



15-441 Computer Networking

Lecture 16 – TCP & Congestion Control
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www.cs.cmu.edu/~prs/15-441-F10

Good Ideas So Far...



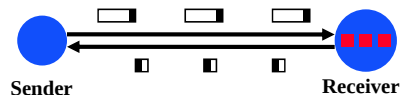
- Flow control
 - Stop & wait
 - Sliding window
- Loss recovery
 - Timeouts
 - Acknowledgement-driven recovery
 - Selective repeat versus selective repeat
 - Cumulative acknowledgement

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Window Flow Control

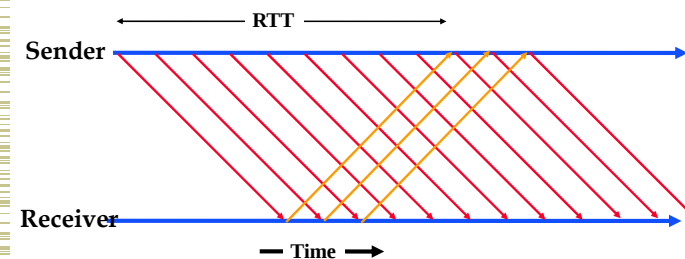


- Stop and wait flow control results in poor throughput for long-delay paths: packet size/ roundtrip-time.
- Solution: receiver provides sender with a window that it can fill with packets.
 - The window is backed up by buffer space on receiver
 - Receiver acknowledges the a packet every time a packet is consumed and a buffer is freed



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Bandwidth-Delay Product

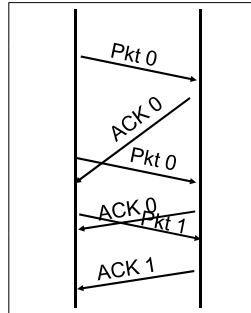


$$\text{Max Throughput} = \frac{\text{Window Size}}{\text{Roundtrip Time}}$$

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Automatic Repeat Request (ARQ)

- Sender retransmits packet after timeout
- Recognize retransmissions using sequence numbers
 - both packets and acks
 - How big should it be?
 - For stop&wait, sliding window?
- Ack can acknowledge one packet or all earlier packets
 - "Selective" vs cumulative



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Outline

- TCP flow control
- Congestion sources and collapse
- Congestion control basics

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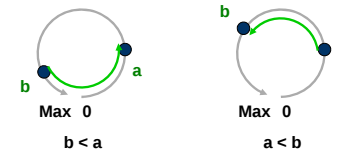
Sequence Numbers

- How large do sequence numbers need to be?
 - Must be able to detect wrap-around
 - Depends on sender/receiver window size
- Example
 - Assume seq number space = 7, window = 7
 - If pkts 0..6 are sent successfully and all acks lost
 - Receiver expects 7,0..5, sender retransmits old 0..6!!!
- Condition: max window size < size sequence number space

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Sequence Numbers

- 32 Bits, Unsigned → for bytes not packets!
 - Circular Comparison



- Why So Big?
 - For sliding window, must have $|\text{Sequence Space}| > |\text{Window}|$
 - No problem
 - Also, want to guard against stray packets
 - With IP, packets have maximum lifetime of 120s
 - Sequence number would wrap around in this time at 286MB/s

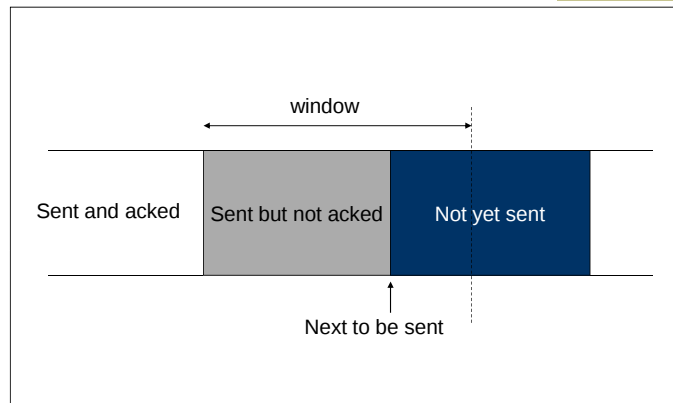
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TCP Flow Control

- TCP is a sliding window protocol
 - For window size n , can send up to n bytes without receiving an acknowledgement
 - When the data is acknowledged then the window slides forward
- Each packet advertises a window size
 - Indicates number of bytes the receiver has space for
- Original TCP always sent entire window
 - Congestion control now limits this

