

Chapter 3 Transport Layer

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Chapter 3 Outline of Lecture 8

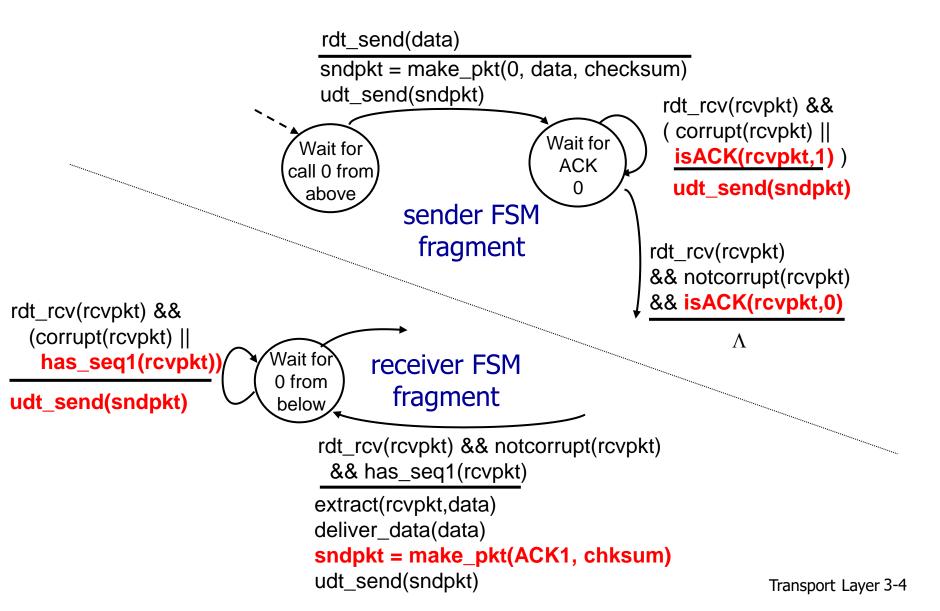
- 3.4 principles of reliable data transfer
 - □rdt2.2: a NAK-free protocol
 - rdt3.0: channels with errors and loss
 - ☐ Pipelined protocols



rdt2.2: a NAK-free protocol

- same functionality as rdt2.1, using ACKs only
- instead of NAK, receiver sends ACK for last pkt received OK
 - receiver must explicitly include seq # of pkt being ACKed
- duplicate ACK at sender results in same action as NAK: retransmit current pkt

rdt2.2: sender, receiver fragments





rdt3.0: channels with errors and loss

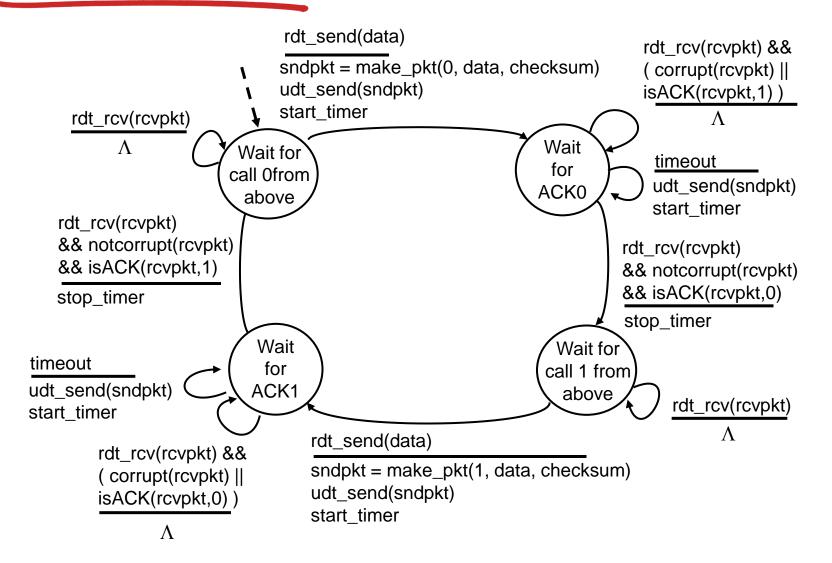
new assumption:

underlying channel can also lose packets (data, ACKs)

