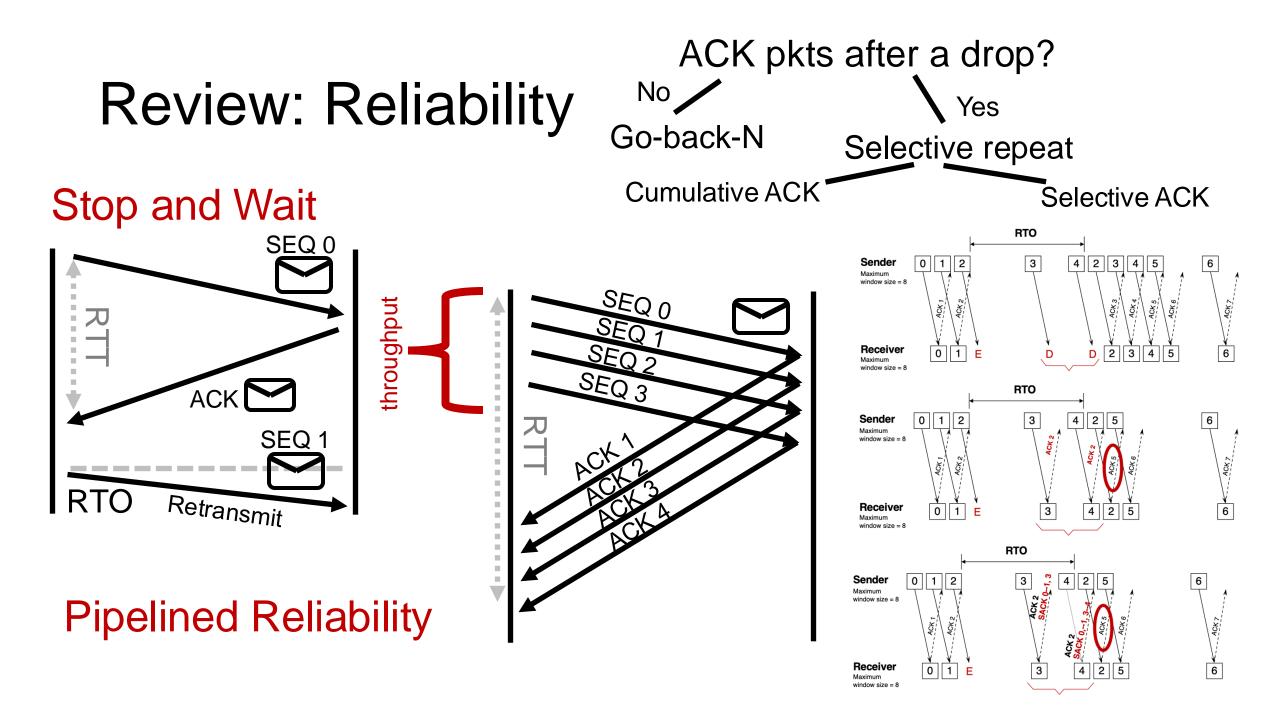
Ordering & Flow Control

Lecture 15

http://www.cs.rutgers.edu/~sn624/352-F24

Srinivas Narayana

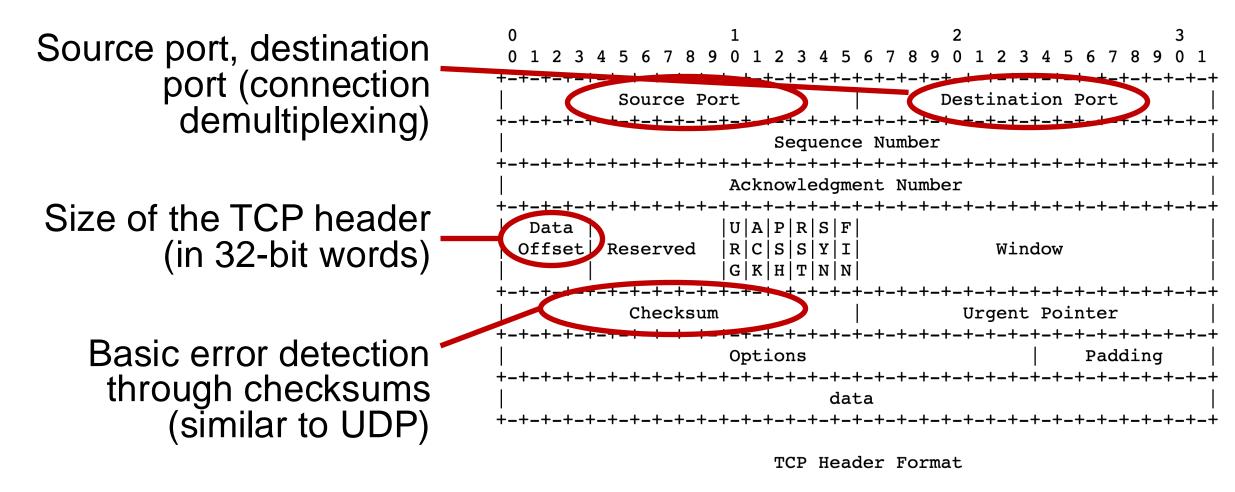




TCP reliability metadata

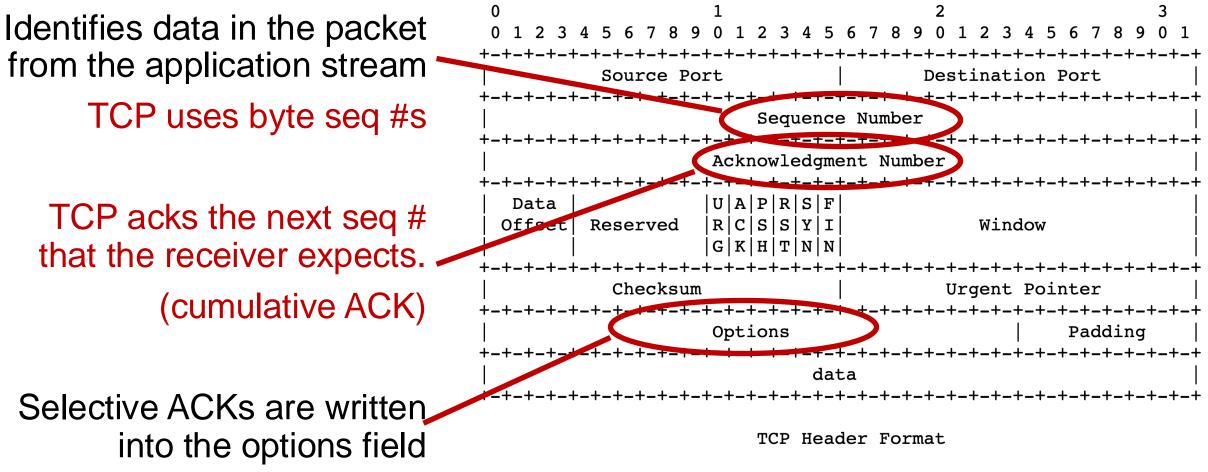
Where are reliability metadata (seq and ack numbers) stored?

TCP header structure



Note that one tick mark represents one bit position.

TCP header structure



Note that one tick mark represents one bit position.

Observing a TCP exchange

- sudo tcpdump -i enol tcp portrange 56000-56010
- curl --local-port 56000-56010
 https://www.google.com > output.html
- Bonus: Try crafting TCP packets with scapy!