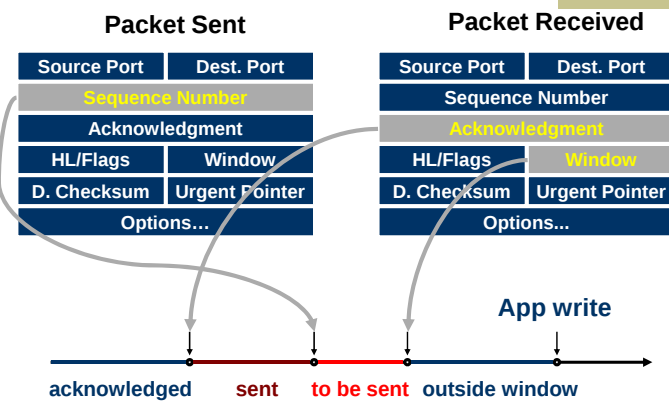
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Window Flow Control: Send Side



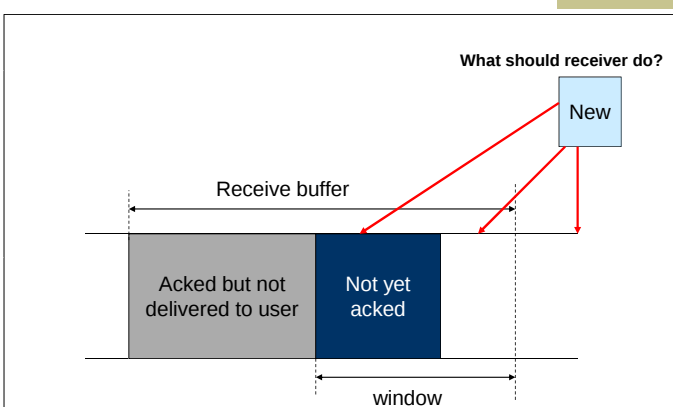
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Window Flow Control: Send Side



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Window Flow Control: Receive Side



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TCP Persist



- What happens if window is 0?
 - Receiver updates window when application reads data
 - What if this update is lost?
- TCP Persist state
 - Sender periodically sends 1 byte packets
 - Receiver responds with ACK even if it can't store the packet

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Performance Considerations



- The window size can be controlled by receiving application
 - Can change the socket buffer size from a default (e.g. 8Kbytes) to a maximum value (e.g. 64 Kbytes)
- The window size field in the TCP header limits the window that the receiver can advertise
 - 16 bits $\rightarrow$ 64 KBytes
 - 10 msec RTT $\rightarrow$ 51 Mbit/second
 - 100 msec RTT $\rightarrow$ 5 Mbit/second
 - TCP options to get around 64KB limit $\rightarrow$ increases above limit

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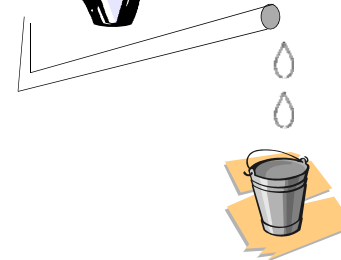
Outline



- TCP flow control
- Congestion sources and collapse
- Congestion control basics

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Internet Pipes?



- How should you control the faucet?

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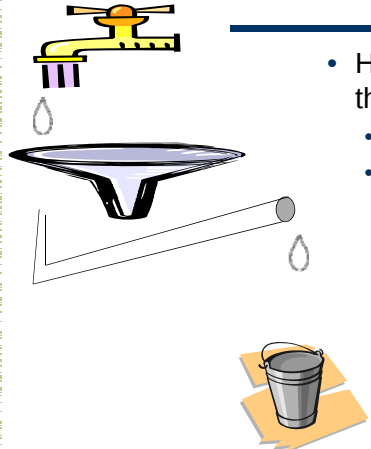
Internet Pipes?



- How should you control the faucet?
 - Too fast – sink overflows!

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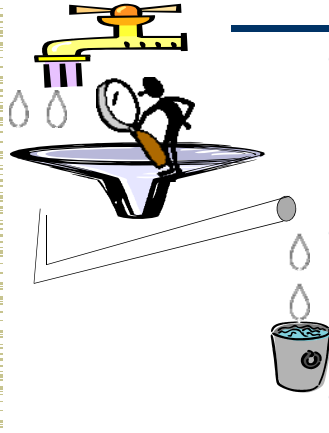
Internet Pipes?



- How should you control the faucet?
 - Too fast – sink overflows!
 - Too slow – what happens?

