rdt3.0: channels with errors and loss

new assumption:

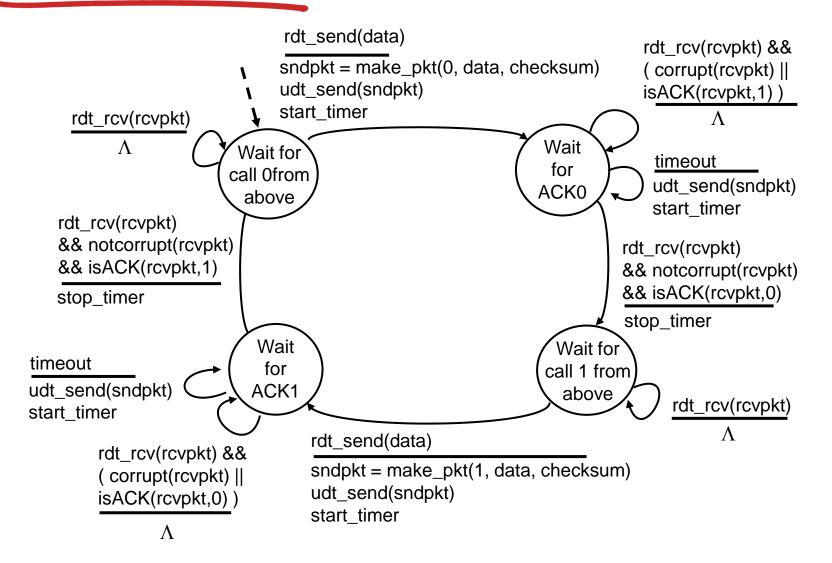
underlying channel can also lose packets (data, ACKs)

checksum, seq. #,
 ACKs, retransmissions
 will be of help ... but
 not enough

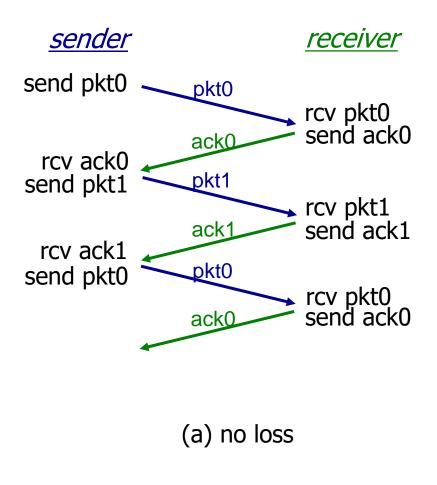
approach: sender waits "reasonable" amount of time for ACK

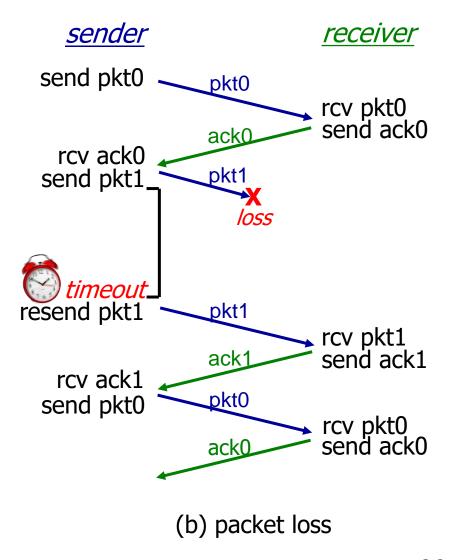
- retransmits if no ACK received in this time
- if pkt (or ACK) just delayed (not lost):
 - retransmission will be duplicate, but seq. #'s already handles this
 - receiver must specify seq # of pkt being ACKed
- requires countdown timer