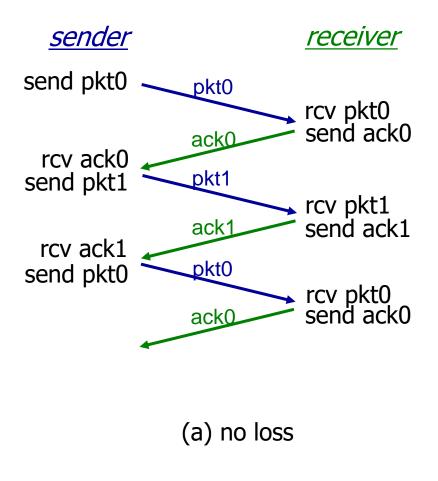
checksum, seq. #,
 ACKs, retransmissions
 will be of help ... but
 not enough

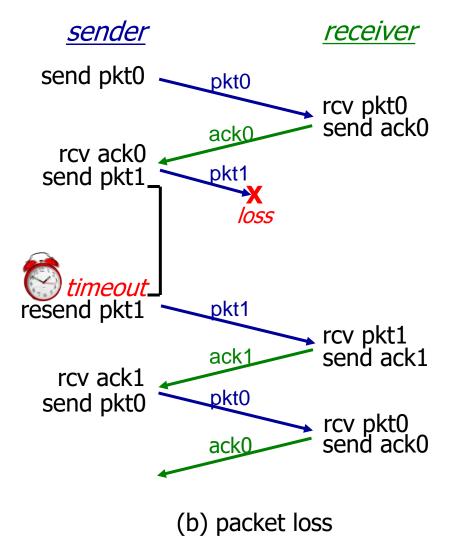
- approach: sender waits
 "reasonable" amount of
 time for ACK
- retransmits if no ACK received in this time
- if pkt (or ACK) just delayed (not lost):
 - retransmission will be duplicate, but seq. #'s already handles this
 - receiver must specify seq # of pkt being ACKed
- requires countdown timer

