# **Transport Layer**



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Lecture 10

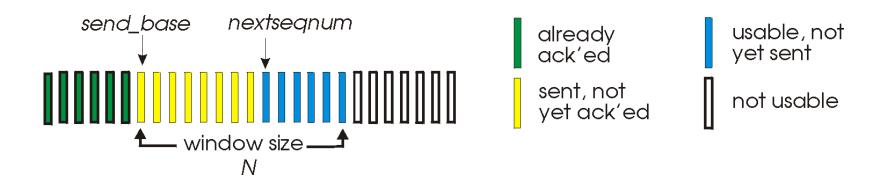
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## Go-Back-N

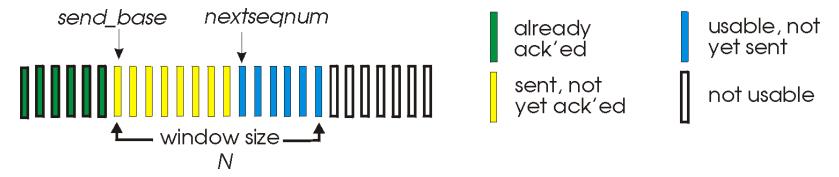
- Sender:
  - allow to transmit multiple packets without waiting for ACK
    - constrained to N unack'ed packets in the pipeline
- Sender's view of the sequence numbers in GBN





# Go-

## Go-Back-N



- send\_base: the seq. number of the oldest unack'ed packet
- nextseqnum: the smallest unused seq. number
- [0, send\_base I]: packets that have already been transmitted and acked
- [send\_base, nextseqnum I]: packets that have been sent but not yet acked
- [nextseqnum, send\_base + N I]: packets can be sent immediately

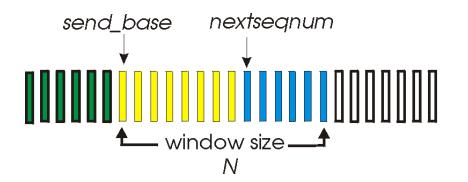




## Go-Back-N

#### Sender:

- allow to transmit multiple packets without waiting for ACK
  - constrained to N unack'ed packets in the pipeline
- "window" of up to N, consecutive unack'ed pkts allowed
- seq. number is carried in a fixed-length field in the packet header
- k-bit seq # in pkt header
  - [0, 2<sup>k</sup> 1]
  - e.g., TCP: 32-bits seq #



already ack'ed sent, not yet ack'ed usable, not yet sent

not usable

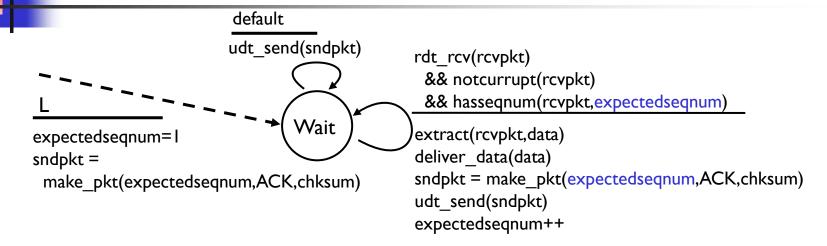


## **GBN: Sender Extended FSM**

rdt send(data)

```
if (nextsegnum < send base+N) {
                           sndpkt[nextseqnum] = make pkt(nextseqnum,data,chksum)
                           udt send(sndpkt[nextseqnum])
                           if (send base == nextseqnum)
                             start timer
                                                      first packet in the window N
                           nextseqnum++
                         else
                           refuse data(data)
   base=I
   nextseqnum=1
                                             timeout
                                             start_timer
                               Wait
                                             udt_send(sndpkt[send_base])
                                             udt_send(sndpkt[send_base+1])
rdt_rcv(rcvpkt)
 && corrupt(rcvpkt)
                                             udt send(sndpkt[nextseqnum-I])
                           rdt rcv(rcvpkt) &&
                             notcorrupt(rcvpkt)
                           send base = getacknum(rcvpkt)+1
                           If (send base == nextseqnum)
                             stop timer
                           else
                             start timer
```