TCP Stream-Oriented Data Transfer

Sequence numbers in the app's stream

Data written by application over time e.g., send() call

|--|

Increasing sequence #s

TCP uses byte sequence numbers

Sequence numbers in the app's stream

Data written by application over time e.g., send() call

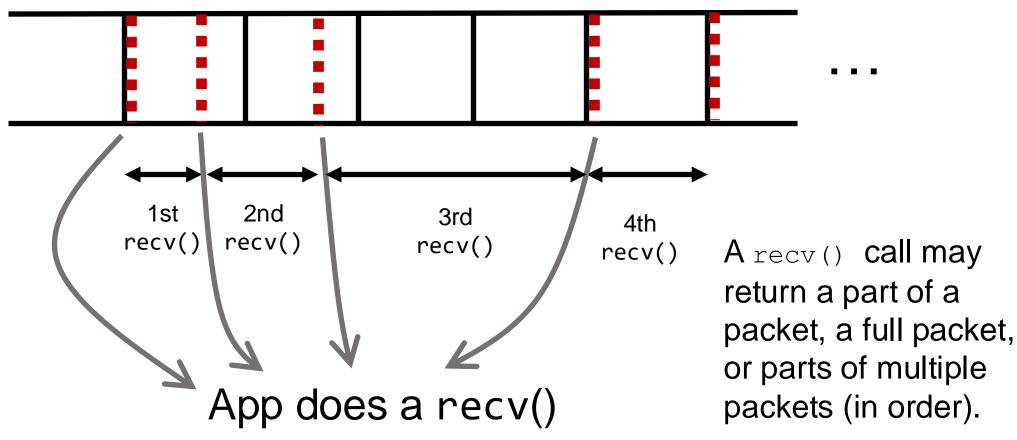
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Increasing sequence #s

Packet boundaries aren't important for TCP software TCP is a stream-oriented protocol (We use SOCK_STREAM when creating sockets)

Sequence numbers in the app's stream

Data written by application over time e.g., send() call



Buffering and Ordering in TCP



Memory Buffers at the Transport Layer

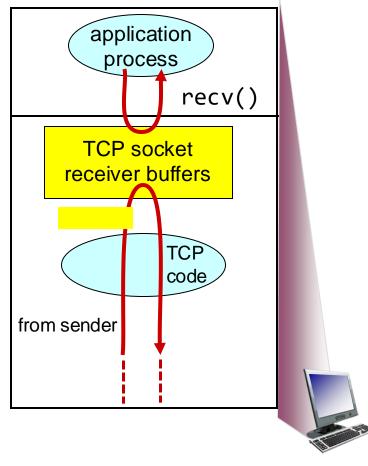
Sockets need receive-side memory buffers

- Since TCP uses selective repeat, the receiver must buffer data that is received after loss:
 - e.g., hold packets so that only the "holes" (due to loss) need to be filled in later, without having to retransmit packets that were received successfully
- Apps read from the receive-side socket buffer when you do a recv() call.
- Even if data is always reliably received, applications may not always read the data immediately
 - What if you invoked recv() in your program infrequently (or never)?
 - For the same reason, UDP sockets also have receive-side buffers

Receiver app's interaction with TCP

- Upon reception of data, the receiver's TCP stack deposits the data in the receive-side socket buffer
- An app with a TCP socket reads from the TCP receive socket buffer

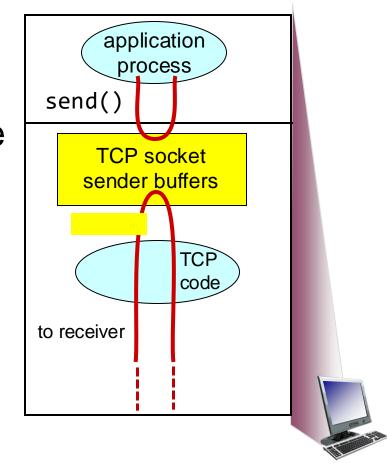
• e.g., when you do data = sock.recv()



receiver TCP interaction

Sockets need send-side memory buffers

- The possibility of packet retransmission in the future means that data can't be immediately discarded from the sender once transmitted.
- App has issued send() and moved on; TCP stack must buffer this data
- Transport layer must wait for ACK of a piece of data before reclaiming (freeing) the memory for that data.

