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Chapter: 6

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Computer Networks Assignment Chapter №6

- (a) What are the source and destination MAC addresses (in Hex)?
- (b) What are the source and destination IP addresses (in dotted decimal)?
- (c) What protocol type is the payload carried by IP packets?
- (d) What is the source port number? How would you classify the source port?
- (e) What is the destination port number? How would you classify the destination port?

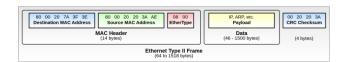


Fig 6-0-1 IPv4 Header Structure Diagram (Task 1)



Fig 6-0-2 Forwarding table (Task 1)

Solution. a) Source and destination MAC addresses

Destination MAC: 34:c9:3d:1d:e1:e7 Source MAC: f0:33:e5:7a:cf:ed

(b) Source and destination IP addresses

The IP header starts after the Ethernet type field (0800) at offset 14 in the hex stream. The IP header length is fixed at 20 bytes.

Source IP: 10.0.6.202 (0a:00:06:ca)

Destination IP: 10.62.33.236 (0a:3e:21:ec)

(c) Protocol type

The protocol type field in the IP header is 06, which corresponds to TCP.

(d) Source port number and classification

The source port is the first field in the TCP header.

Source port: 44347 (0xadc7 in hex)

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Classification: Ephemeral port (typically for outgoing connections)

(e) Destination port number and classification

The destination port is the second field in the TCP header.

Destination port: 80 (0x0050 in hex)

Classification: Well-known port (for HTTP traffic)

Task 2. You are hired to design a reliable byte-stream protocol that uses a sliding window (like TCP). This protocol will run over a 1-Gbps network. The RTT of the network is 100 ms, and the maximum segment lifetime is 30 seconds. How many bits would you include in the AdvertisedWindow and SequenceNum fields of your protocol header?

Solution. 1 Gbps = $1\ 000\ 000\ 000$ bits per second

Bandwidth-Delay Product (BDP) = $Bandwidth \cdot RTT = 1000000000bps \cdot 0.1s = 12500000$ bytes

AdvertisedWindow size: Needs to be able to represent up to 12.5 MB

 $log_2(12500000) => 24$ bits

The sequence number space should cover the maximum segment lifetime (MSL)

With 1 Gbps bandwidth and MSL of 30 seconds, the maximum sequence number range needed can be calculated as $MaxDataSentinMSL = Bandwidth \cdot MSL$

Bandwidth = $1 \text{ Gbps} = 10^9 \text{ bits per second}$

MSL = 30 second

 $MaxDataSentinMSL = 10^9 bps \cdot 30s = 3.75 \cdot 10^9$ bytes

Number of Bits for Sequence Number: $log_2(3.75 \cdot 10^9) => 32$ bits

Summary:

AdvertisedWindow Field: 24 bits SequenceNum Field: 32 bits

Task 3. Consider Fig. 6-1. Assuming TCP Reno is the protocol experiencing the behavior shown above, answer the following questions. In all cases, you should provide a short discussion justifying your answer

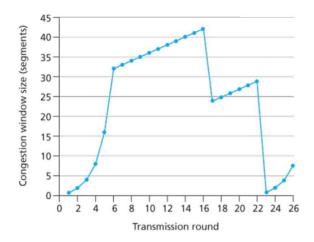


Fig 6-1 Forwarding table (Task 3)

- (a) Identify the intervals of time when TCP slow start is operating.
- (b) Identify the intervals of time when TCP congestion avoidance is operating.
- (c) After the 16th transmission round, is segment loss detected by a triple duplicate ACK or by a timeout?
- (d) After the 22nd transmission round, is segment loss detected by a triple duplicate ACK or by a timeout?
- (e) What is the initial value of ssthresh at the first transmission round?
- (f) What is the value of ssthresh at the 18th transmission round?
- (g) What is the value of ssthresh at the 24th transmission round?
- (h) During what transmission round is the 70th segment sent?
- (i) Assuming a packet loss is detected after the 26th round by the receipt of a triple duplicate ACK, what will be the values of the congestion window size and of ssthresh?
- (j) Suppose TCP Tahoe is used (instead of TCP Reno), and assume that triple duplicate ACKs are received at the 16th round. What are the ssthresh and the congestion window size at the 19th round?
- (k) Again suppose TCP Tahoe is used, and there is a timeout event at 22nd round. How many packets have been sent out from 17th round till 22nd round, inclusive.

