Communication Networks (0368-3030) / Spring 2011

The Blavatnik School of Computer Science,
Tel-Aviv University

Allon Wagner

Reliable Data Transfer

Kurose & Ross, Chapter 3.4 (5th ed.)

Many slides adapted from:

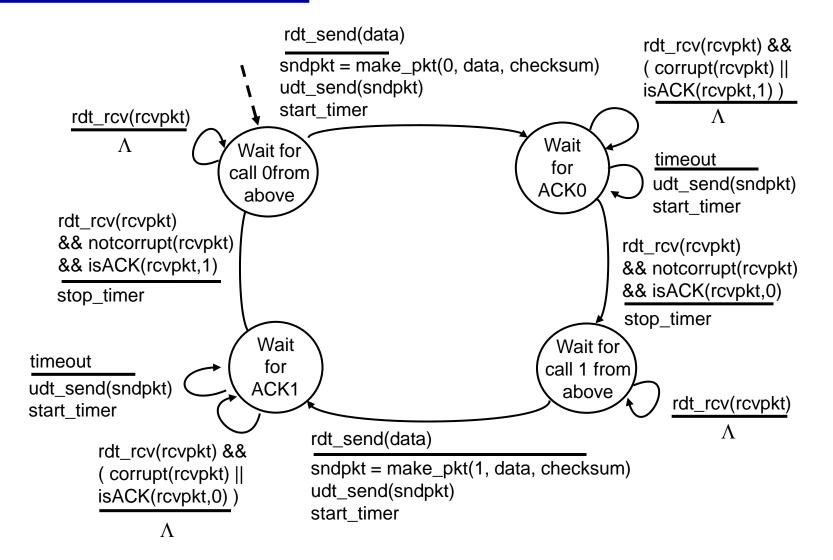
J. Kurose & K. Ross \

Computer Networking: A Top Down Approach (5th ed.)

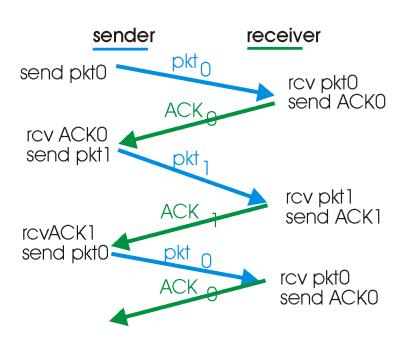
Addison-Wesley, April 2009.

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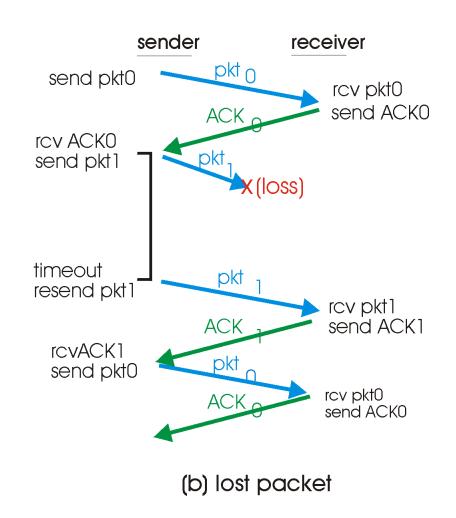
rdt3.0 sender



