the value of the next sequence number the sender of the segment is expecting to receive
- This also acknowledges receipt of all data segments before this value

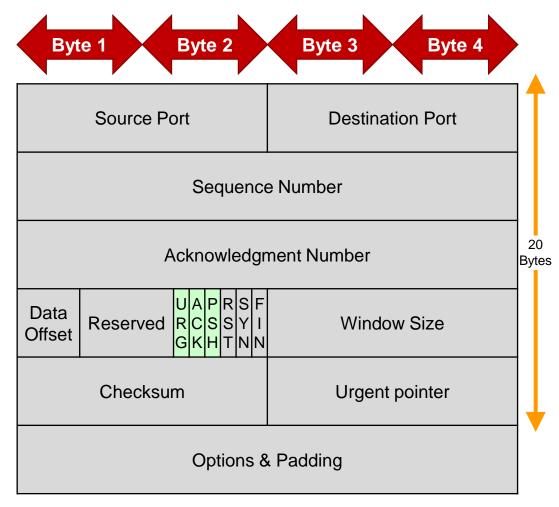


Data Offset 4 bits:

- Sets the number of 32 bit words in the header
- Indicated where the data begins

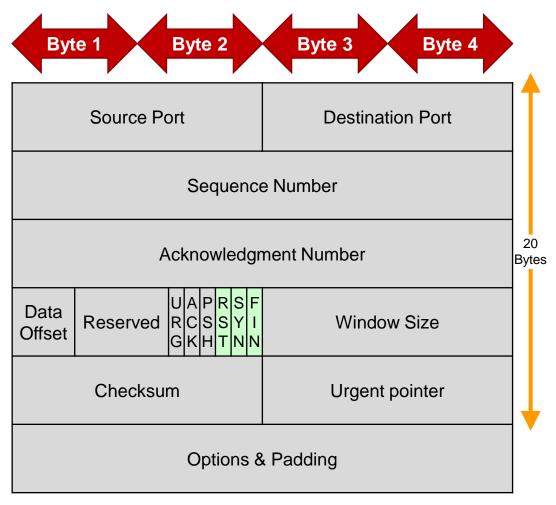
Reserved 6 bits:

- Reserved for future use
- The latter 3 bits are specified in RFC 3540/RFC 3168 and used for congestion control



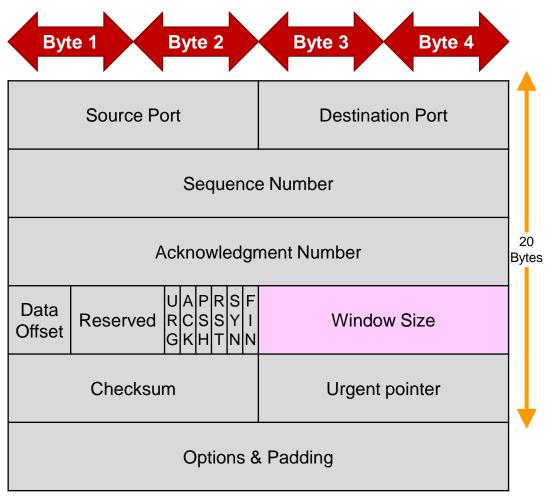
Flags 6 bits:

- URG Urgent Pointer field significant
- ACK –
 Acknowledgment field significant
- PSH Push Function: Push buffered data to the application



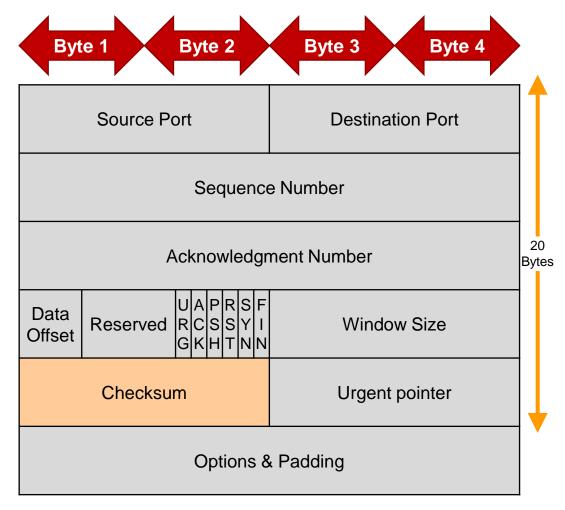
Flags 6 bits:

- RST Reset the connection
- **SYN** Synchronize sequence numbers
- FIN No more data from sender



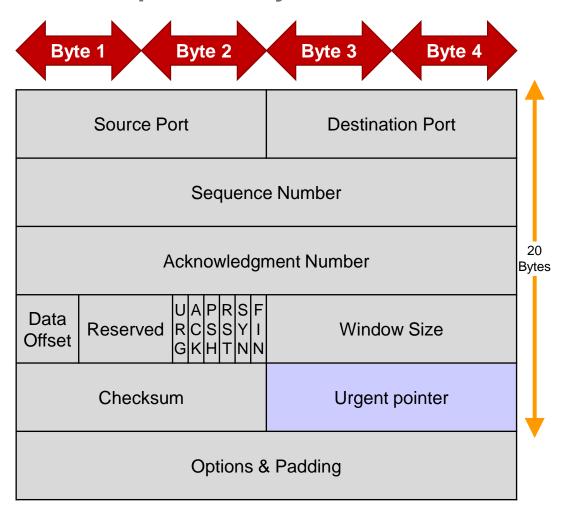
Window Size 16 bits:

 The amount of data in bytes beyond that identified in the acknowledgement field that the sender is willing to accept



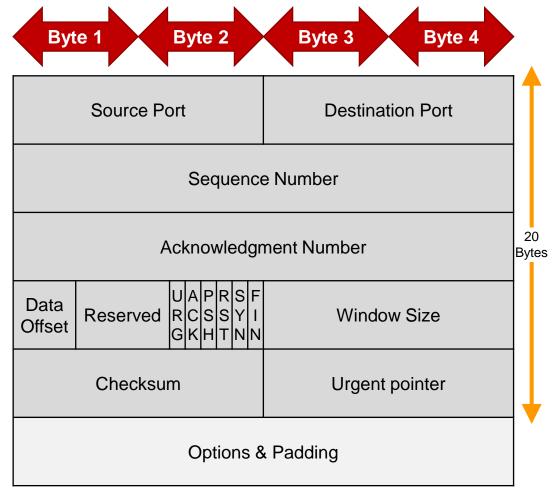
Checksum 16 bits:

- Detects corruption and transmission errors based on a pseudoheader
- The pseudo-header contains:
 - Source/Destination IP Addresses
 - Protocol Field
 - Length of the TCP header and payload



Urgent Pointer 16 bits:

- Indicates the end of the urgent data offset in relation to the sequence number
- Only used if the URG flag is set



Options varies:

 Used to communicate various control information such as: Maximum Segment Size or Window Scale

Padding varies:

 A number of zeros that is added when options are used to ensure the segment is a multiple of 32 bits

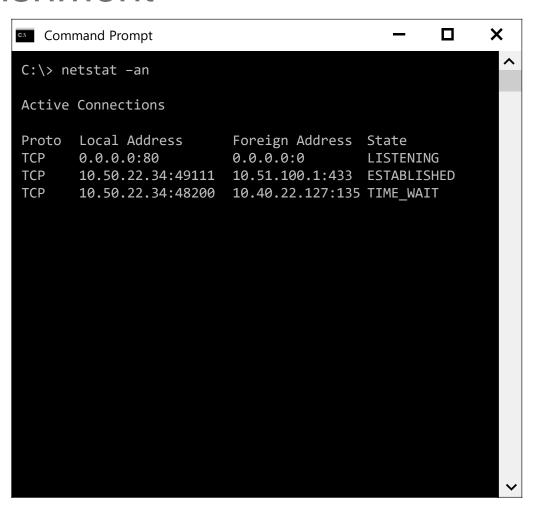
TCP Session and Connection Management

- TCP must setup sessions before application data can be transferred
- Additionally, sessions must be managed and kept in an "alive" state to continue to transfer information
- Session termination occurs when TCP connections are no longer required, or sessions reach a timeout limit



