Transport Layer

Instructor: C. Pu (Ph.D., Assistant Professor)

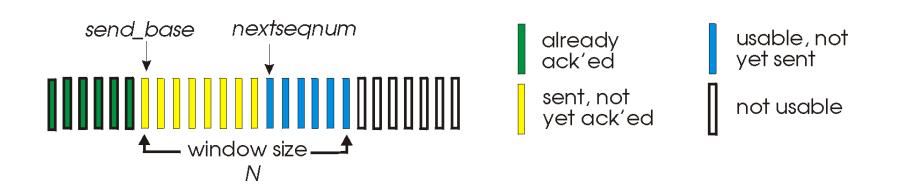
Lecture 10

puc@marshall.edu

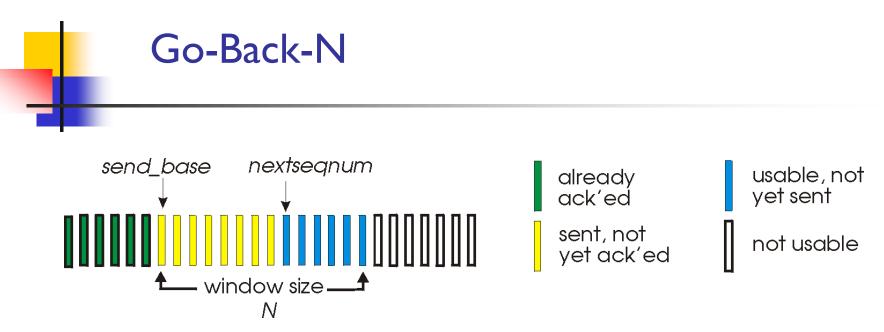




- Sender:
 - allow to transmit multiple packets without waiting for ACK
 - constrained to N unack'ed packets in the pipeline
- Sender's view of the sequence numbers in GBN





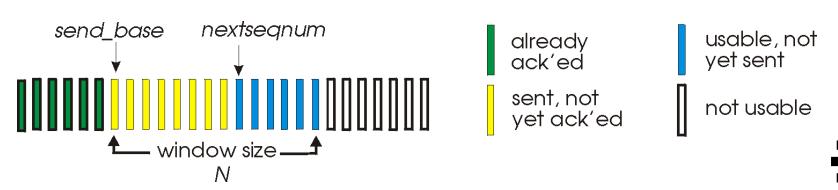


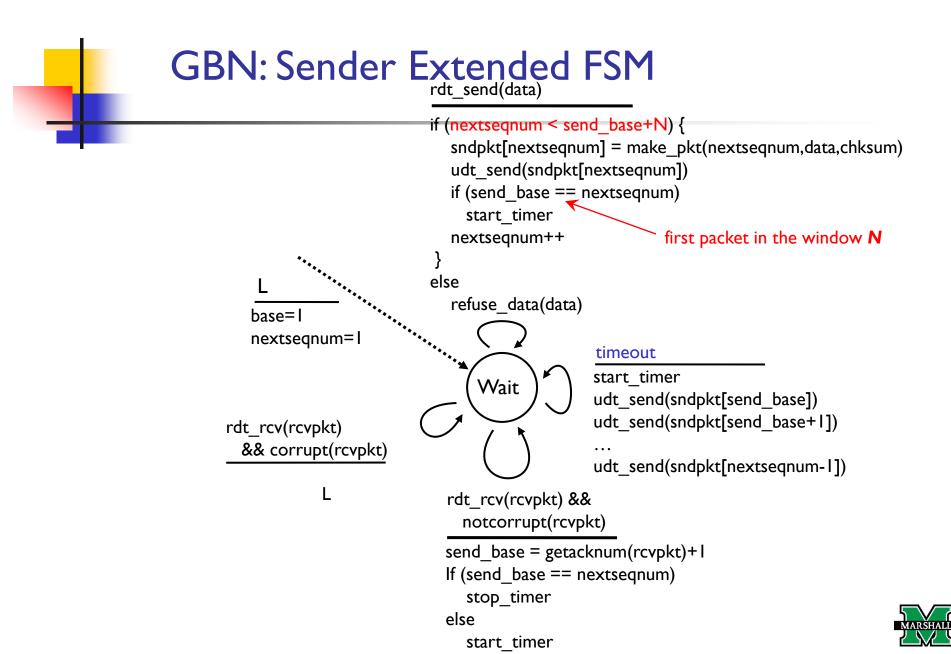
- send_base: the seq. number of the oldest unack'ed packet
- nextseqnum: the smallest unused seq. number
- [0, send_base I]: packets that have already been transmitted and acked
- [send_base, nextseqnum 1]: packets that have been sent but not yet acked
- [nextseqnum, send_base + N I]: packets can be sent immediately