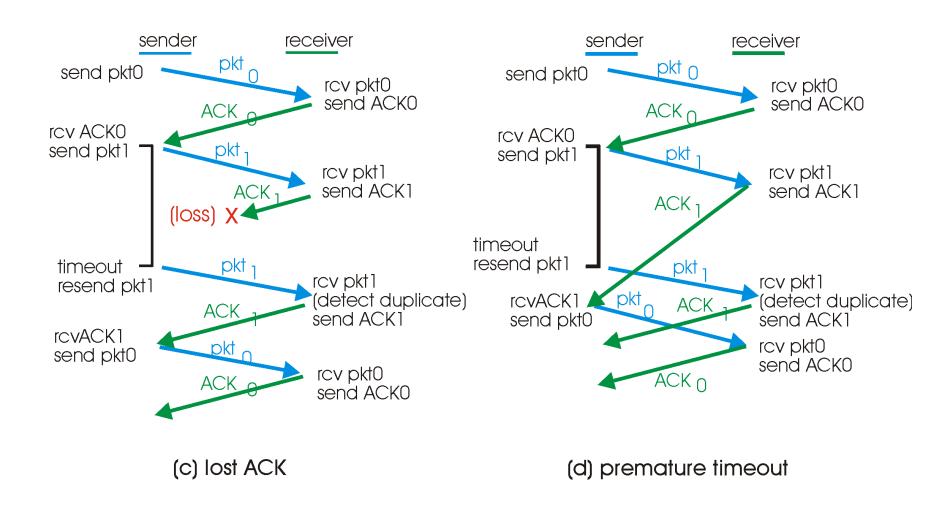
rdt3.0 in action



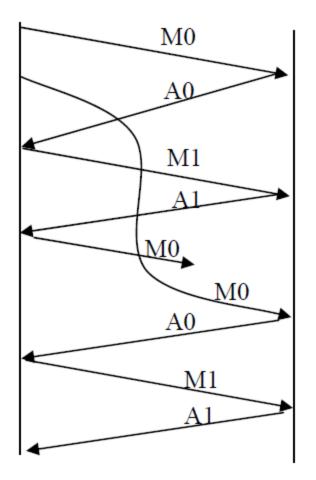
(a) operation with no loss



rdt3.0 in action



- rdt 3.0 is correct only under a FIFO channel assumption.
 - Correct = guarantees reliable transmission. Data sent by sender is exactly the data reconstructed in the receiver side.
- Show a case where a non-FIFO channel (i.e., one that can cause packet reordering) causes rdt 3.0 to deliver incorrect data.



old version of M0 accepted!

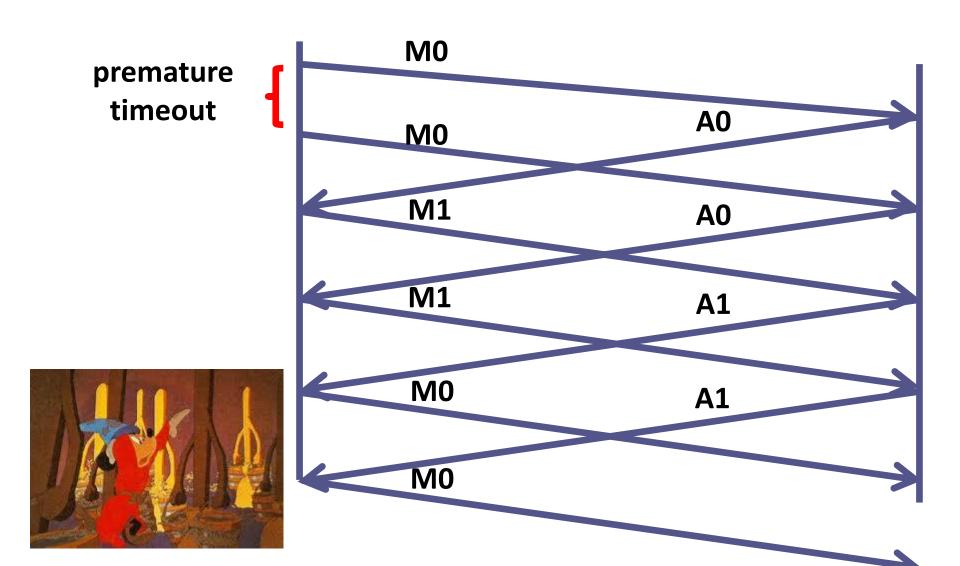
Transport Layer

- The sender of rdt 3.0 simply ignores all received packets that are either in error or have the wrong value in the acknum field of an ack packet.
- Suppose that in such circumstances, rdt 3.0 were simply to transmit the current data packet.
- Would the protocol still work?
- Would it be more or less efficient than before?

- Would the protocol still work?
 - Yes. A retransmission is exactly what would happen if the sender's timeout expired (for instance, because an ack was completely lost instead of garbled).
 - The receiver can't even distinguish between the two events.

- Would it be more or less efficient than before?
 - Depends on the length of the sender timeout,
 compared to the expected channel delay.
 - If the timeout is very long, then the immediate retransmit can save us the long wait until the timeout expires.
 - However, premature timeouts can cause a pathologies.