rdt3.0 sender

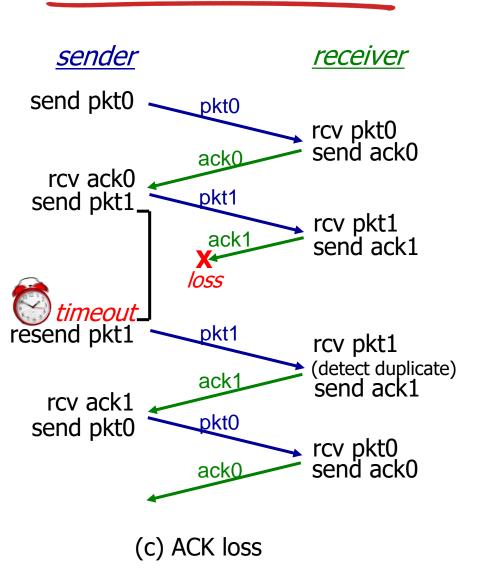


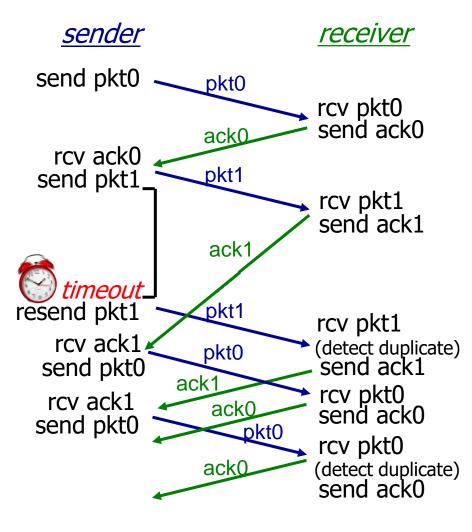
rdt3.0 in action





rdt3.0 in action





(d) premature timeout/ delayed ACK



Performance of rdt3.0

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

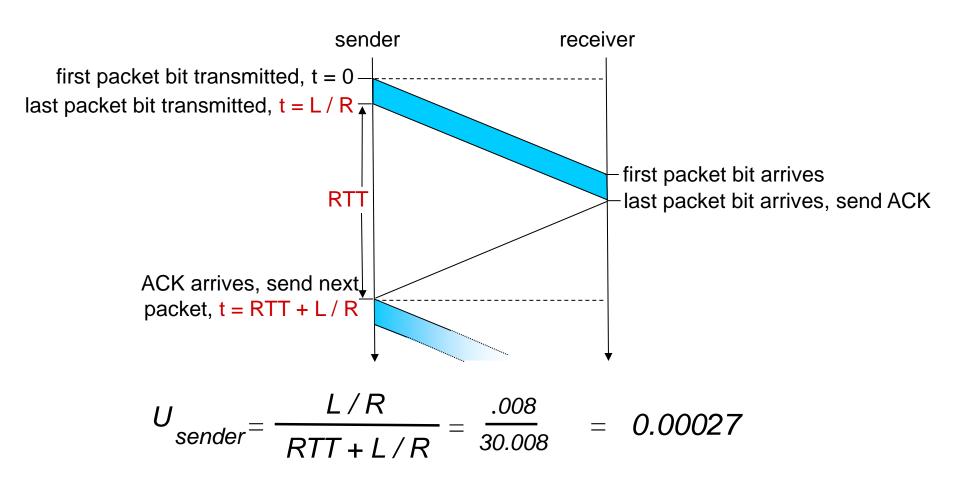
$$D_{trans} = \frac{L}{R} = \frac{8000 \text{ bits}}{10^9 \text{ bits/sec}} = 8 \text{ microsecs}$$

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!

rdt3.0: stop-and-wait operation

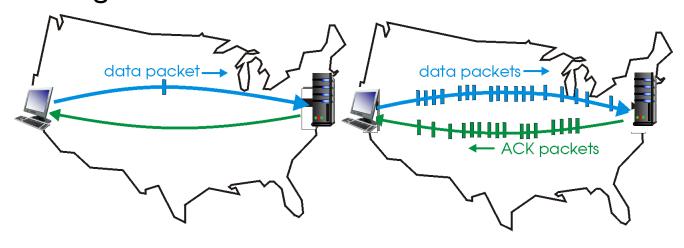


Pipelined protocols



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

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

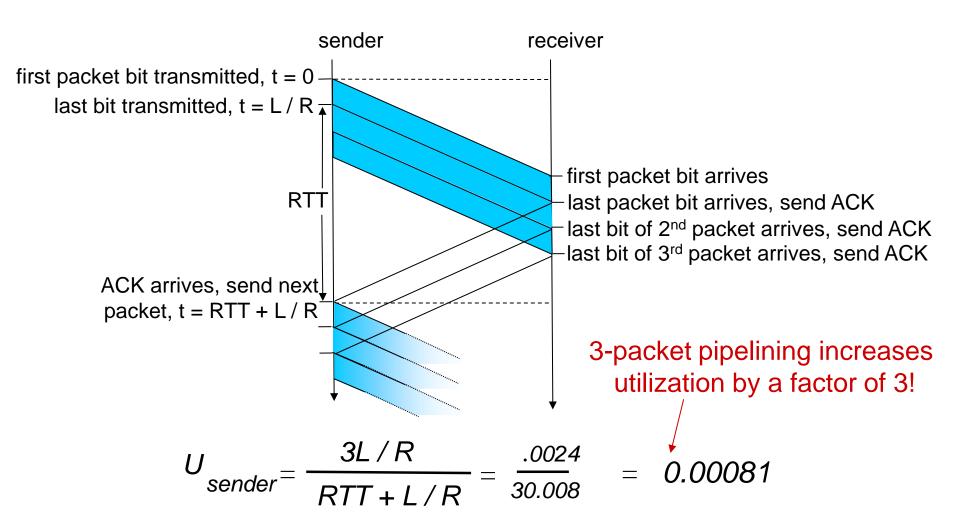


(a) a stop-and-wait protocol in operation

(b) a pipelined protocol in operation

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

Pipelining: increased utilization





Pipelined protocols: 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:

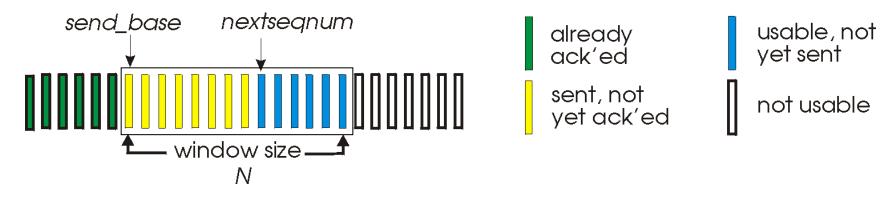
- sender can have up to N unack ed packets in pipeline
- rcvr sends individual ack for each packet

- 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



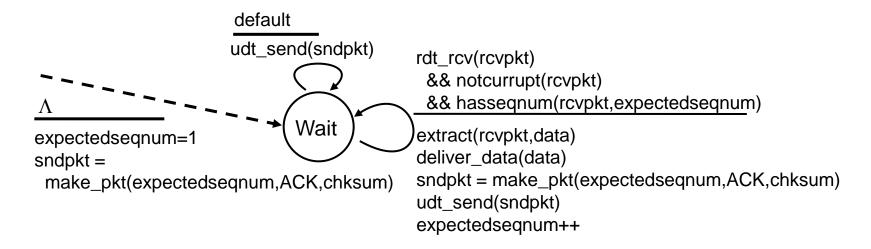
- ACK(n):ACKs all pkts up to, including seq # n "cumulative ACK"
 - may receive duplicate ACKs (see receiver)
- timer for oldest in-flight pkt
- timeout(n): retransmit packet n and all higher seq # pkts in window