## **GBN:** Receiver Extended FSM



- 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
- ACK-only: always send ACK for correctly-received pkt with highest in-order seq #
  - may generate duplicate ACKs
  - need only remember expectedseqnum



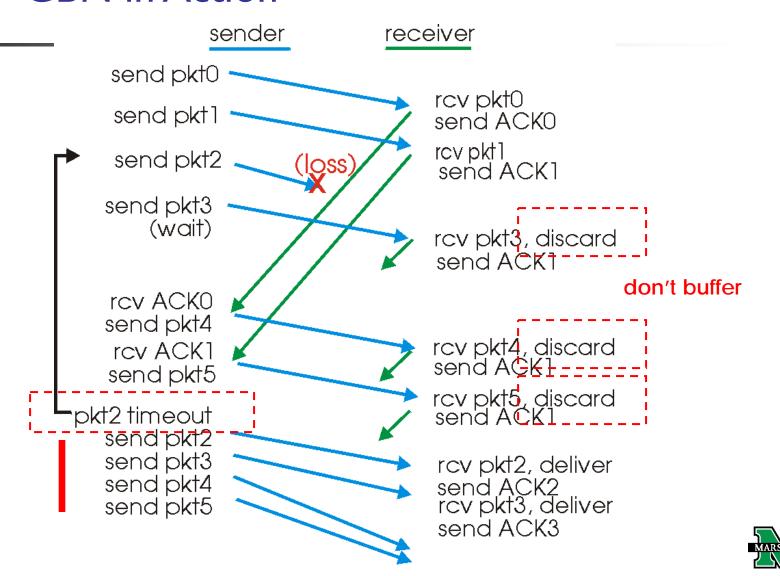


# GBN: Receiver Extended FSM (cont.)

- out-of-order pkt:
  - packet n is expected, but packet n+1 arrives
  - discard (don't buffer) -> no receiver buffering! why?
    - based on GBN, the sender transmit the packet
    - adv.?
      - simplicity of receiver buffering
    - disadv.?
      - throwing away a correctly received packet
      - subsequent retransmission of that packet might be lost or garbled
      - even more retransmission would be required



## **GBN** in Action





## Selective Repeat

- In GBN protocol,
  - allow the sender to potentially "fill the pipeline"
  - avoiding the channel utilization problem
  - → suffer from performance problem, e.g. many packets in the pipeline, but a single packet error?
    - retransmit a large number of packets (many unnecessarily)

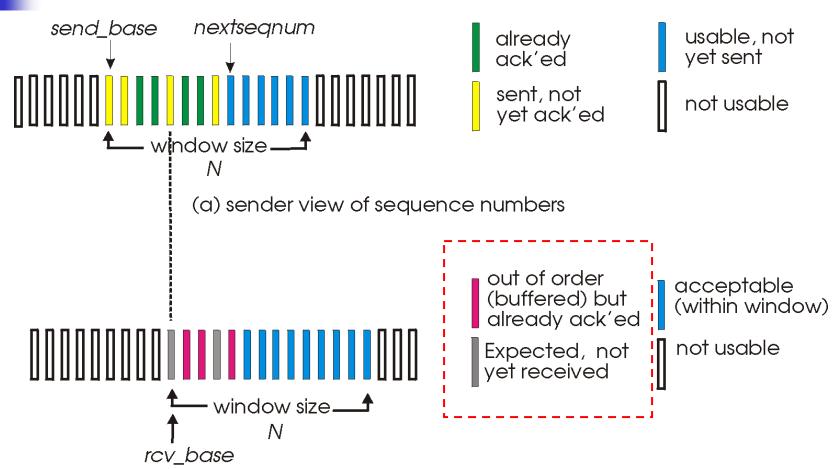


# Selective Repeat (cont.)

- selective-repeat protocol avoids unnecessary retransmissions by having the sender retransmit only those packets that it suspects were received in error at the receiver
- receiver individually acknowledges all correctly received pkts, whether or not they are in order
  - 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, unACKed pkts