- Would it be more or less efficient than before?
- We will show a scenario in which one premature timeout causes duplication of all the packets in the session from a certain time point.
- This is the "Sorcerer's Apprentice Syndrome"



Performance of rdt3.0

- rdt3.0 works, but performance stinks
- * ex: 1 Gbps link, 15 ms prop. delay, 8000 bit packet:

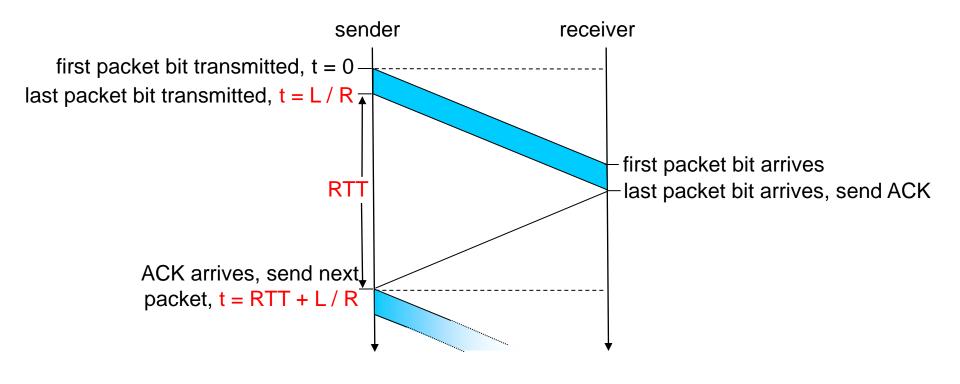
$$d_{trans} = \frac{L}{R} = \frac{8000 \text{bits}}{10^9 \text{bps}} = 8 \text{ microseconds}$$

U sender: utilization - fraction of time sender busy sending

$$U_{\text{sender}} = \frac{L/R}{RTT + L/R} = \frac{.008}{30.008} = 0.00027$$

- if RTT=30 msec, 1KB pkt every 30 msec -> 33kB/sec thruput over 1 Gbps link
- network protocol limits use of physical resources!

rdt3.0: stop-and-wait operation

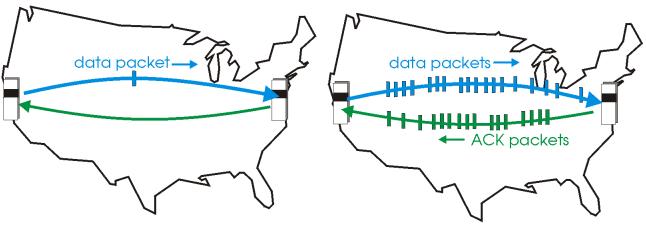


$$U_{\text{sender}} = \frac{L/R}{RTT + L/R} = \frac{.008}{30.008} = 0.00027$$

Pipelined protocols

pipelining: sender allows multiple, "in-flight", yet-tobe-acknowledged pkts

- range of sequence numbers must be increased
- buffering at sender and/or receiver

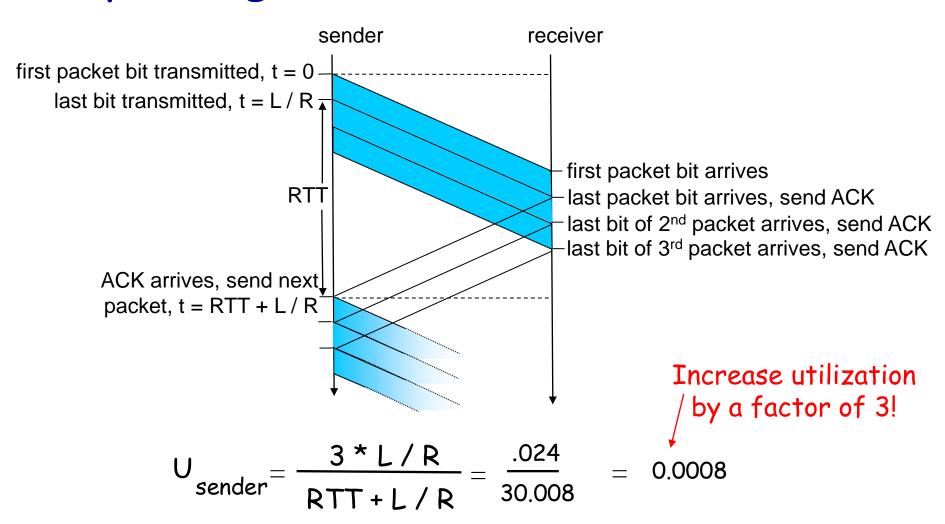


(a) a stop-and-wait protocol in operation

(b) a pipelined protocol in operation

two generic forms of pipelined protocols: go-Back-N, selective repeat

Pipelining: increased utilization



Pipelined Protocols

Go-back-N: big picture:

- * sender can have up to N unacked packets in pipeline
- rcvr only sends cumulative acks
 - doesn't ack packet if there's a gap
- * sender has timer for oldest unacked packet
 - if timer expires, retransmit all unack'ed packets

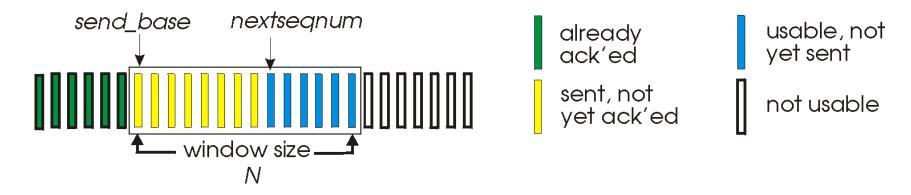
Selective Repeat: big pic

- * sender can have up to N unack'ed packets in pipeline
- * rcvr sends individual ack for each packet
- sender maintains timer for each unacked packet
 - when timer expires, retransmit only unack'ed packet

Go-Back-N

Sender:

- k-bit seq # in pkt header
- "window" of up to N, consecutive unack'ed pkts allowed

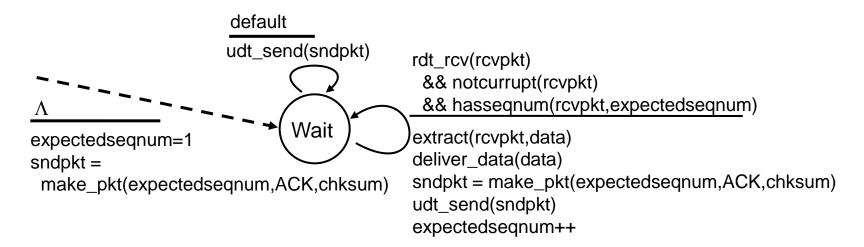


- ACK(n): ACKs all pkts up to, including seq # n "cumulative ACK"
 - may receive duplicate ACKs (see receiver)
- * timer for each in-flight pkt
- timeout(n): retransmit pkt n and all higher seq # pkts in window

GBN: sender extended FSM

```
rdt send(data)
                       if (nextseqnum < base+N) {
                          sndpkt[nextseqnum] = make_pkt(nextseqnum,data,chksum)
                          udt send(sndpkt[nextsegnum])
                          if (base == nextsegnum)
                            start_timer
                          nextseqnum++
                       else
   Λ
                         refuse_data(data)
  base=1
  nextseqnum=1
                                           timeout
                                           start timer
                             Wait
                                           udt_send(sndpkt[base])
                                           udt send(sndpkt[base+1])
rdt_rcv(rcvpkt)
 && corrupt(rcvpkt)
                                           udt_send(sndpkt[nextsegnum-1])
                         rdt_rcv(rcvpkt) &&
                           notcorrupt(rcvpkt)
                         base = getacknum(rcvpkt)+1
                         If (base == nextseqnum)
                           stop_timer
                          else
                            start_timer
```

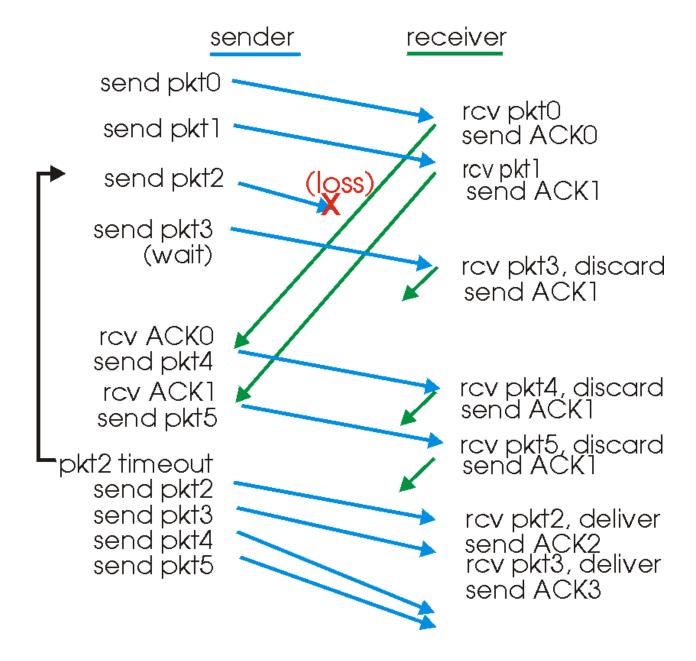
GBN: receiver extended FSM



ACK-only: always send ACK for correctly-received pkt with highest in-order seq

- may generate duplicate ACKs
- need only remember expectedseqnum
- out-of-order pkt:
 - discard (don't buffer) -> no receiver buffering!
 - Re-ACK pkt with highest in-order seq #