TCP Session Establishment

- TCP uses a process known as the three-way handshake to establish a connection with a remote device
- Before this can occur, a service must bind to a port and listen for incoming connection requests
- Use the flags SYN, SYN/ACK and ACK



TCP Session Establishment

Step 1 (SYN)

 Host A initiates an active open by sending a SYN request to a server

Step 2 (SYN+ACK)

- Host B acknowledges Host A's request by incrementing the Host A's sequence number
- Host B also requests a return session by setting the SYN flag

Step 3 (ACK)

 Host A acknowledges Host B's request by incrementing Host B's sequence number







```
rame 3: 54 bytes on wire (432 bits), 54 bytes captured (432 bits) on interface 0
Ethernet II, Src: IntelCor_59:c2:84 (e8:2a:ea:59:c2:84), Dst: AsustekC_cf:d6:f0 (d8:50:e6:cf:d6:f0
Internet Protocol Version 4, Src: 192.168.1.121, Dst: 23.3.98.34
Transmission Control Protocol, Src Port: 52744, Dst Port: 80, Seq: 1, Ack: 1, Len: 0
  Source Port: 52744
  Destination Port: 80
  [Stream index: 0]
  [TCP Segment Len: 0]
  Sequence number: 1
                        (relative sequence number)
   [Next sequence number: 1
                              (relative sequence number)]
  Acknowledgment number: 1
                              (relative ack number)
  0101 .... = Header Length: 20 bytes (5)
  Flags: 0x010 (ACK)
  Window size value: 513
  [Calculated window size: 131328]
  [Window size scaling factor: 256]
  Checksum: 0xb5ce [unverified]
  [Checksum Status: Unverified]
  Urgent pointer: 0
```

 When a SYN is received, the host starts the SYN-RECEIVED timer, allowing 75 seconds for the handshake to complete



TCP Session Termination

Step 1 (FIN)

 When host A is ready to terminate the session it sends a TCP segment with the FIN flag set to host B

Step 2 (FIN+ACK)

- Host B acknowledges the request by responding and setting the ACK flag
- Host B also sends a FIN request to close the connection

Step 3 (ACK)

 Host A acknowledges Host B's FIN by responding with an ACK

TCP session termination can be in the form of a three-way or four-way handshake





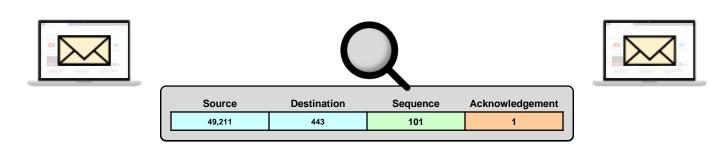


```
rame 10: 54 bytes on wire (432 bits), 54 bytes captured (432 bits) on interface 0
Ethernet II, Src: IntelCor_59:c2:84 (e8:2a:ea:59:c2:84), Dst: AsustekC_cf:d6:f0 (d8:50:e6:cf:d6:f0)
Internet Protocol Version 4, Src: 192.168.1.121, Dst: 23.3.98.34
Transmission Control Protocol, Src Port: 52744, Dst Port: 80, Seq: 300, Ack: 386, Len: 0
   Source Port: 52744
   Destination Port: 80
   [Stream index: 0]
   [TCP Segment Len: 0]
   Sequence number: 300
                           (relative sequence number)
   [Next sequence number: 300
                                 (relative sequence number)]
   Acknowledgment number: 386
                                 (relative ack number)
   0101 .... = Header Length: 20 bytes (5)
  Flags: 0x010 (ACK)
   Window size value: 511
   [Calculated window size: 130816]
   [Window size scaling factor: 256]
   Checksum: 0xb324 [unverified]
   [Checksum Status: Unverified]
   Urgent pointer: 0
```

 The host that closed the connection starts the **Time Wait** timer, which allows any outstanding segments to be received.

