

1.

Check all of the services below that are provided by the UDP protocol.

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- ☐ A congestion control service to ensure that multiple senders do not overload network links.
- ☒ A message abstraction, that preserves boundaries between message data sent in different socket send calls at the sender.
- ☐ A guarantee on the *minimum* amount of throughput that will be provided between sender and receiver.
- ☐ A flow-control service that ensures that a sender will not send at such a high rate so as to overflow receiving host buffers.
- ☐ A byte stream abstraction, that does not preserve boundaries between message data sent in different socket send calls at the sender.
- ☐ Reliable data delivery.
- ☐ A guarantee on the maximum amount of time needed to deliver data from sender to receiver.
- ☐ In-order data delivery

That's Correct!

## TRANSPORT LAYER SERVICES USING TCP.

Check all of the services below that are provided by the TCP protocol.

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- ☐ A message abstraction, that preserves boundaries between message data sent in different socket send calls at the sender.
- ☒ A flow-control service that ensures that a sender will not send at such a high rate so as to overflow receiving host buffers.
- ☒ Reliable data delivery.
- ☒ A congestion control service to ensure that multiple senders do not overload network links.
- ☒ In-order data delivery
- ☐ A guarantee on the maximum amount of time needed to deliver data from sender to receiver.
- ☐ A guarantee on the *minimum* amount of throughput that will be provided between sender and receiver.
- ☒ A byte stream abstraction, that does not preserve boundaries between message data sent in different socket send calls at the sender.

That's Correct!

2.

## UDP SEGMENT LENGTH FIELD.

Why is the UDP header length field needed?

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- ☐ (a) and (b) above
- ☒ Because the payload section can be of variable length, and this lets UDP know where the segment ends.
- ☐ Because this field is needed in TCP as well.
- ☐ To make the header and even number of bytes

That's Correct!

3.

## WHAT IS A CHECKSUM?

Which of the following statements are true about a checksum? Hint: more than one statement is true.

- ☒ A checksum is computed at a sender by considering each byte within a packet as a number, and then adding these numbers (each number representing a bytes) together to compute a sum (which is known as a checksum).
- ☒ The sender-computed checksum value is often included in a checksum field within a packet header.
- ☐ The receiver of a packet with a checksum will add up the received bytes, just as the sender did, and compare this locally-computed checksum with the checksum value in the packet header. If these two values are the **same** then the receiver **knows** that all of the bits in the received packet are correct, i.e., that no bits have been changed during transmission from sender to receiver.
- ☒ The receiver of a packet with a checksum field will add up the received bytes, just as the sender did, and compare this locally-computed checksum with the checksum value in the packet header. If these two values are **different** then the receiver **knows** that one of the bits in the received packet has been changed during transmission from sender to receiver.

That's Correct!

## UDP HEADER FIELDS.

Which of the fields below are in a UDP segment header? [Hint: note the use of the word "header" in this question statement.]

- ☒ Destination port number
- ☐ Data (payload)
- ☐ Source IP address
- ☐ Upper layer protocol
- ☒ Length (of UDP header plus payload)
- ☒ Internet checksum
- ☐ Sequence number
- ☒ Source port number

That's Correct!



4.

Consider the purposes/goals/use of different reliable data transfer protocol mechanisms. For the given purpose/goal/use match it to the RDT mechanism that is used to implement the given purpose/goal/use.

#### QUESTION LIST:

Lets the sender know that a packet was NOT received correctly at the receiver.

Used by sender or receiver to detect bits flipped during a packet's transmission.

Allows for duplicate detection at receiver.

Lets the sender know that a packet was received correctly at the receiver.

Allows the receiver to eventually receive a packet that was corrupted or lost in an earlier transmission.

#### ANSWER LIST:

A. NAK

B. Sequence numbers

C. Checksum

D. ACK

E. Retransmission

## PIPELINING.

Which of the following statements about pipelining are true? One or more statements may be true.