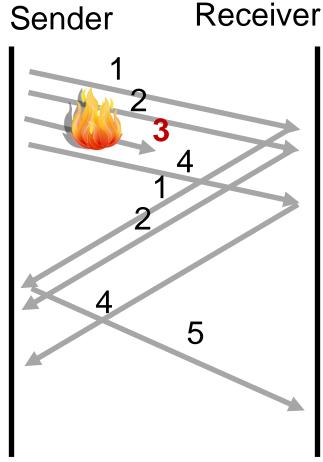


sender TCP interaction

Ordered Delivery

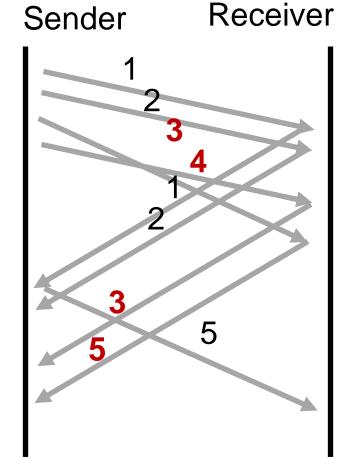
Reordering packets at the receiver side

- Let's suppose receiver gets packets 1, 2, and 4, but not 3 (dropped)
- Suppose you're trying to download a document containing a report
- What would happen if transport at the receiver directly presents packets 1, 2, and 4 to the application (i.e., receiving 1,2,4 through the recv() call)?



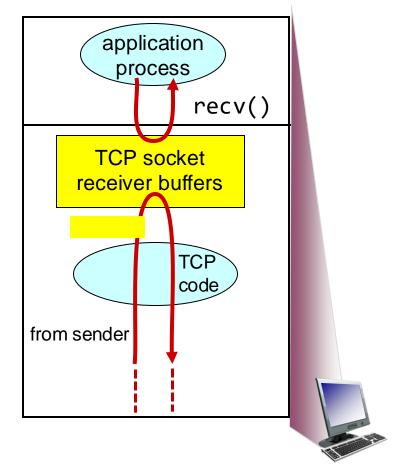
Reordering packets at the receiver side

- Reordering can happen for a few reasons:
 - Drops
 - Packets taking different paths through a network
- Receiver needs a general strategy to ensure that data is presented to the application in the same order that the sender pushed it. Ideas?
- To implement ordered delivery, the receiver uses
 - Sequence numbers
 - Receiver socket buffer
- We've already seen the use of these for reliability; but they can be used to order too!