rdt3.0 sender

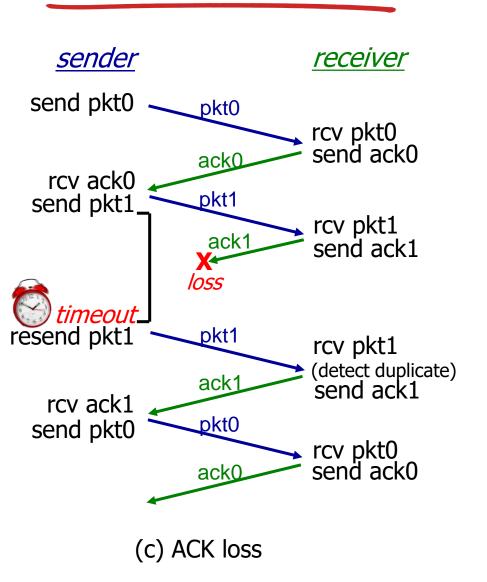


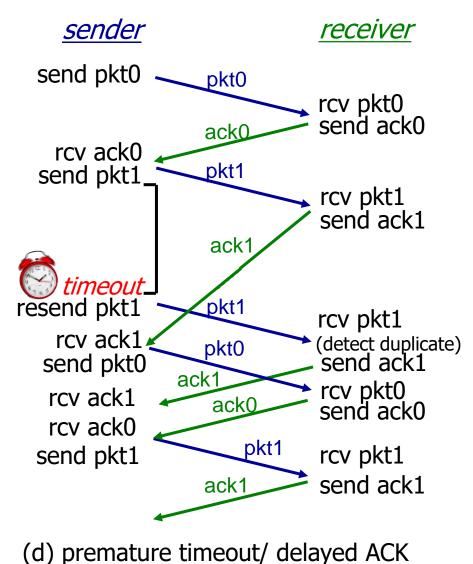
rdt3.0 in action





rdt3.0 in action





Performance of rdt3.0

- rdt3.0 is correct, but performance is poor
- e.g.: I Gbps link, 15 ms prop. delay, 8000 bit packet:

$$D_{trans} = \frac{L}{R} = \frac{8000 \text{ bits}}{10^9 \text{ bits/sec}} = 8 \text{ microsecs}$$

U sender: utilization – fraction of time sender busy sending

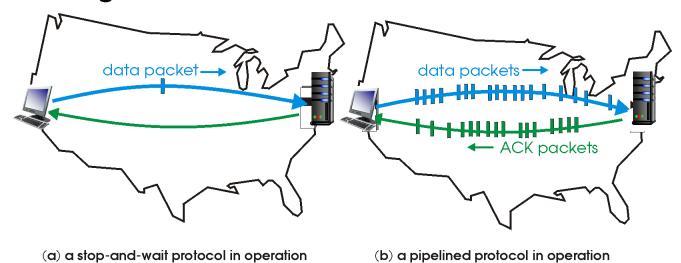
$$U_{\text{sender}} = \frac{L/R}{RTT + L/R} = \frac{.008}{30.008} = 0.00027$$

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

Pipelined protocols

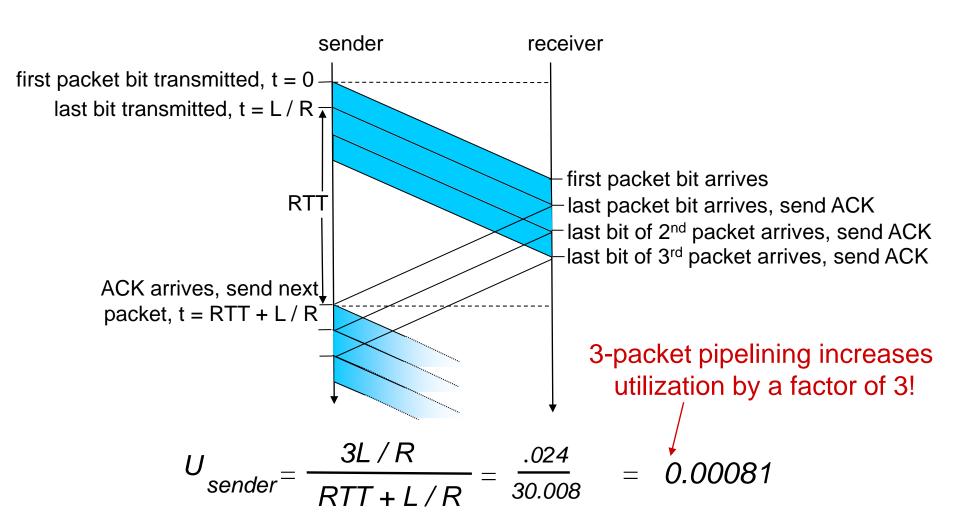
pipelining: sender allows multiple, "in-flight", yetto-be-acknowledged pkts

- range of sequence numbers must be increased
- buffering at sender and/or receiver. What for?



