

Reliable data transfer with TCP

- efficiency
 - stop-and-wait
 - pipelining
 - sliding-window

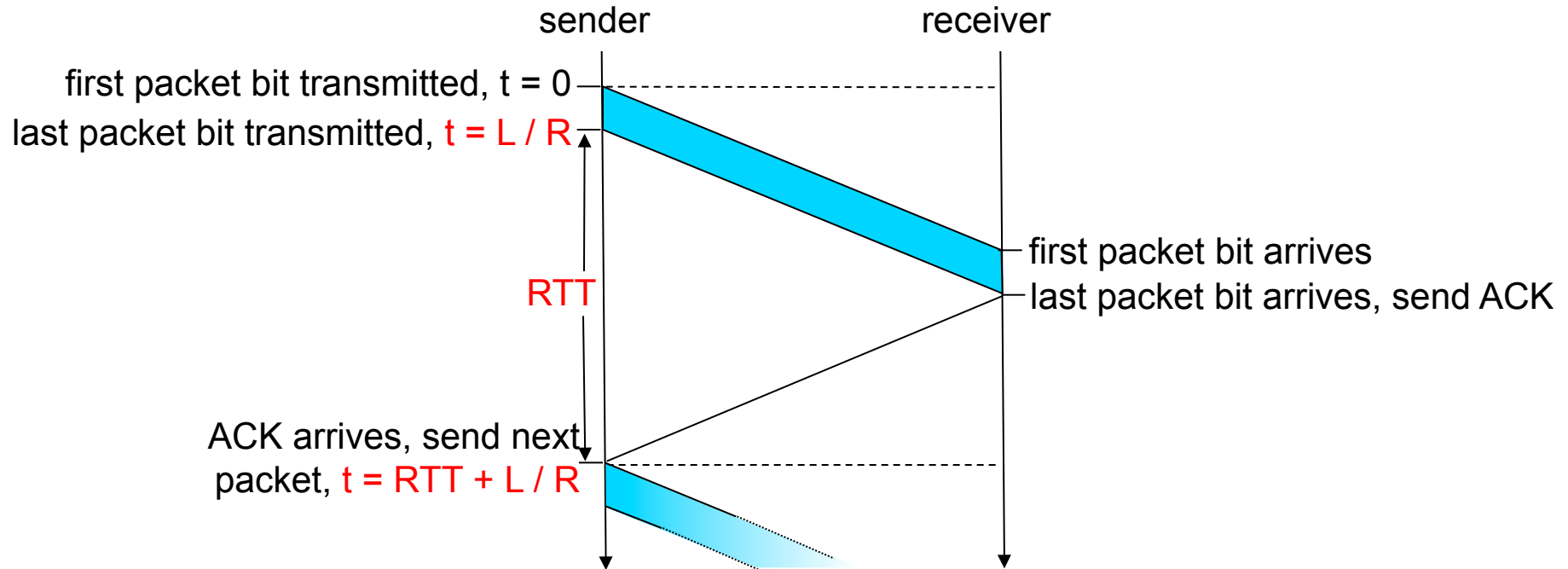
Note: Many of the lecture slides are based on presentations that accompany *Computer Networking: A Top Down Approach*, 6th edition, by Jim Kurose & Keith Ross, Addison-Wesley, 2013.

Efficiency considerations

- TCP's acknowledgement model is great for reliability ...
- ... but with what tradeoffs?
 - Segment overhead
 - Each packet requires at least one RTT
 - send segment, wait for ACK
 - May increase queuing delay
 - Increases network congestion

Performance of “stop-and-wait”

- Example: $R=1$ Gbps, 15 ms end-to-end propagation delay, $L=1000$ Byte packet:

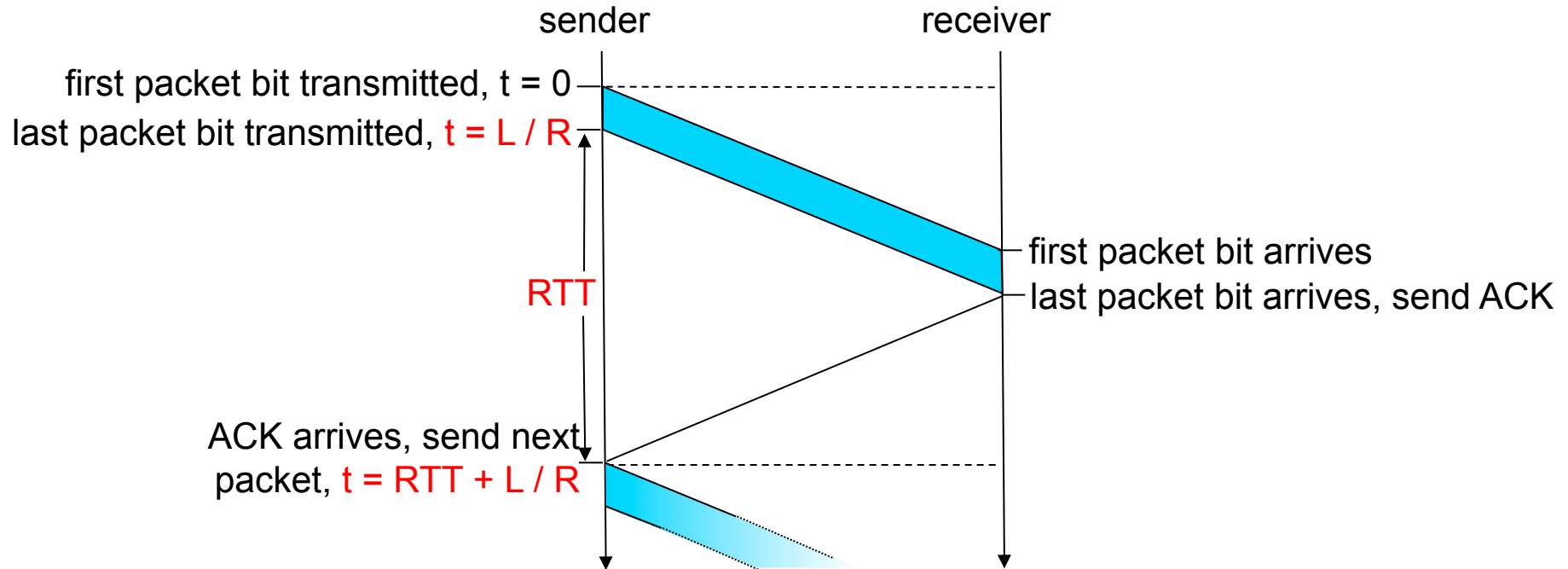


$$\text{transmission delay} = \frac{L}{R} = \frac{8 \times 10^3 \text{ bits}}{10^9 \text{ bps}} = 0.008 \text{ ms}$$

$$\text{round-trip time (RTT)} = 30 \text{ ms}$$

Performance of “stop-and-wait”

- Example: $R=1$ Gbps, 15 ms end-to-end propagation delay, $L=1000$ Byte packet:

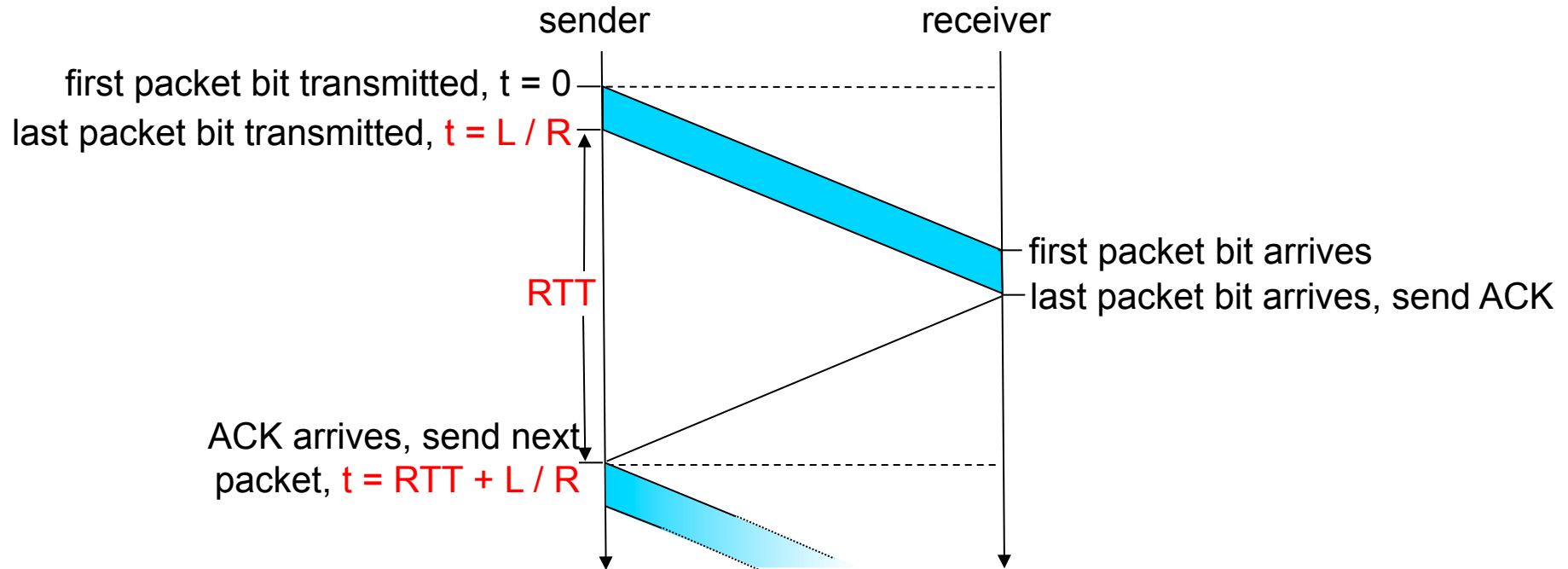


U_{sender} : utilization = fraction of time sender is busy sending

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

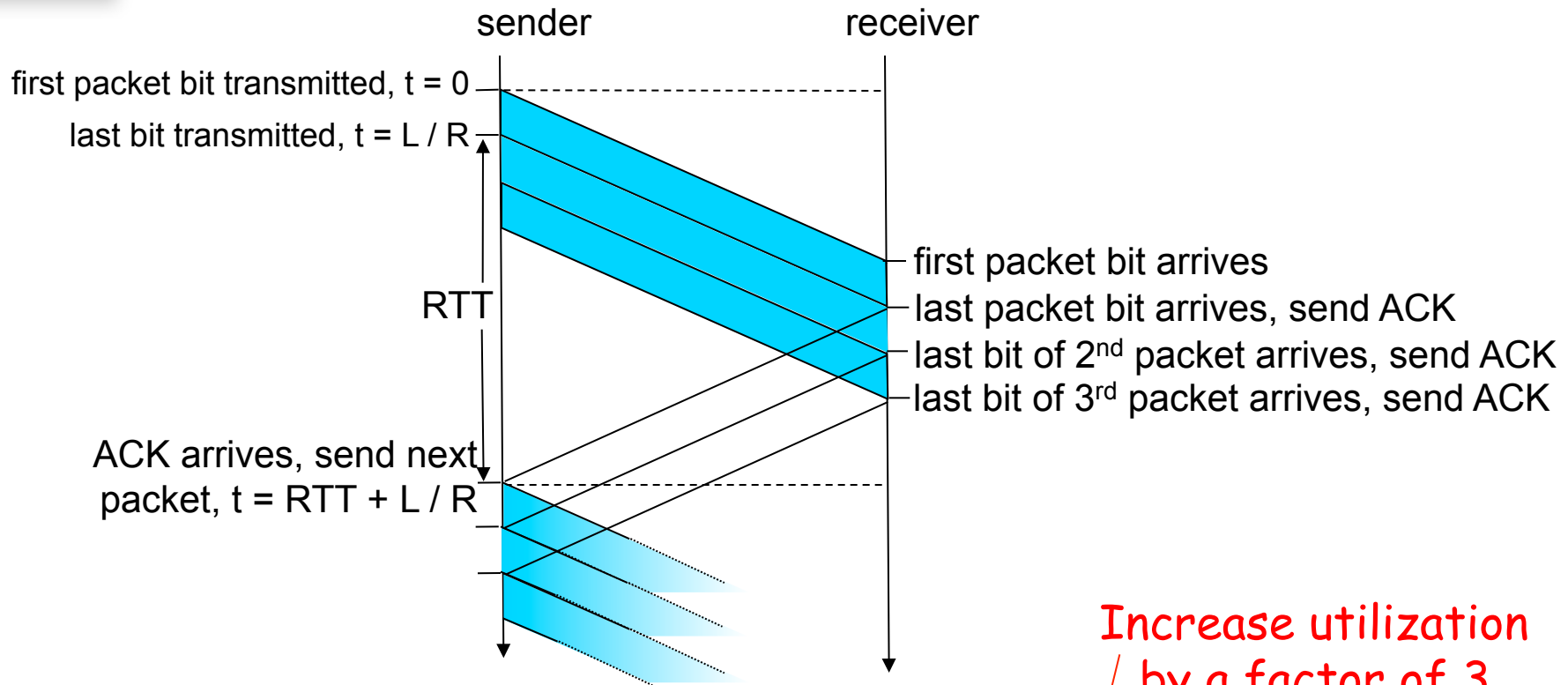
Performance of “stop-and-wait”

- Example: $R=1$ Gbps, 15 ms end-to-end propagation delay, $L=1000$ Byte packet:



- 1KB packet every 30 msec -> **33Kbps** throughput over **1 Gbps** link !!!
- “stop-and-wait” protocol limits use of physical resources.