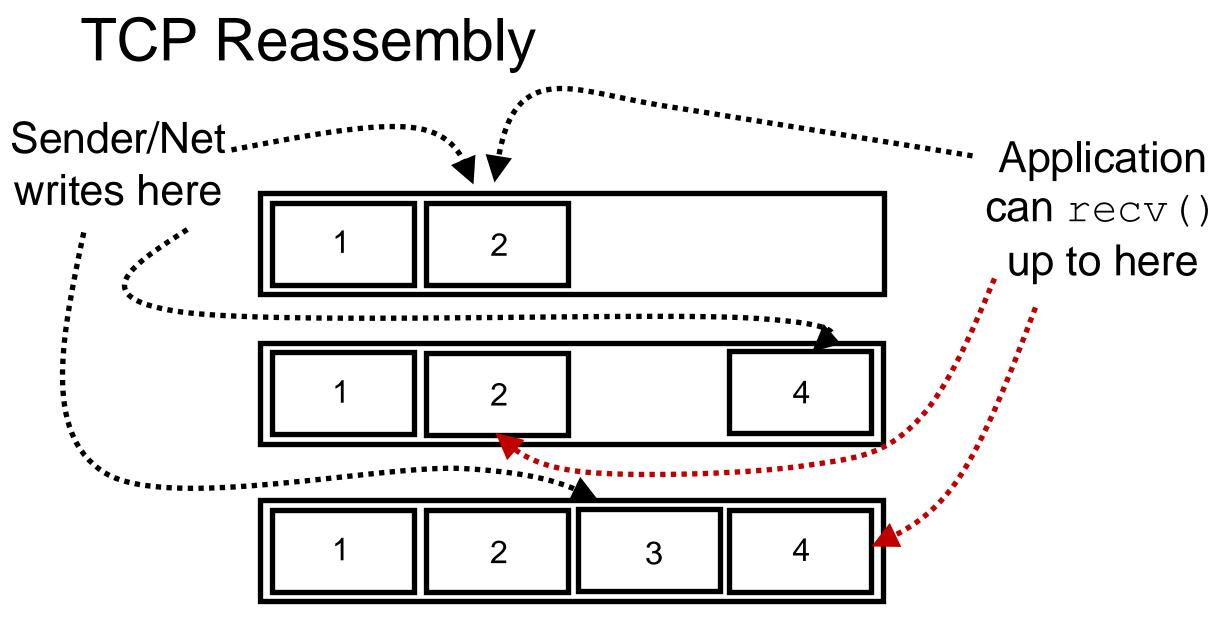


Receive-side app and TCP

- TCP receiver software only releases the data from the receive-side socket buffer to the application if:
 - the data is in order relative to all other data already read by the application
- This process is called TCP reassembly