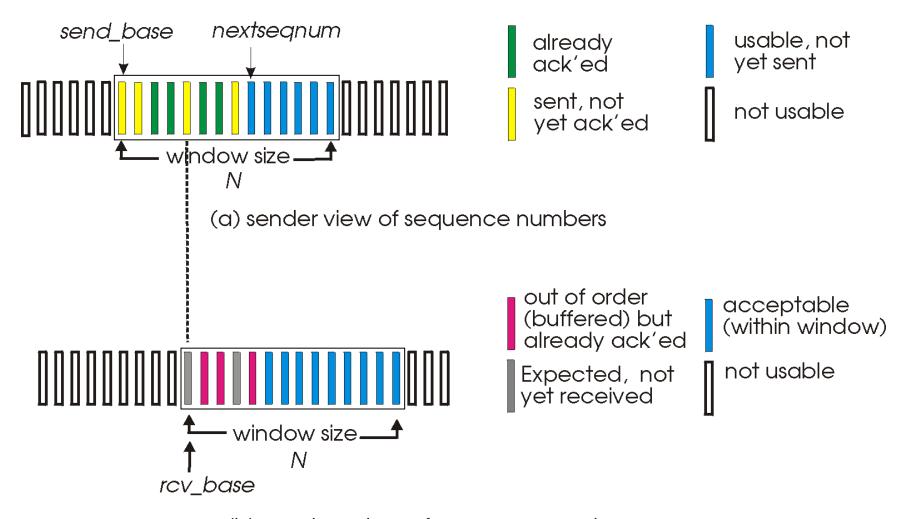
GBN in action



Selective Repeat

- receiver individually acknowledges all correctly received pkts
 - buffers pkts, as needed, for eventual in-order delivery to upper layer
- sender only resends pkts for which ACK not received
 - sender timer for each unACKed pkt
- sender window
 - N consecutive seq #'s
 - again limits seq #s of sent, unACK'ed pkts

Selective repeat: sender, receiver windows



(b) receiver view of sequence numbers

Selective repeat

-sender

data from above:

if next available seq # in window, send pkt

timeout(n):

resend pkt n, restart timer

ACK(n) in [sendbase,sendbase+N]:

- mark pkt n as received
- if n smallest unACKed pkt, advance window base to next unACKed seq #

receiver

pkt n in [rcvbase, rcvbase+N-1]

- send ACK(n)
- out-of-order: buffer
- in-order: deliver (also deliver buffered, in-order pkts), advance window to next not-yet-received pkt

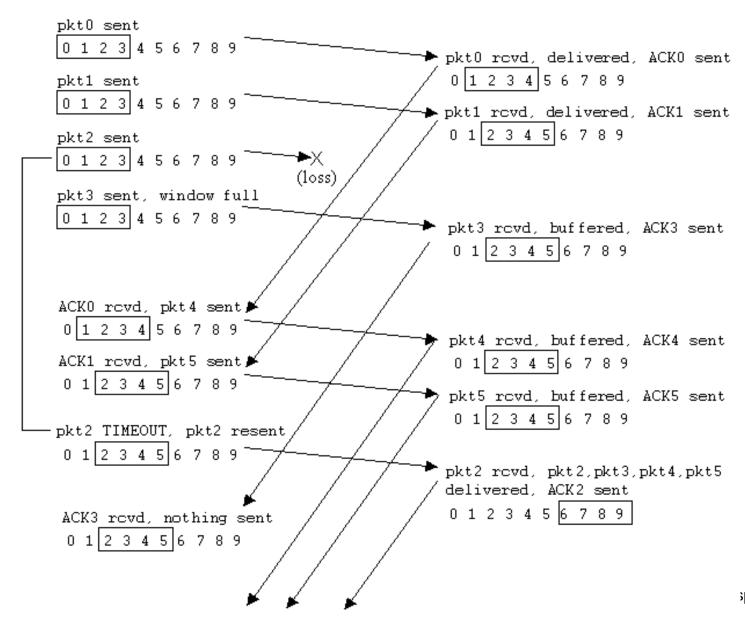
pkt n in [rcvbase-N,rcvbase-1]