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

Pipelining: increased utilization



Pipelined protocols: overview

Go-back-N:

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

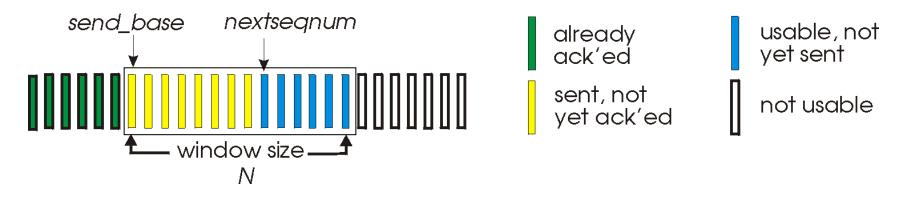
Selective Repeat:

- 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 that unacked packet

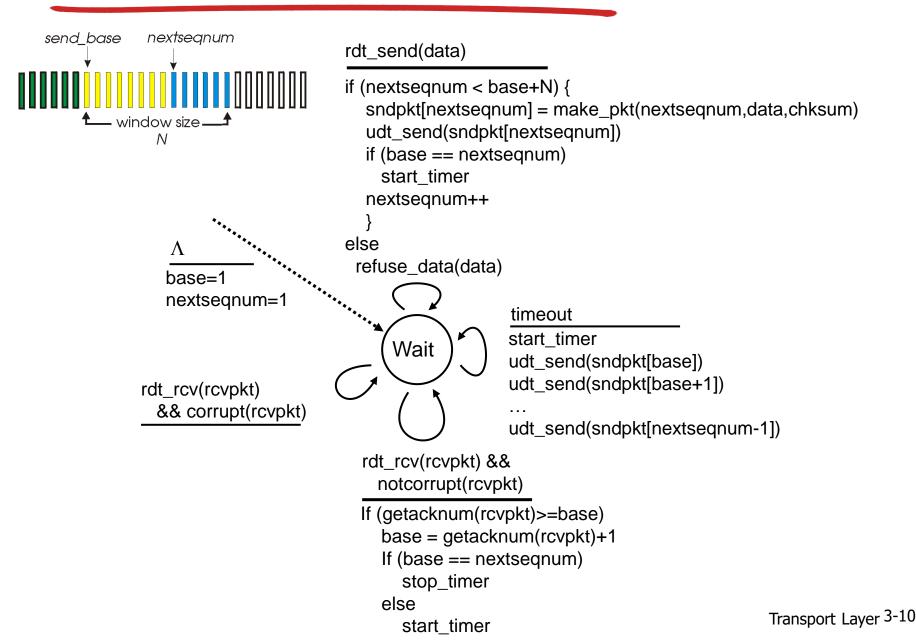
Go-Back-N: sender

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

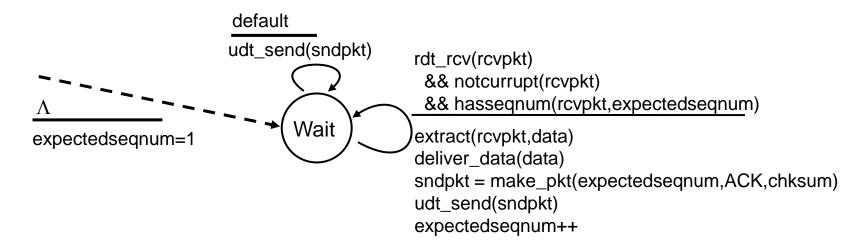


- ACK(n):ACKs all pkts up to, including seq # n "cumulative ACK"
 - may receive duplicate ACKs (see receiver)
- timer for oldest in-flight pkt
- timeout(n): retransmit packet n and all higher seq # pkts in window