- ☒ With a pipelined sender, there may be transmitted packets "in flight" – propagating through the channel – packets that the sender has sent but that the receiver has not yet received.
- ☐ With pipelining, a receiver will have to send fewer acknowledgments as the degree of pipelining increases
- ☒ A pipelined sender can have transmitted multiple packets for which the sender has yet to receive an ACK from the receiver.
- ☐ With pipelining, a packet is only retransmitted if that packet, or its ACK, has been lost.

That's Correct!



## PACKET BUFFERING IN GO-BACK-N.

What are some reasons for discarding received-but- out-of-sequence packets at the receiver in GBN? Indicate one or more of the following statements that are correct.

- ☐ If some packets are in error, then its likely that other packets are in error as well.
- ☐ Discarding an out of sequence packet will really force the sender to retransmit.
- ☒ The implementation at the receiver is simpler.
- ☒ The sender will resend that packet in any case.

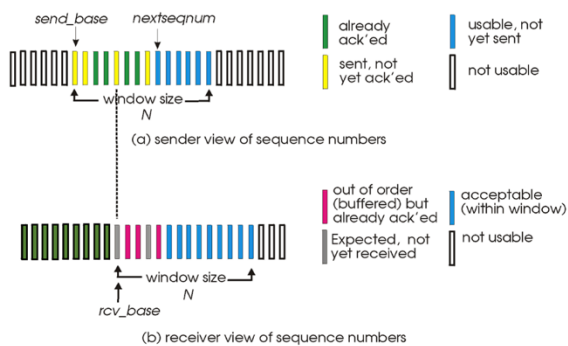
That's Correct!



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In SR, why does the receiver have to acknowledge packets with sequence numbers that are less than (and to the left of) those in its window, which starts at *rcv\_base*.

### Selective repeat: sender, receiver windows



- ☐ Actually, this ACK retransmission can be ignored and the protocol will still function correctly, but its performance won't be as good.
- ☐ Because, at the time of the data packet arrival at the receiver, the sender has definitely still not received an ACK for that packet.
- ☒ Because the sender may not have received an ACK for that packet yet.

That's Correct!



What are some reasons for *not* discarding received-but- out-of-sequence packets at the receiver in GBN? Indicate one or more of the following statements that are correct.

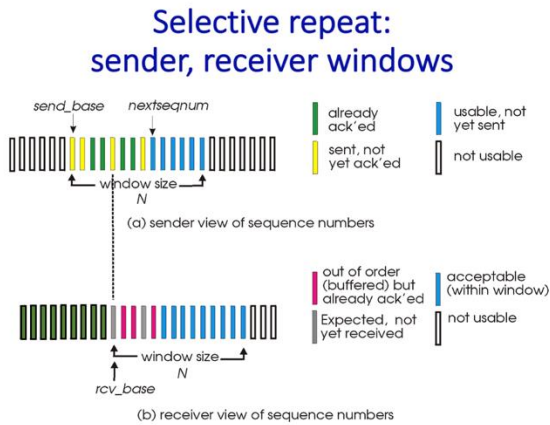
- ☒ Even though that packet will be retransmitted, its next retransmission could be corrupted, so don't discard a perfectly well-received packet, silly!
- ☐ Complex protocols are always better.
- ☐ By not discarding, the receiver can implicitly let the sender know that it (the sender) does not necessarily have to retransmit that packet.

That's Correct!



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In the SR receiver window (see diagram below, taken from PPT slides and video), why haven't the red packets been delivered yet? Check the one or more reasons below that apply.



- ☐ Red packets have a lower delivery priority up to the application.
- ☒ There is a packet with a lower sequence number than any of the red packets that has yet to be received, so in-order delivery of data in the red packets up to the application layer is not possible.
- ☐ There is a packet with a higher sequence number than any of the red packets that has yet to be received, so in-order delivery of data in the red packets to the application layer is not yet possible.

That's Correct!