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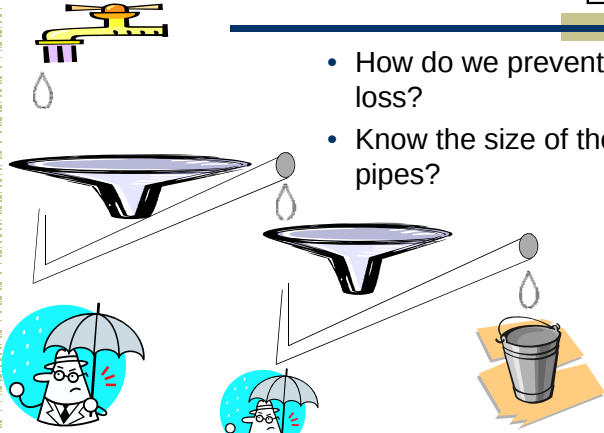
Internet Pipes?



- How should you control the faucet?
 - Too fast – sink overflows
 - Too slow – what happens?
- Goals
 - Fill the bucket as quickly as possible
 - Avoid overflowing the sink
- Solution – watch the sink

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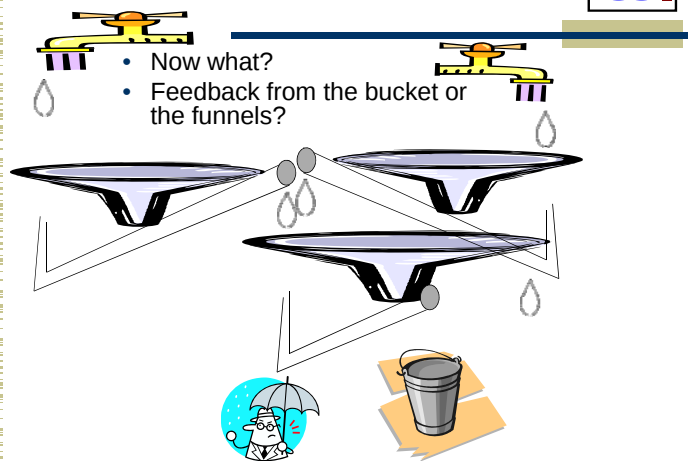
Plumbers Gone Wild!



- How do we prevent water loss?
- Know the size of the pipes?

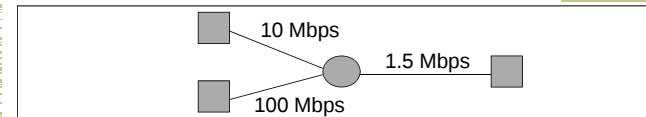
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Plumbers Gone Wild 2!



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Congestion

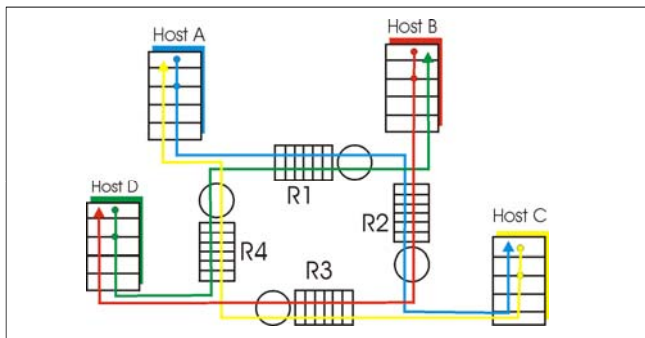


- Different sources compete for resources inside network
- Why is it a problem?
 - Sources are unaware of current state of resource
 - Sources are unaware of each other
- Manifestations:
 - Lost packets (buffer overflow at routers)
 - Long delays (queuing in router buffers)
 - Can result in throughput less than bottleneck link (1.5Mbps for the above topology) → a.k.a. congestion collapse

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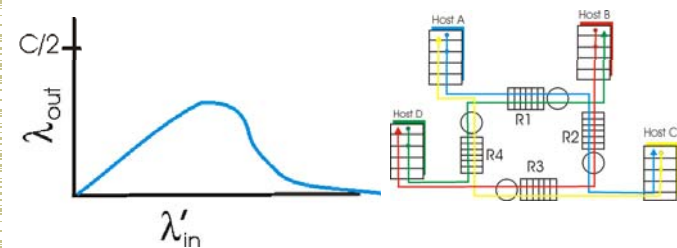
Causes & Costs of Congestion

- Four senders – multihop paths Q: What happens as rate increases?
- Timeout/retransmit



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Causes & Costs of Congestion



- When packet dropped, any “upstream transmission capacity used for that packet was wasted!