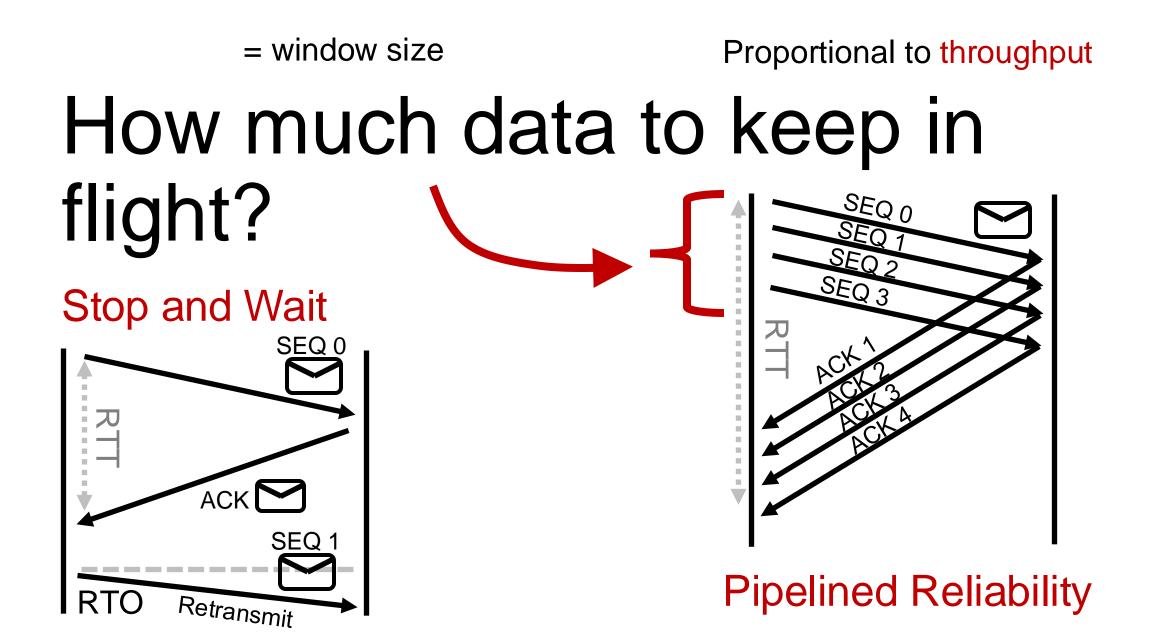
receiver protocol stack



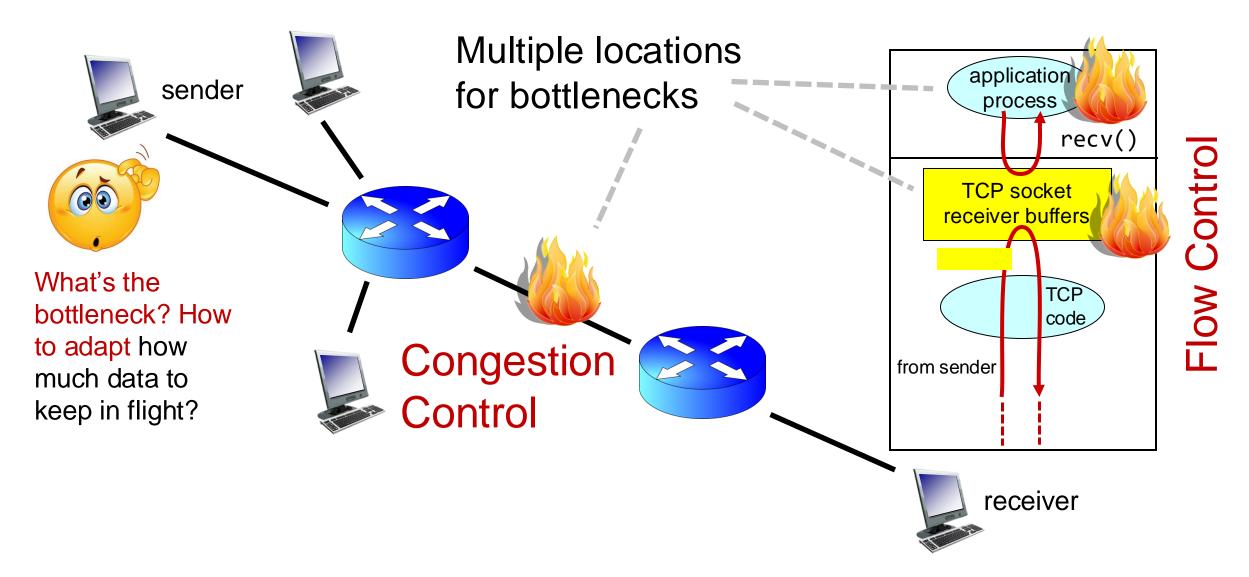
Socket buffer memory on the receiver

Implications of ordered delivery

- Packets cannot be delivered to the application if there is an inorder packet missing from the receiver's buffer
 - The receiver can only buffer so much out-of-order data
 - Subsequent out-of-order packets dropped
 - It won't matter that those packets successfully arrive at the receiver from the sender over the network
- TCP application-level throughput will suffer if there is too much packet reordering in the network
 - Data may have reached the receiver, but won't be delivered to apps upon a recv() (...or may not even be buffered!)



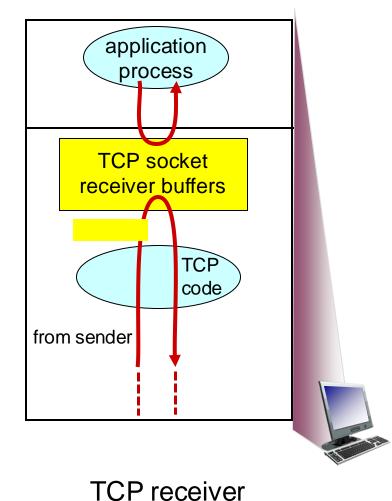
We want to increase throughput, but ...