TCP Sequence Numbers & Acknowledgements

- TCP uses sequence numbers to identify every byte of data
- As each byte of data leaves the source host, the sequence number is incremented to match the amount of data transmitted
- Destination hosts confirm receipt of data using the acknowledgement field in the TCP header
- Sequence numbers are also used to reorder received data should it arrive out of sequence



TCP Sequence Numbers & Acknowledgements

- When retransmissions occur, sequence numbers are used to identify duplicate transmissions
- In RFC 793, the initial sequence number is generated based on a clock that increments every 4 microseconds and resets every 4.55 hours (2³²)
- This provided attackers the ability to guess sequence numbers based and have the ability to hijack TCP sessions
- Newer implementations of TCP use a random number generator or other methods to choose the ISN



TCP Sequence Numbers & Ordered Delivery











Datagram 1

Datagram 5

Datagram 4

Datagram 2

Datagram 3



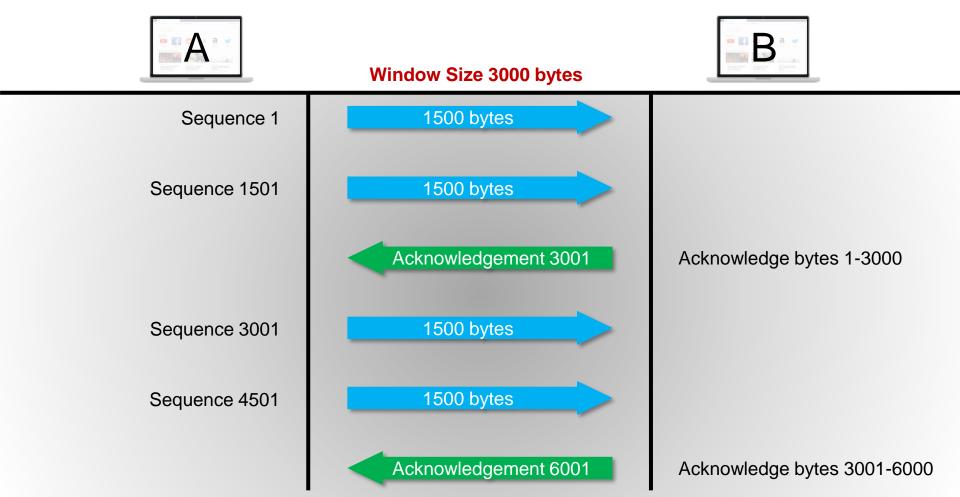






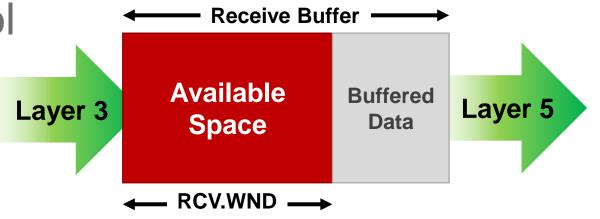


TCP Window Size



TCP Flow Control

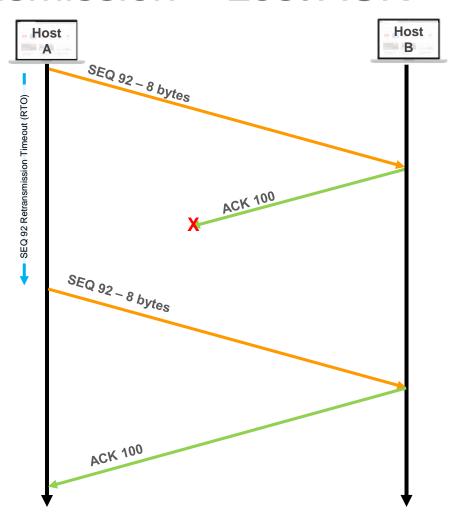
 TCP utilizes endto-end flow control mechanisms to limit the amount of data a sender can



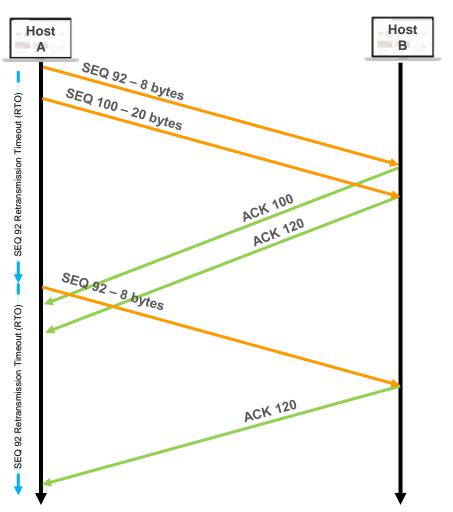
transmit; to prevent it from overwhelming the receiving device