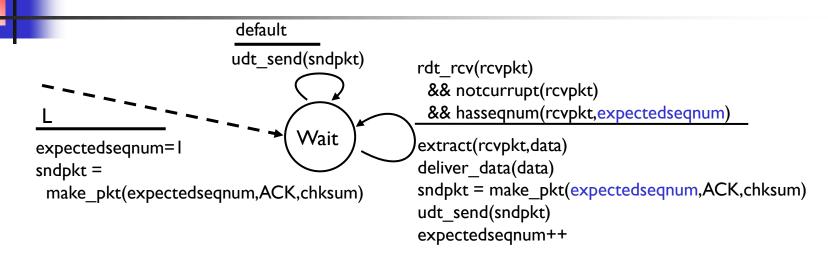


- Sender:
 - allow to transmit multiple packets without waiting for ACK
 - constrained to N unack'ed packets in the pipeline
 - "window" of up to N, consecutive unack'ed pkts allowed
 - seq. number is carried in a fixed-length field in the packet header
 - k-bit seq # in pkt header
 - [0, 2^k 1]
 - e.g.,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





GBN: Receiver Extended FSM



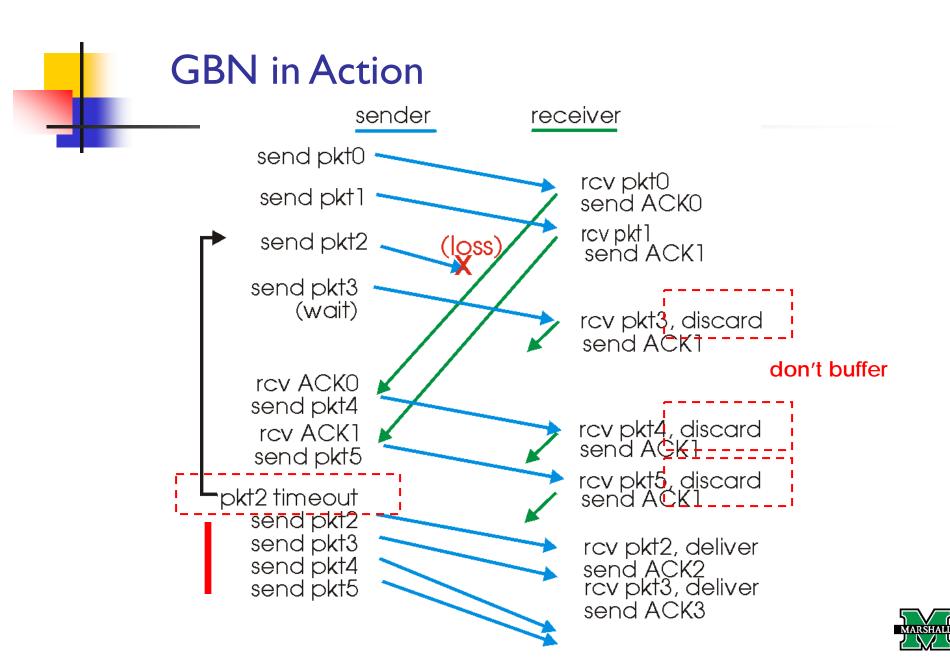
- ACK(n): ACKs all pkts up to, including seq # n "cumulative ACK"
 - may receive duplicate ACKs (see receiver)
- timer for each in-flight pkt
- timeout(n): retransmit pkt n and all higher seq # pkts in window
- ACK-only: always send ACK for correctly-received pkt with highest in-order seq #
 - may generate 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







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)

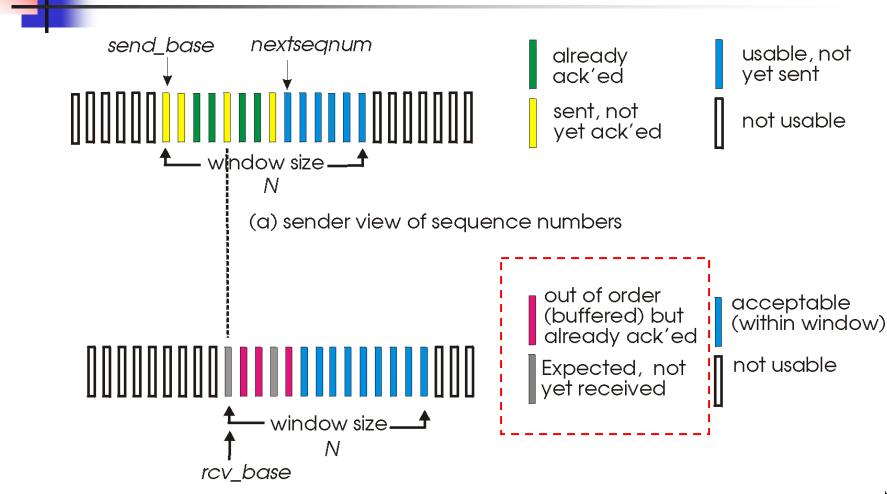


Selective Repeat (cont.)