Flow Control

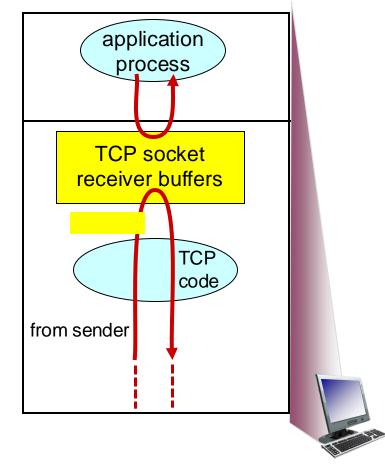
Socket buffers can become full

- Applications may read data slower than the sender is pushing data in
 - Example: what if an app infrequently or never calls recv()?
- There may be too much reordering or packet loss in the network
 - What if the first few bytes of a window are lost or delayed?
- Receivers can only buffer so much before dropping subsequent data



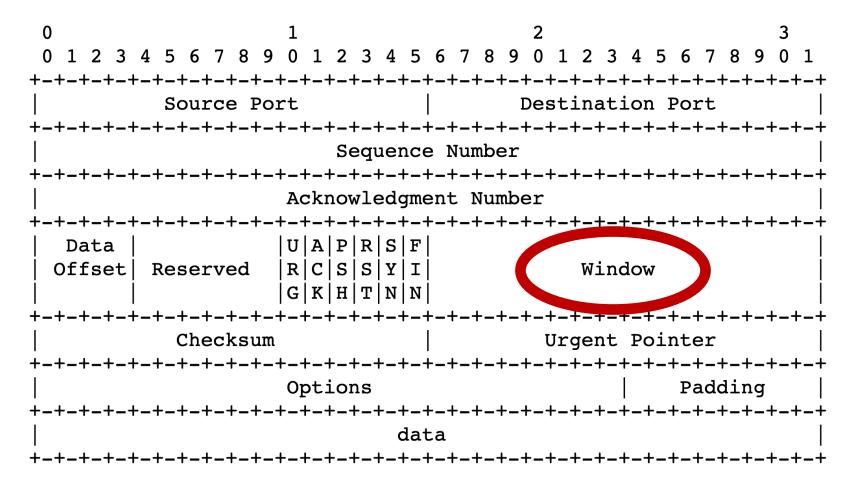
Goal: avoid drops due to buffer fill

- Have a TCP sender only send as much as the free buffer space available at the receiver.
- Amount of free buffer varies over time!
- TCP implements flow control
- Receiver's ACK contains the amount of data the sender can transmit without running out the receiver's socket buffer
- This number is called the advertised window size



receiver protocol stack

Flow control in TCP headers

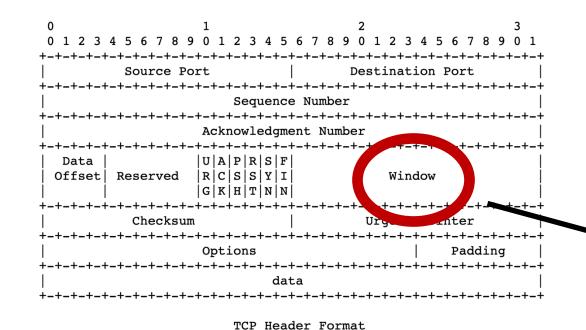


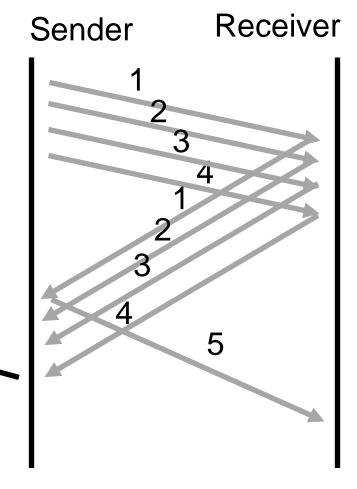
TCP Header Format

Note that one tick mark represents one bit position.

TCP flow control

 Receiver advertises to sender (in the ACK) how much free buffer is available





Note that one tick mark represents one bit position.

TCP flow control

- Subsequently, the sender's sliding window cannot be larger than this value
- Restriction on new sequence numbers that can be transmitted
- == restriction on sending rate