GBN: sender extended FSM

```
rdt send(data)
                       if (nextseqnum < base+N) {
                          sndpkt[nextseqnum] = make_pkt(nextseqnum,data,chksum)
                          udt_send(sndpkt[nextseqnum])
                          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

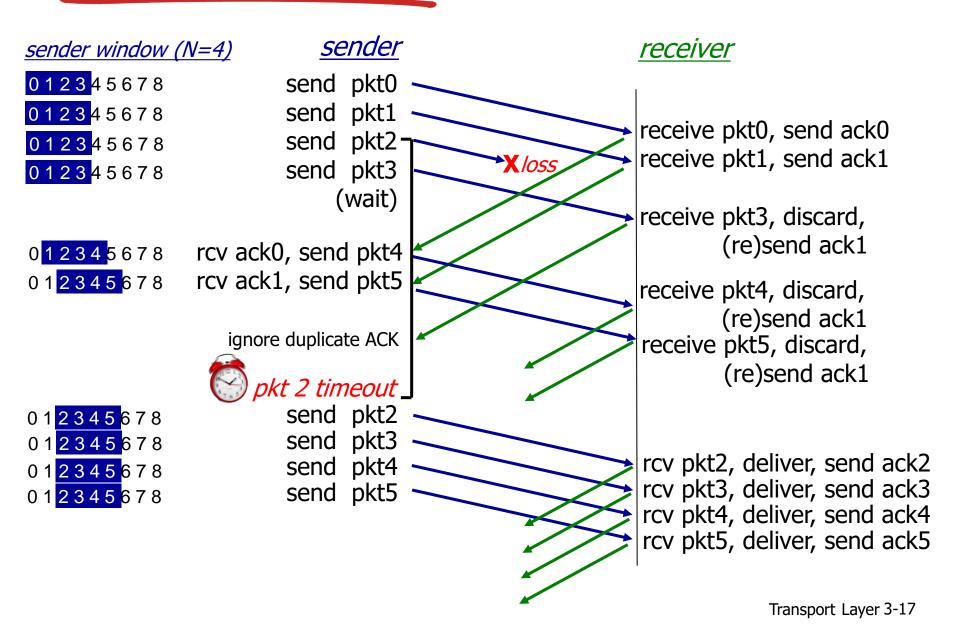




ACK-only: always send ACK for correctly-received pkt with highest *in-order* seq

- 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

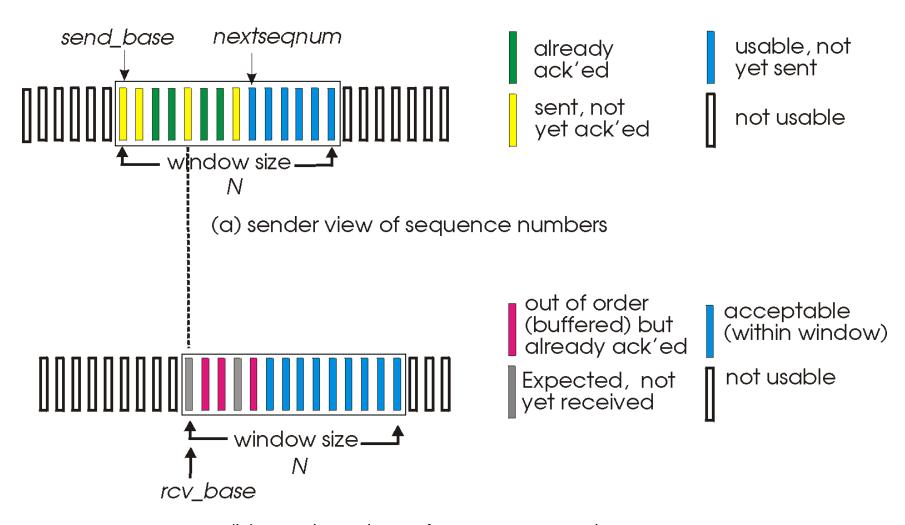




Selective repeat

- 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
 - limits seq #s of sent, unACKed pkts

Selective repeat: sender, receiver windows



(b) receiver view of sequence numbers





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

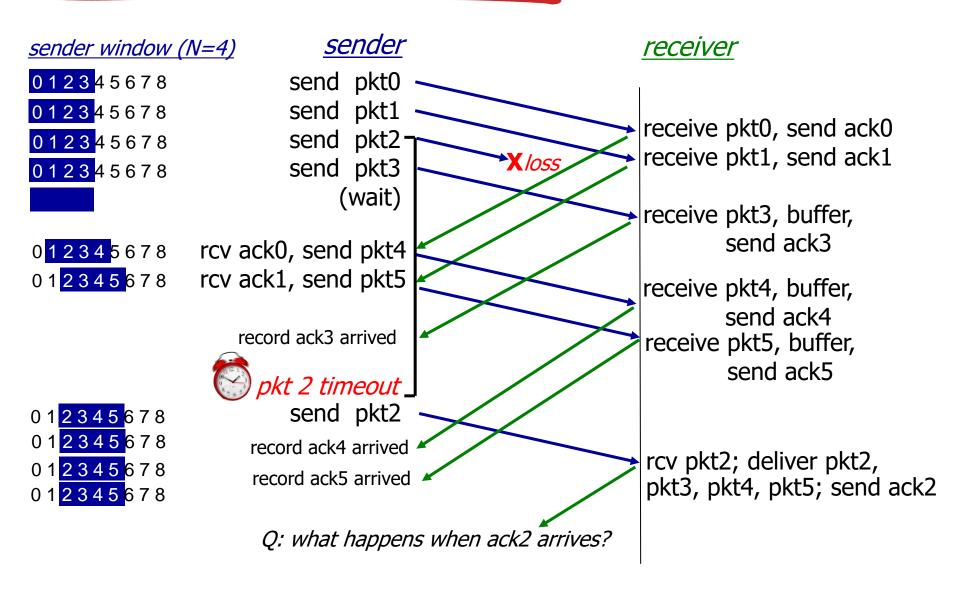
pkt n in [rcvbase-N,rcvbase-1]

ACK(n)

otherwise:

ignore

Selective repeat in action



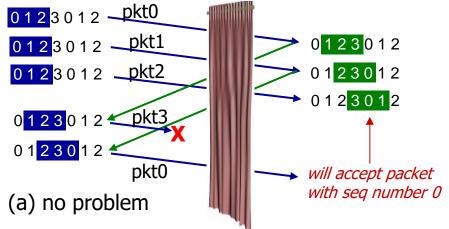
Selective repeat: dilemma

example:

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

sender window (after receipt)





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