- To accomplish this, the receiving device advertises the amount of available space in it's buffer to the sending device in a value known as the receive window
- When the receiver advertises a receive window of 0, the sending device stops sending data and starts the persist timer, which is employed to prevent a deadlock situation

TCP Retransmission – Lost ACK



TCP Retransmission – Premature Timeout

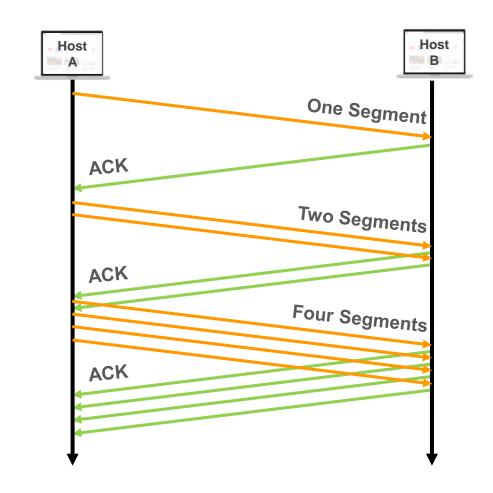


TCP Congestion Control

- Updated with RFC5681, TCP utilizes four mechanisms to improve performance and provide congestion control
- The congestion detection in TCP is decentralized, and comes from inferred feedback from the TCP hosts:
 - if transmitted segments are acknowledged, there is no congestion
 - If transmitted segments are not acknowledged, the network is congested
- The algorithms that provide congestion control are:
 - Slow-Start
 - Congestion Avoidance
 - Fast Retransmit
 - Fast Recovery

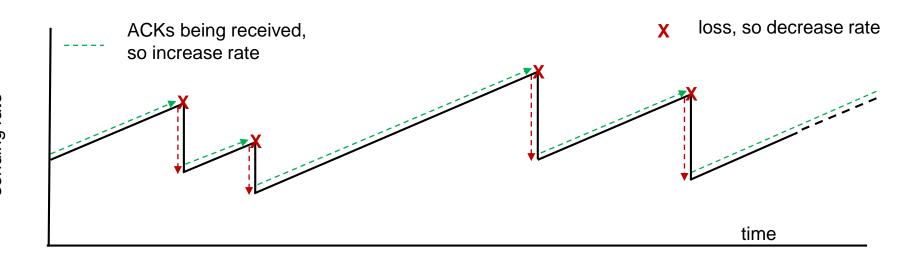
TCP Slow-Start

- When the connection commences, slow start has a congestion window of 1, 2, 4 or 10 maximum segment size (MSS)
- Each time transmitted segments are acknowledged, the congestion window is adjusted by that amount, essentially doubling the amount of transmitted data until a loss is detected, or the limit of the Receive Window is reached



TCP Congestion Avoidance

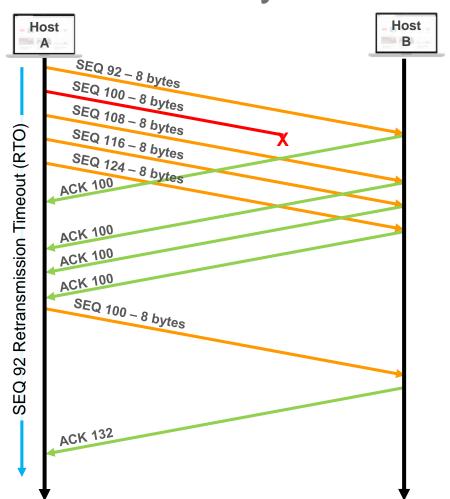
- Congestion avoidance is closely tied to Slow-Start.
- Slow-Start is used to increase the size of the congestion window, but congestion avoidance manages the window size
- When a loss is detected, the congestion window is cut in half and slow-start is restarted to increase the window size



sending rate

TCP Fast Retransmit & Fast Recovery

- Fast Retransmit allows a sending host to reduce the amount to time it takes to recover from a lost segment
- If three or more duplicate acknowledgements are received, fast retransmit sends the next segment again
- Fast Recovery is invoked until all transmitted segments are acknowledged