- 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

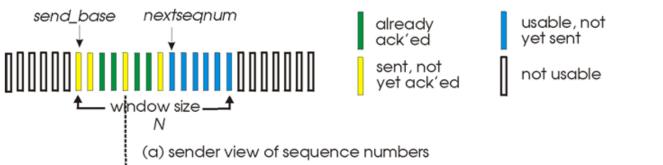


(b) receiver view of sequence numbers



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-I]:
 - mark pkt n as received
 - If the packet's sequence# == send_base
 - advance the window base to the unacked packet with the smallest sequence #



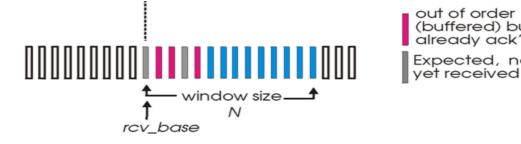


Selective Repeat (cont.)

- **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

(b) receiver view of sequence numbers

otherwise: ignore the packet



out of order (buffered) but already ack'ed Expected, not

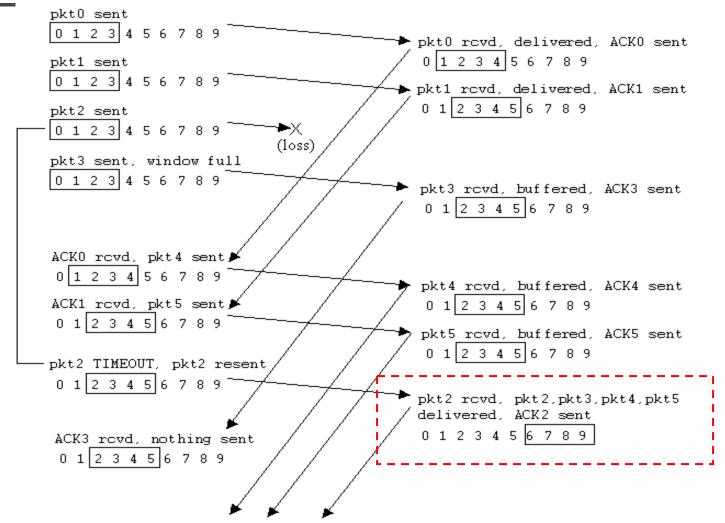
acceptable (within window)

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



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

- TCP is said to be connection-oriented
 - 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 TCP "connection" is not an end-to-end TDM or FDM circuit as in a circuits switched network
 - 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

- ATCP connection provides a *full-duplex service*
 - 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
- ATCP 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



TCP: How a TCP Connection is Established

- 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
 - I. 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



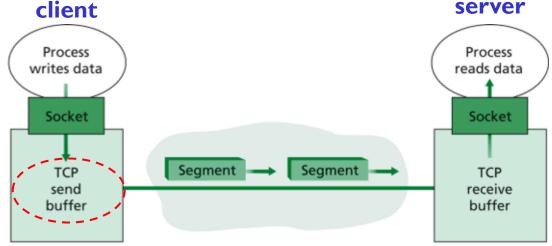
TCP: How a TCP Connection is Established

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



client

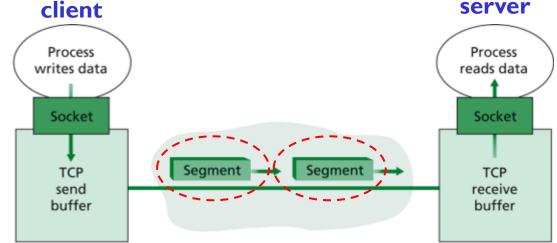
TCP: How a TCP Connection is Established

 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

segment:

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





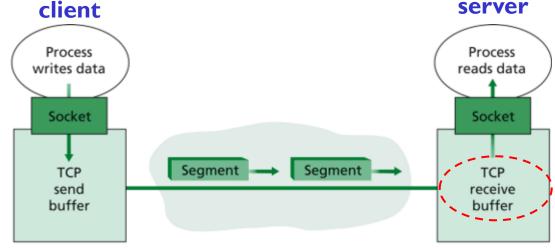
TCP: How a TCP Connection is Established

 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

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





TCP: How a TCP Connection is Established

- 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