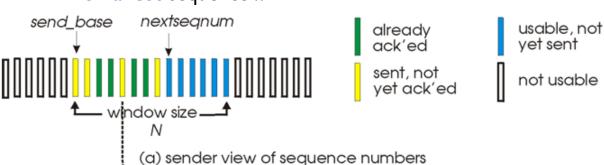
# Selective Repeat: Sender, Receiver Windows





# Selective Repeat (cont.)

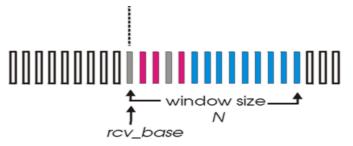
- Sender:
  - data received from above :
    - if next available seq # in window, send pkt
  - timeout(n):
    - resend pkt n, restart timer
  - ACK(n) in [send\_base, send\_base+N-1]:
    - mark pkt n as received
    - If the packet's sequence# == send\_base
      - advance the window base to the unacked packet with the smallest sequence #

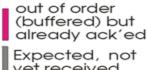






- Receiver:
  - pkt n in [rcv\_base, rcv\_base+N-1]
    - send ACK(n)
    - out-of-order: buffering
    - in-order: deliver (also deliver buffered, in-order pkts), advance window to next not-yet-received pkt
  - pkt n in [rcv\_base-N, rcv\_base-1]
    - ACK(n)
    - although, the receiver has previously acknowledged
  - otherwise: ignore the packet





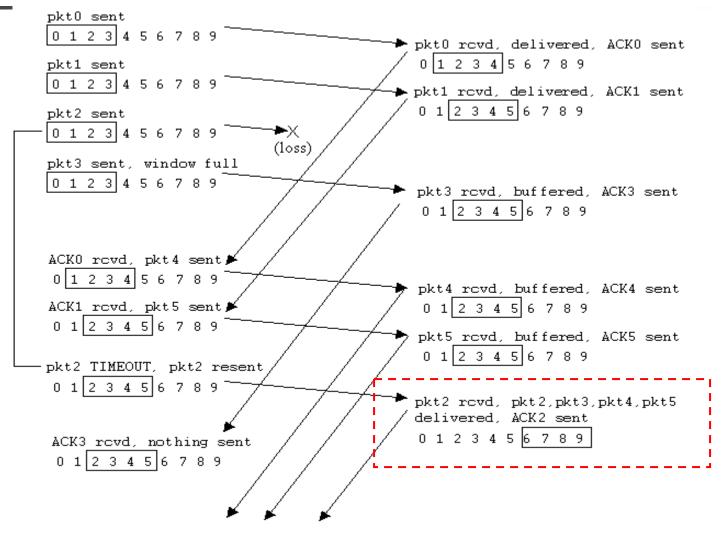
vet received

acceptable (within window)

not usable



## Selective repeat in Action







# Pipelined Protocols: Summary

#### Go-back-N:

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

### **Selective Repeat:**

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





RFCs: 793, 1122, 1323, 2018, 2581

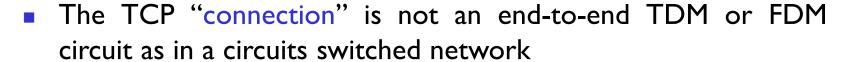


- before one application process can begin to send data to another
- the two processes must first "handshake" with each other
  - send some preliminary segments to each other to establish the parameters of the ensuing data transfer
- both sides of the connection will initialize many TCP state variables associated with the TCP connection



## TCP: Overview

RFCs: 793, 1122, 1323, 2018, 2581



- the connection state resides entirely in the two end systems
  - because the TCP protocol runs only in the end systems and not in the intermediate network elements (routers and linklayer switches)
  - the intermediate network elements do not maintain TCP connection state



