

1.

Since most firewalls are configured to block UDP traffic, using TCP for video and voice traffic lets the traffic through the firewalls.

2.

Yes. The application developer can put reliable data transfer into the application layer protocol. This would require a significant amount of work and debugging, however.

3.

A timer would still be necessary in the protocol rdt 3.0. If the round trip time is known then the only advantage will be that, the sender knows for sure that either the packet or the ACK (or NACK) for the packet has been lost, as compared to the real scenario, where the ACK (or NACK) might still be on the way to the sender, after the timer expires. However, to detect the loss, for each packet, a timer of constant duration will still be necessary at the sender.

4.

a) 20 bytes

b) ack number = 90

5.

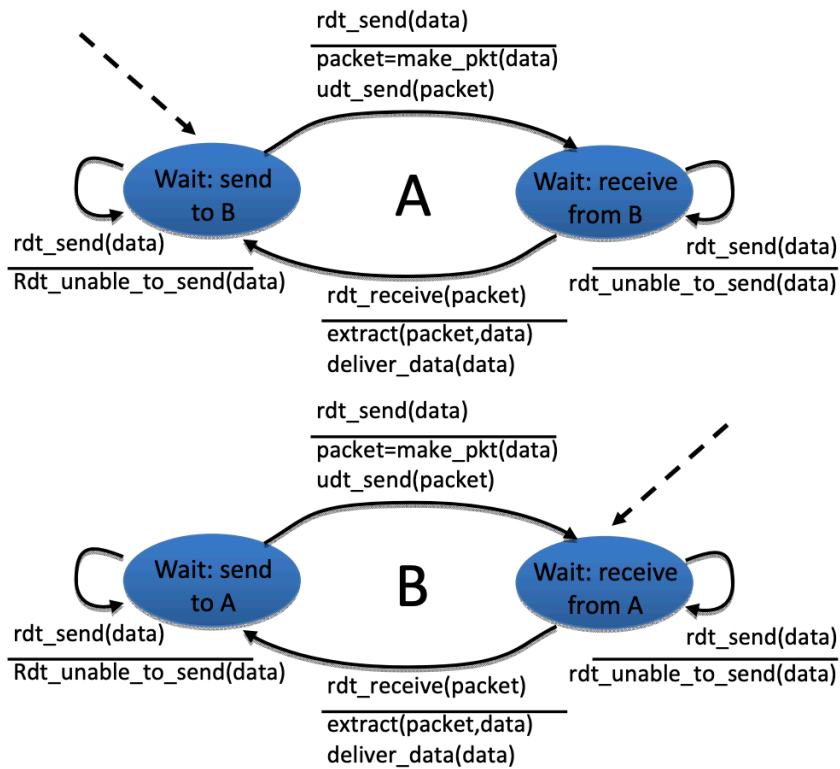
To detect errors, the receiver adds the four words (the three original words and the checksum). If the sum contains a zero, the receiver knows there has been an error. All one-bit errors will be detected, but two-bit errors can be undetected (e.g., if the last digit of the first word is converted to a 0 and the last digit of the second word is converted to a 1).

6.

a) Here we have a window size of  $N=3$ . Suppose the receiver has received packet  $k-1$ , and has ACKed that and all other preceding packets. If all of these ACK's have been received by sender, then sender's window is  $[k, k+N-1]$ . Suppose next that none of the ACKs have been received at the sender. In this second case, the sender's window contains  $k-1$  and the  $N$  packets up to and including  $k-1$ . The sender's window is thus  $[k-N, k-1]$ . By these arguments, the sender's window is of size 3 and begins somewhere in the range  $[k-N, k]$ .

b) If the receiver is waiting for packet  $k$ , then it has received (and ACKed) packet  $k-1$  and the  $N-1$  packets before that. If none of those  $N$  ACKs have been yet received by the sender, then ACK messages with values of  $[k-N, k-1]$  may still be propagating back. Because the sender has sent packets  $[k-N, k-1]$ , it must be the case that the sender has already received an ACK for  $k-N-1$ . Once the receiver has sent an ACK for  $k-N-1$  it will never send an ACK that is less than  $k-N-1$ . Thus the range of inflight ACK values can range from  $k-N-1$  to  $k-1$ .

7.



8.

- TCP slowstart is operating in the intervals [1,6] and [23,26]
- TCP congestion avoidance is operating in the intervals [6,16] and [17,22]
- After the 16th transmission round, packet loss is recognized by a triple duplicate ACK. If there was a timeout, the congestion window size would have dropped to 1.
- After the 22nd transmission round, segment loss is detected due to timeout, and hence the congestion window size is set to 1.
- The threshold is initially 32, since it is at this window size that slow start stops and congestion avoidance begins.
- The threshold is set to half the value of the congestion window when packet loss is detected. When loss is detected during transmission round 16, the congestion windows size is 42. Hence the threshold is 21 during the 18th transmission round.
- The threshold is set to half the value of the congestion window when packet loss is detected. When loss is detected during transmission round 22, the congestion windows size is 29. Hence the threshold is 14 (taking lower floor of 14.5) during the 24th transmission round.
- During the 1st transmission round, packet 1 is sent; packet 2-3 are sent in the 2nd transmission round; packets 4-7 are sent in the 3rd transmission round; packets 8-15 are sent in the 4th transmission round; packets 16-31 are sent in the 5th

transmission round; packets 32-63 are sent in the 6th transmission round; packets 64 – 96 are sent in the 7th transmission round. Thus packet 70 is sent in the 7th transmission round.

- i) The threshold will be set to half the current value of the congestion window (8) when the loss occurred and congestion window will be set to the new threshold value + 3 MSS . Thus the new values of the threshold and window will be 4 and 7 respectively.
- j) threshold is 21, and congestion window size is 1.
- k) round 17, 1 packet; round 18, 2 packets; round 19, 4 packets; round 20, 8 packets; round 21, 16 packets; round 22, 21 packets. So, the total number is 52.