Solution. a) Slow start is operating from 1 to 6 and from 23 to 26 transmission rounds where the congestion window (cwnd) increases exponentially

- b) Congestion avoidance is operating from 6 to 16 and from 17 to 23 where cwnd increases linearly
- c) After the 16th transmission round, the segment loss is detected using a triple duplicate ACK, because we see that the cwnd has not decreased to zero (decreased to half), which happens after a fast retransmission, which in turn happens after receiving 3 duplicate confirmations
- d) After the 22nd transmission round, the segment loss is detected using a timeout, because we see that the cwnd has decreased to zero, which happens after timeout
- e) The initial value of ssthresh at the first transmission round is 32 because after cwnd == 32 congestion avoidance starts
- f) The value of ssthresh at the 18th transmission round is cwnd /2 = 42 / 2 = 21 (that is why from 16 cwnd grows linearly, cwnd starts there from 24)
- g) The value of ssthresh at the 24th transmission round is cwnd /2 = 28 / 2 = 14 (that is why from 23 cwnd grows exponentially)
- h) The 70th segment sent during 7th transmission round because 1 + 2 + 4 + 8 + 16 + 32 + 33 = 96 segments > 70 segments (each term is one round)
- i) Assuming a packet loss is detected after the 26th round by the receipt of a triple duplicate ACK, the values of the congestion window size and of ssthresh will be ssthresh = cwnd/2 = 8 / 2 = 4; cwnd = ssthresh = 4
- j) Supposing TCP Tahoe is used (instead of TCP Reno), and assuming that triple duplicate ACKs are received at the 16th round, the ssthresh and the congestion window size at the 19th round are 21 and 4 respectfully. Because ssthresh is the same as with TCP Reno but congestion window size = 1 at 17th round and then slow start executed. 18th -> 2, 19th -> 4
- k) Supposing TCP Tahoe is used, and there is a timeout event at 22nd round, from 17th round till 22nd round, inclusive, have been sent 1 + 2 + 4 + 8 + 16 + 32 = 63 segments (packets). After round 22, the condition cwnd \geq stresh should be met, but it is not yet fulfilled at round 22, so exponential growth

Task 4. Assume that TCP implements an extension that allows window sizes much larger than 64 KB. Suppose that you are using this extended TCP over a 1-Gbps link with a latency of 50 ms to transfer a 10-MB file, and the TCP receive window is 1 MB. If TCP sends 1-KB packets (assuming no congestion and no lost packets):

- (a) How many RTTs does it take until slow start opens the send window to 1 MB?
- (b) How many RTTs does it take to send the file?
- (c) If the time to send the file is given by the number of required RTTs multiplied by the link latency, what is the effective throughput for the transfer? What percentage of the link bandwidth is utilized?

Solution. (a) RTTs until window opens to 1 ${\rm MB}$ Starting from 1 MSS (1 KB), each RTT the window doubles until it hits 1 MB: $log_2(\frac{1MB}{1KB}) = 10$ RTT's

(b) RTTs to send the file

Once the window is 1 MB, sending a 10 MB file will take: $\frac{10MB}{1MB}=10$ RTT's Total RTTs: 10+10=20 RTT's

(c) Effective throughput

Time to send the file: $20RTT's \cdot 50ms = 1$ second

Effective throughput: 10 MB Percentage of bandwidth: $\frac{10MB}{125MB} \cdot 100 = 8 \%$

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