## TCP: Overview

RFCs: 793, 1122, 1323, 2018, 2581



- if there is a TCP connection between Process A on one host and Process B on another host
- then application layer data can flow from Process A to Process B at the same time as application layer data flows from Process B to Process A
- A TCP connection is always point-to-point:
  - one sender and one receiver
- multicasting
  - the transfer of data from one sender to many receivers in a single send operation
  - not possible with TCP





- suppose a process running in one host wants to initiate a connection with another process in another host
  - client process
    - the process that is initiating the connection
  - server process
    - the other process
  - three-way handshake
  - 1. the client first sends a special TCP segment
  - 2. the server responds with a second special TCP segment
  - 3. the client responds again with a third special segment
    - the first two segments carry no payload
    - the third of these segments may carry a payload



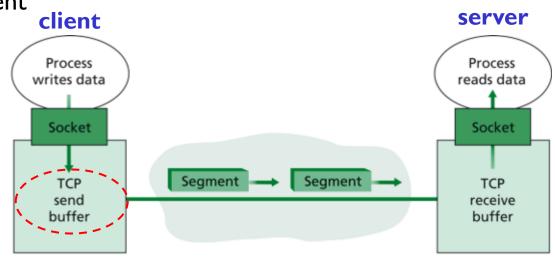


- Once a TCP connection is established, the two application processes can send data to each other
  - the client process passes a stream of data through the socket

 once the data passes through the door, the data is in the hands of TCP running in the client

#### send buffer:

- set aside during initial 3-way handshake
- grab chunks of data from send buffer and pass the data to the network layer
- maximum segment size (MSS): the maximum amount of data that can be grabbed and placed in a segment
  - MSS is typically set by first determining the length of the largest link-layer frame that can be sent by the sending host



TCP send and receive buffers



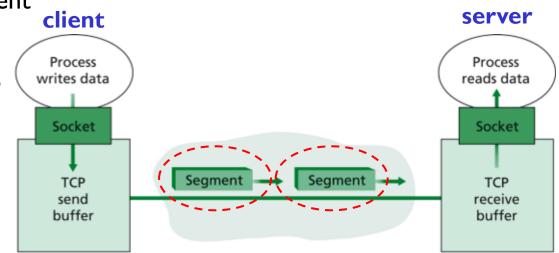


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#### segment:

 TCP pairs each chunk of client data with a TCP header, thereby forming TCP segments





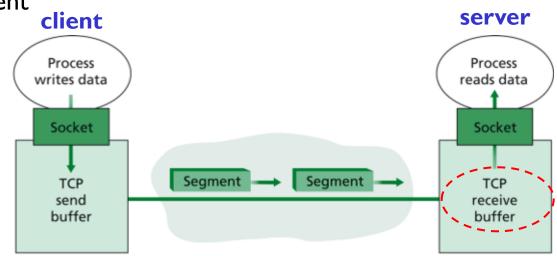


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#### receiver buffer:

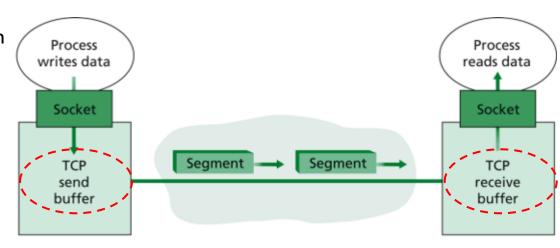
- when TCP receives a segment at the other end, the segment's data is placed in the TCP connection's receive buffer
- app. reads the stream of data from buffer







- Once a TCP connection is established, the two application processes can send data to each other
  - the client process passes a stream of data through the socket
  - once the data passes through the door, the data is in the hands of TCP running in the client
- each side of the connection has its own send buffer and its own receive buffer







source port # dest port # sequence number acknowledgement number head Receive window len checksum Urg data pnter Options (variable length) application data (variable length)