GBN: sender extended FSM



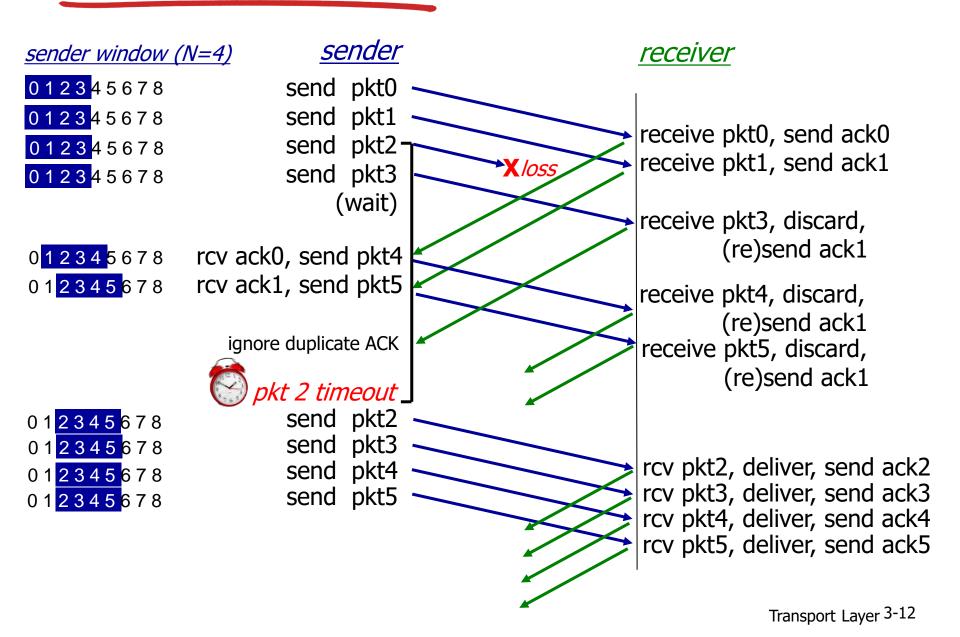
GBN: receiver extended FSM



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 #

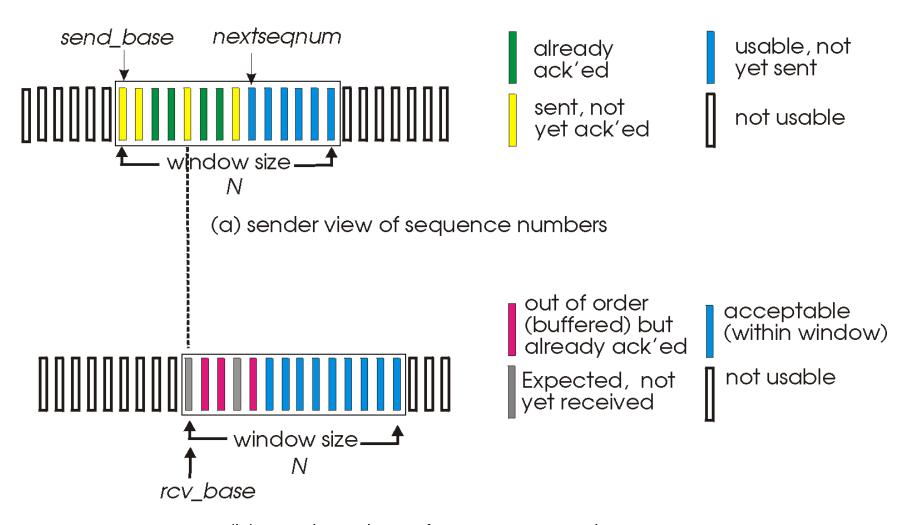
GBN in action



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
 - limits seq #s of sent, unACKed 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-1]:

- 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

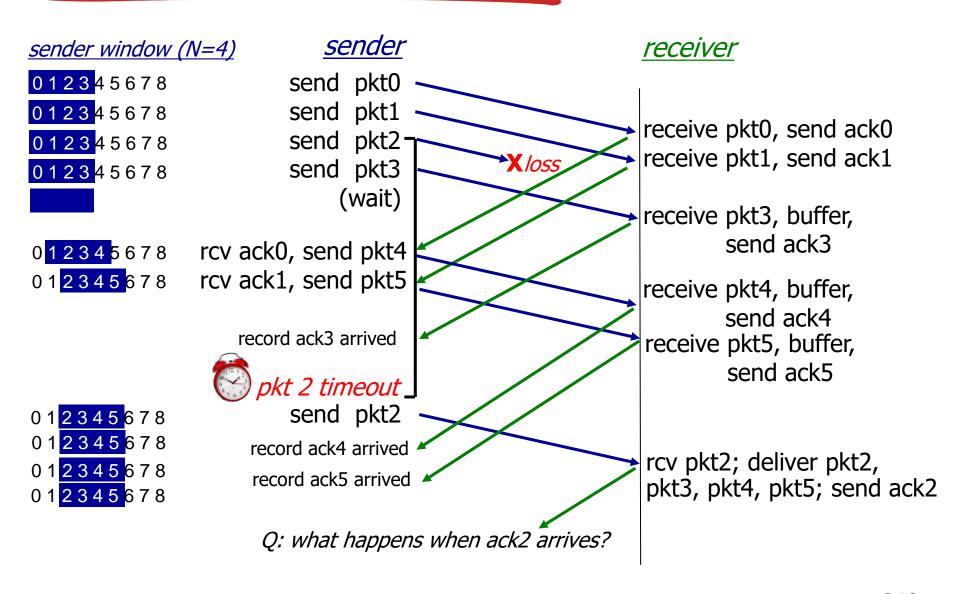
pkt n in [rcvbase-N,rcvbase-I]