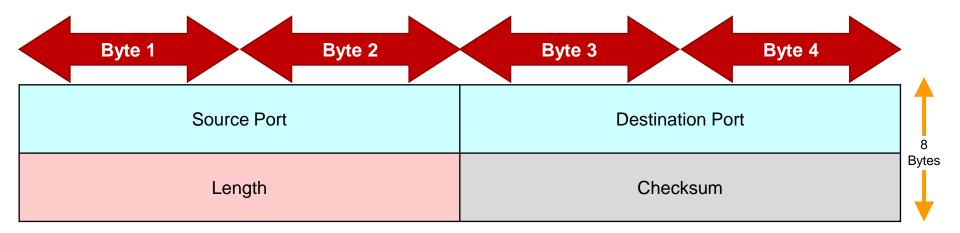


User Datagram Protocol (UDP)

- Originally defined in RFC 768
- UDP is a connectionless protocol and no session is established before application data is sent

UDP is used as the transport layer protocol when the communication requires one or more of the following:

- Fastest Possible Delivery
 - UDP is ideally suited to communications that are time-sensitive
- Tolerance to Dropped/Lost Datagrams
 - UDP does not provide mechanisms to retransmit drop or lost datagrams
- Tolerance to Unordered Datagrams
 - UDP does not reorder datagram delivery before passing received data up the OSI model

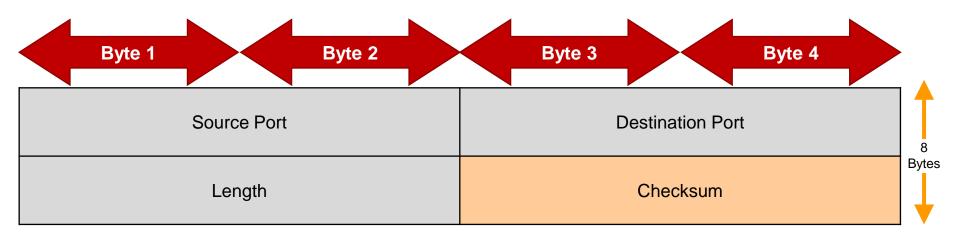


Source/Destination Port 16 bits:

Identifies the sending and receiving processes

Length 16 bits:

The size of the header and data in bytes



Checksum 16 bits:

- Detects corruption and transmission errors based on a pseudoheader
- The pseudo-header contains:
 - Source/Destination IP Addresses
 - Protocol Field
 - Length of the UDP header and payload

5 – The Session Layer

Session

Application High-level APIs that provide access to network resources

Data translation services including encoding, compression and encryption

Management of communication sessions

Provides segmentation and process-to-process message delivery

Provides routing and node-to-node delivery

Transmission of bits over a medium; includes

5 – The Session Layer

- The session layer is responsible for the setup, maintenance and teardown of connections (sessions) between source and destination applications
- The transfer of data is managed by lower layers of the OSI model, but the session layer manages the synchronization of connections
- Transmission modes in the session layer include simplex, half-duplex and full-duplex



6 – The Presentation Layer

High-level APIs that provide access to network Data translation services including encoding, **Presentation** compression and encryption

6 – The Presentation Layer

- The presentation layer is responsible for transferring data to endpoints in a format that the endpoint is capable of receiving such as ASCII or binary data streams
- Encryption and compression are commonly performed in the presentation layer
- The presentation layer is also capable of serializing and deserializing of complex data objects



7 – The Application Layer

_		
7	Application	High-level APIs that provide access to network resources
		Data translation services including encoding, compression and encryption
		Management of communication sessions
		Provides segmentation and process-to-process message delivery
		Provides routing and node-to-node delivery
		Error-free transmission of frames between nodes
		Transmission of bits over a medium; includes

7 – The Application Layer

- The application layer enable user and system applications to interface with the network stack and communicate across a network
- Functions of the application layer include identification of communication endpoints, availability determination and interface management
- The OSI model does not include specifications for specific socket interfaces, and their design is implementation specific



References

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