Pipelining: increased utilization

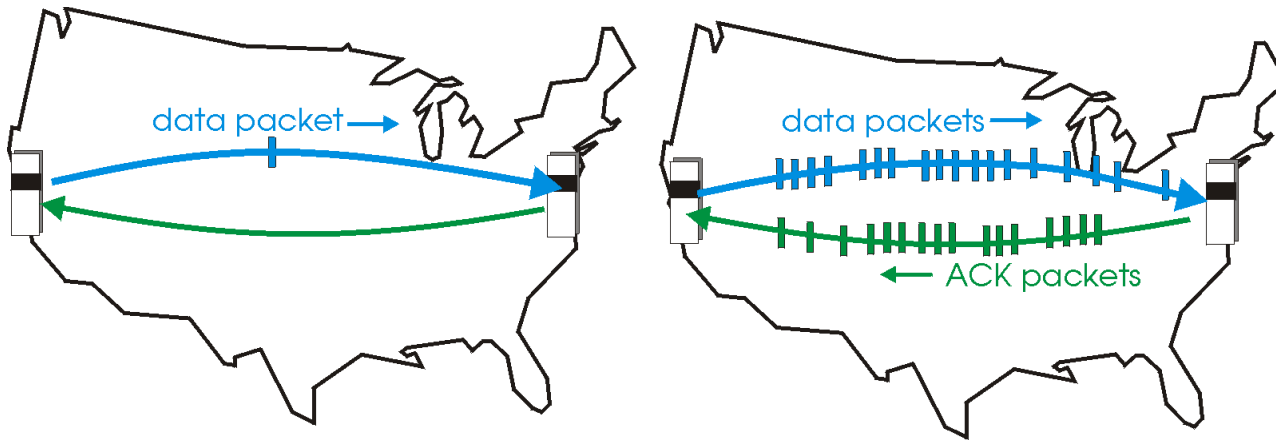


$$U_{\text{sender}} = \frac{3 * L / R}{RTT + L / R} = \frac{.024}{30.008} = 0.0008$$

Increase utilization
by a factor of 3

Pipelined protocols

Pipelining: Sender transmits multiple packets.
Packets “in-flight” have yet to be acknowledged



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

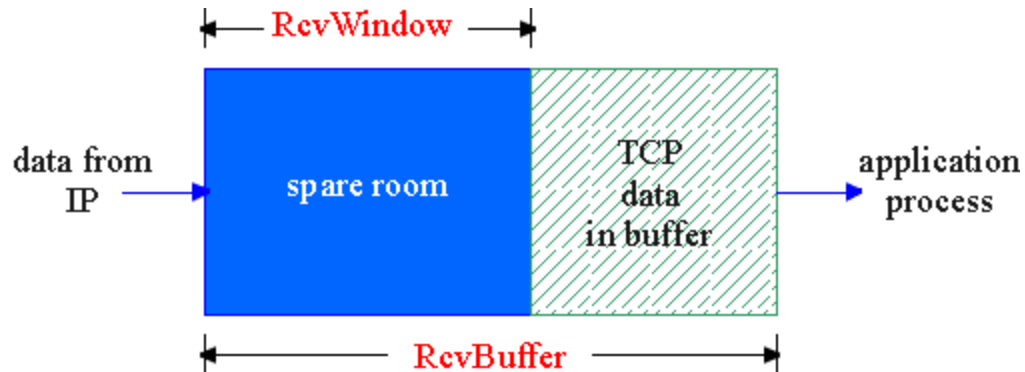
(b) a pipelined protocol in operation

Problem: receiver might not be able to handle that many packets as fast as they arrive.

Problem: packets might arrive out-of-order

TCP flow control

- receive side of TCP connection has a receive buffer:



- receiver application layer process reading from buffer may be slow

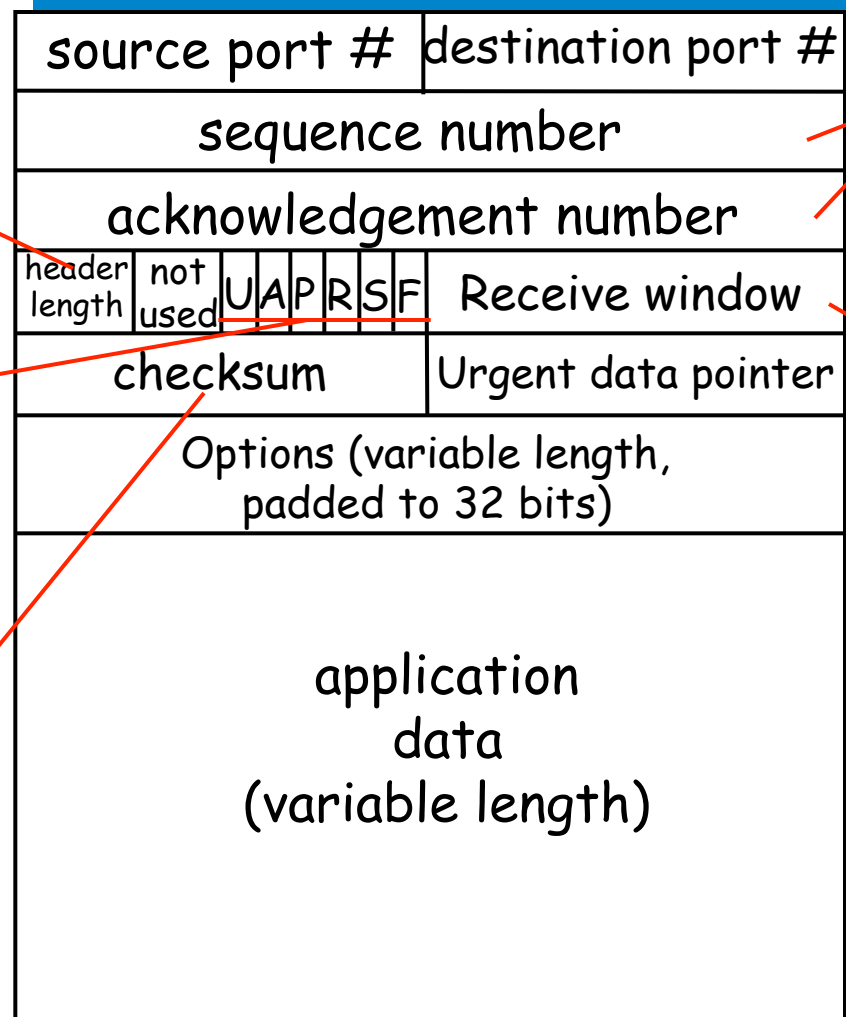
flow control

sender won't overflow receiver's buffer by transmitting too much, too fast

- speed-matching service: matching the send rate to the receiving application's drain rate
- See animations on textbook's website.

TCP segment structure

← 32 bits →



4-bit header size.
Number of 32-bit
“lines” (minimum=5,
maximum=15)

Flags for urgent data,
ACK validity, push,
reset, synchronize,
final data

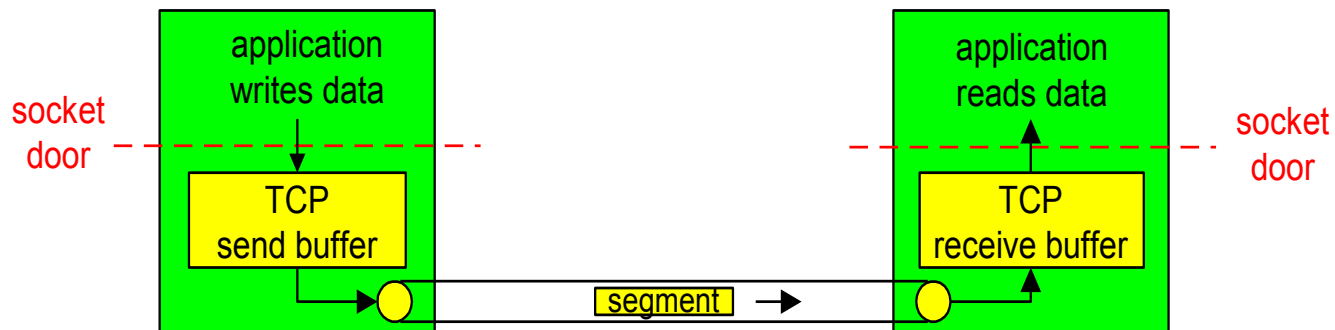
Internet
checksum
(as in UDP)

counting
by bytes
of data
(not segments!)

bytes
receiver
is willing
to accept

TCP flow control

- TCP uses **sliding window** for flow control
 1. Sender transmits one segment
 2. Receiver specifies window size (**window advertisement**) in ACK header “receive window”
 - Specifies how many bytes in the data stream can be sent
 3. Sender limits unACKed data to “receive window”
 - guarantees receive buffer doesn't overflow



- Efficiency
 - stop-and-wait
 - pipelining
- flow control
- “receive window”
- sliding-window protocol
- Next hurdle:
 - What happens if there are errors in the pipeline?