ACK(n)

otherwise:

ignore

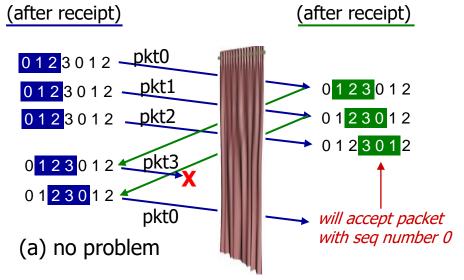
Selective repeat in action



Selective repeat: dilemma

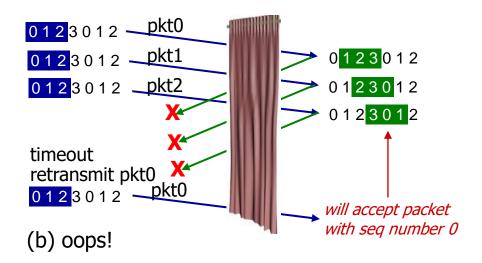
example:

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



sender window

receiver can't see sender side.
receiver behavior identical in both cases!
something's (very) wrong!



receiver window

Transport Layer: outline

- 3.1 transport-layer services
- 3.2 multiplexing and demultiplexing
- 3.3 connectionless transport: UDP
- 3.4 principles of reliable data transfer

- 3.5 connection-oriented transport: TCP
 - segment structure
 - reliable data transfer
 - flow control
 - connection management
- 3.6 principles of congestion control
- 3.7 TCP congestion control

TCP: Overview RFCs: 793,1122,1323, 2018, 2581

point-to-point:

- one sender, one receiver
- reliable, in-order byte stream:
 - no "message boundaries"
- pipelined:
 - TCP congestion and flow control set window size

full duplex data:

- bi-directional data flow in same connection
- MSS: maximum segment size

connection-oriented:

 handshaking (exchange of control msgs) inits sender, receiver state before data exchange

flow controlled:

sender will not overwhelm receiver

TCP segment structure

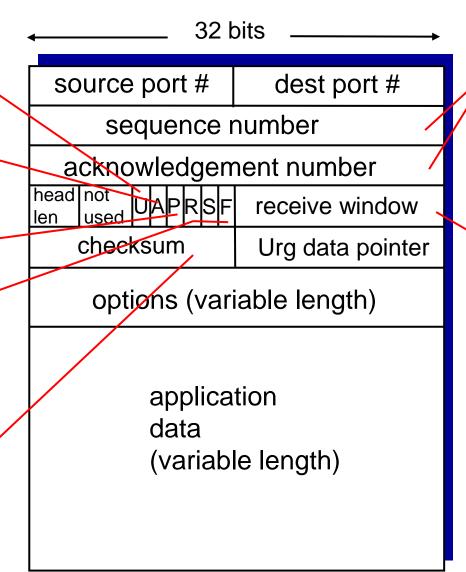
URG: urgent data (generally not used)

ACK: ACK # valid

PSH: push data now (generally not used)

RST, SYN, FIN: connection estab (setup, teardown commands)

Internet checksum (as in UDP)



by bytes of data (not segments!)

> # bytes rcvr willing to accept

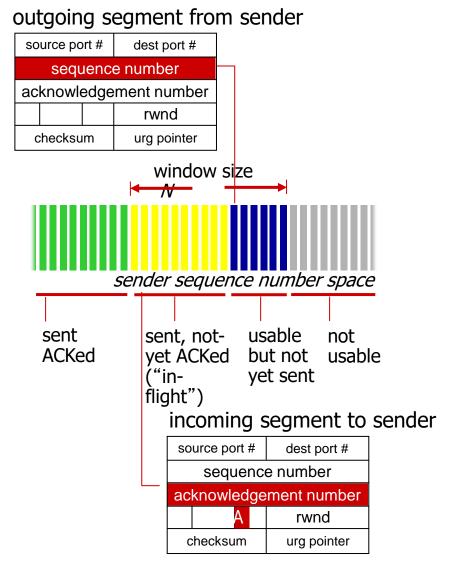
TCP seq. numbers, ACKs

sequence numbers:

byte stream "number" of first byte in segment's data

acknowledgements:

- seq # of next byte expected from other side
- cumulative ACK
- Q: how receiver handles out-of-order segments
 - A: TCP spec doesn't say,
 - up to implementor



TCP seq. numbers, ACKs