* ACK(n)

otherwise:

ignore

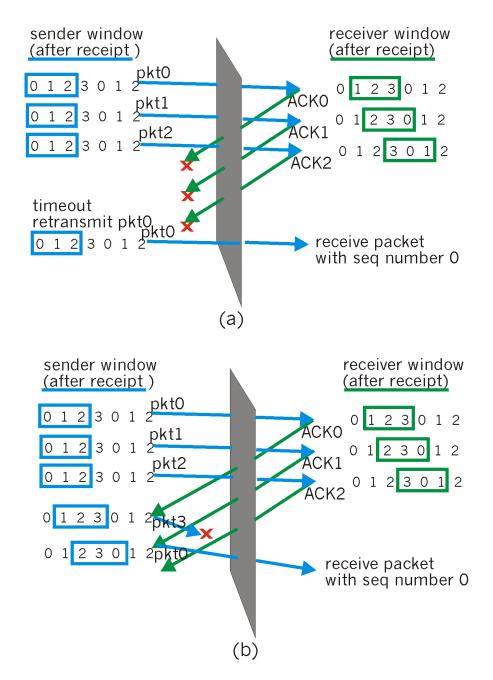
Selective repeat in action



Selective repeat: dilemma

Example:

- seq #'s: 0, 1, 2, 3
- window size=3
- * receiver sees no difference in two scenarios!
- incorrectly passes duplicate data as new in (a)
- Q: what relationship between seq # size and window size?



Transport Layer

Minimal sequence range

- Assume we want to use a sender window of size N.
- What is the minimal number of unique sequence numbers we should allow to prevent such errors?
- The cyclic sequence number should never cause the sender and receiver's window to ambiguously overlap
- In FIFO channels:
 - □ GBN: *N* + 1
 - □ SR: 2*N*
 - Proof: on-board

Minimal sequence range (cont.)

- In non-FIFO channel, this cannot be guaranteed!
 - We assume that in realistic channels, old packets are cleared from the network after a reasonable time, so accidental overlap does not occur of the range of sequence numbers is "big enough".

- Are the following statements true or false?
- With SR, it is possible for the sender to receive an ACK for a packet that falls outside of its current window.
- True. Suppose sender has a window size of 3.
 - Time t_0 : it sends packets 1, 2, 3.
 - Time $t_1 > t_0$: receiver acks 1, 2, 3.
 - Time $t_2 > t_1$: sender times out and retransmits 1, 2, 3.
 - Time $t_3 > t_2$: receiver gets the duplicates and reacks 1, 2, 3.
 - Time $t_4 > t_3$: sender gets the ack sent at t_1 , advances its window to 4, 5, 6.
 - Time $t_5 > t_4$: sender receives the acks sent at t_2 , that fall outside of its current window.
- With GBN, it is possible for the sender to receive an ACK for a packet that falls outside of its current window.
- True, with the same scenario as described above. Only need to replace the selective acks with cumulative acks.

- Are the following statements true or false?
- rdt 3.0 is the same as SR with a sender and receiver window size of 1.
- rdt 3.0 is the same as GBN with a sender and receiver window size of 1.
- Both are true. With a window size of 1, SR, GBN, and the rtd 3.0 are functionally equivalent.
 - The window size of 1 precludes the possibility of out-oforder packets (within the window).
 - A cumulative ACK is just an ordinary ACK in this situation, since it can only refer to the single packet within the window.

Exercise

- Recall the GBN receiver: assume it is waiting for packet m (i.e., it received correctly all the packets up to m-1 inclusive).
 - When a data packet with sequence n=m is received, the receiver accepts it and advances its window.
 - Whenever a data packet with sequence $n \neq m$ is received, the receiver discards it and resends ack m ("I am still waiting for m").
- Assume a FIFO channel and an infinite sequence number. Does the protocol remain correct if we perform the following changes?
- If n < m the receiver discards the packet and does not send an ack. Otherwise, operate as before.
- Incorrect. Let the sender send packets 1, ..., m-1. All received correctly, but all acks are lost.
 - The receiver waits for packet m.
 - But whenever the sender times-out expires, it resends packets 1, ..., m-1.
 - Receiver discards them and does not ack.
 - Deadlock.

Exercise

- if n > m, the receiver discards the packet and does not send an ack. Otherwise, operate as before.
- Correct. If n > m was received, but the receiver is waiting for m, it means we have a gap. The sender will eventually timeout for m, and resend packet n then.