32 bits



# 4

# TCP Segment Structure

\_\_\_\_\_ 32 bits

32-bit sequence number field

32-bit acknowledgment number field

source port#	dest port#							
(sequence number)								
acknowledgement number								
head not len used UAPRSF	Receive window							
checksum	Urg data pnter							
Options (variable length)								
application data (variable length)								





32 bits

length field: the header length of the TCP header in 32-bit words

used when a sender the receiver negotiate size maximum segment (MSS)

SO	urce	ро	rt	#	dest port#				
sequence number									
acknowledgement number									
head len	not used	UA	P	R S	F	Receive window			
	chec	ksuı	m		Urg data pnter				
	Options (variable length)								

application data (variable length) flow control: # bytes rcvr willing to accept





URG: urgent data (generally not used)

Г									
	SO	urce	P	0	rt	<i>‡</i>	dest port#		
sequence number									number
$\frac{1}{2}$	acknowledgement number								
	head Ien	not used	U	Α	Р	R	S	F	Receive window
	checksum								Urg data pnter
	Options (variable length)								

32 bits

application data (variable length)





# TCP Segment Structure

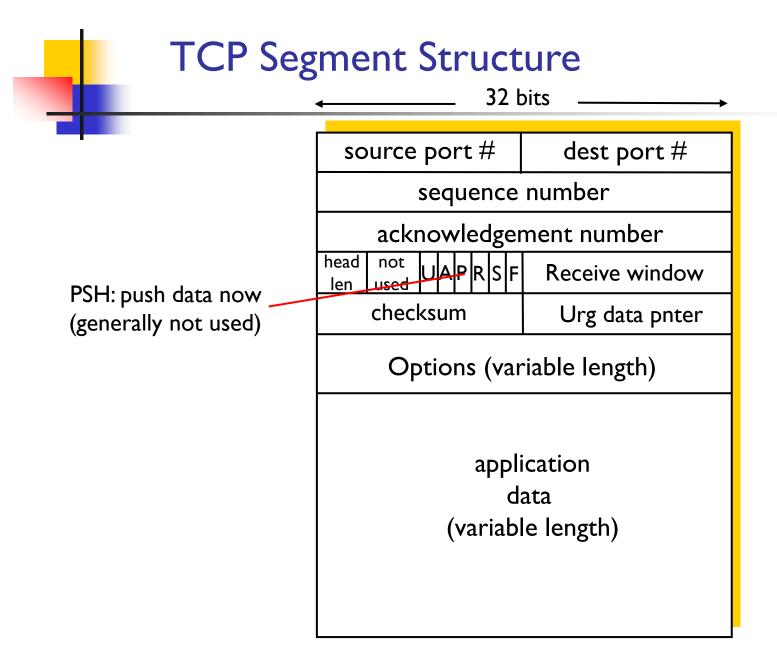
32 bits

ACK bit: indicate that the value carried in the acknowledgment field is valid;

source por	t #	dest port #							
sequence number									
acknowledgement number									
head not UAP	RSF	Receive window							
checksum	l	Urg data pnter							
Options (variable length)									
application									

application
data
(variable length)







# TCP Segment Structure

32 bits ———

source port # dest port #

sequence number

acknowledgement number

head not used UAPRSF Receive window
checksum Urg data pnter

Options (variable length)

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

application data (variable length)



# TCP Segment Structure

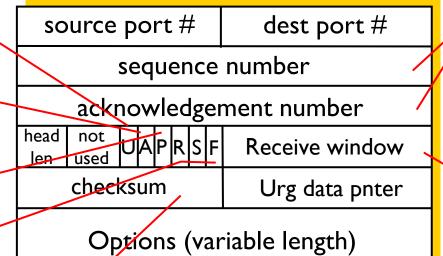
URG: urgent data (generally not used)

ACK:ACK # valid

PSH: push data now (generally not used)

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

> Internet checksum' (as in UDP)



32 bits

application data (variable length) by bytes
of data
(not segments!)

# bytes
rcvr willing
to accept