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Congestion Collapse



- Definition: *Increase in network load results in decrease of useful work done*
- Many possible causes
 - Spurious retransmissions of packets still in flight
 - Classical congestion collapse
 - How can this happen with packet conservation
 - Solution: better timers and TCP congestion control
 - Undelivered packets
 - Packets consume resources and are dropped elsewhere in network
 - Solution: congestion control for ALL traffic

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Congestion Control and Avoidance



- A mechanism that:
 - Uses network resources efficiently
 - Preserves fair network resource allocation
 - Prevents or avoids collapse
- Congestion collapse is not just a theory
 - Has been frequently observed in many networks

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Approaches Towards Congestion Control



- Two broad approaches towards congestion control:
- | | |
|--|--|
| <ul style="list-style-type: none">• End-end congestion control:<ul style="list-style-type: none">• No explicit feedback from network• Congestion inferred from end-system observed loss, delay• Approach taken by TCP | <ul style="list-style-type: none">• Network-assisted congestion control:<ul style="list-style-type: none">• Routers provide feedback to end systems<ul style="list-style-type: none">• Single bit indicating congestion (SNA, DECbit, TCP/IP ECN, ATM)• Explicit rate sender should send at• Problem: makes routers complicated |
|--|--|

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Example: TCP Congestion Control



- Very simple mechanisms in network
 - FIFO scheduling with shared buffer pool
 - Feedback through packet drops
- TCP interprets packet drops as signs of congestion and slows down
 - This is an assumption: packet drops are not a sign of congestion in all networks
 - E.g. wireless networks
- Periodically probes the network to check whether more bandwidth has become available.

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Objectives



- Simple router behavior
- Distributedness
- Efficiency: $X = \sum x_i(t)$
- Fairness: $(\sum x_i)^2 / n(\sum x_i^2)$
 - What are the important properties of this function?
- Convergence: control system must be stable

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Basic Control Model



- Reduce speed when congestion is perceived
 - How is congestion signaled?
 - Either mark or drop packets
 - How much to reduce?
- Increase speed otherwise
 - Probe for available bandwidth – how?

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Linear Control

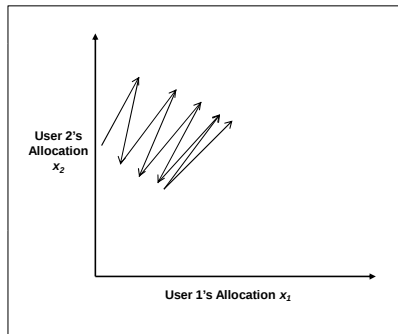


- Many different possibilities for reaction to congestion and probing
 - Examine simple linear controls
 - $\text{Window}(t + 1) = a + b \text{Window}(t)$
 - Different a_i/b_i for increase and a_d/b_d for decrease
- Supports various reaction to signals
 - Increase/decrease additively
 - Increased/decrease multiplicatively
 - Which of the four combinations is optimal?

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Phase Plots

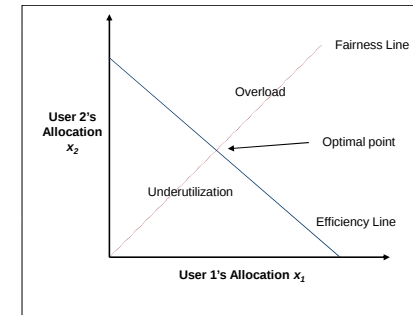
- Simple way to visualize behavior of competing connections over time



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Phase Plots

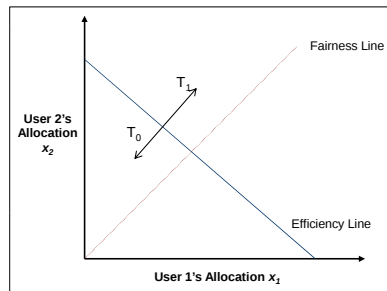
- What are desirable properties?
- What if flows are not equal?



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Additive Increase/Decrease

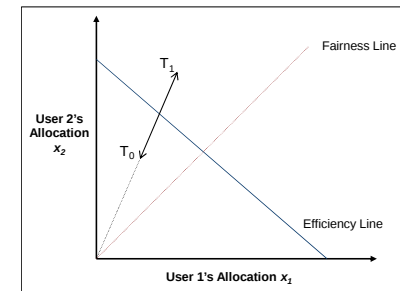
- Both x_1 and x_2 increase/ decrease by the same amount over time
 - Additive increase improves fairness and additive decrease reduces fairness



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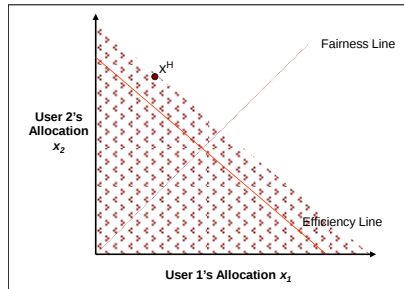
Multiplicative Increase/Decrease

- Both x_1 and x_2 increase by the same factor over time
 - Extension from origin – constant fairness



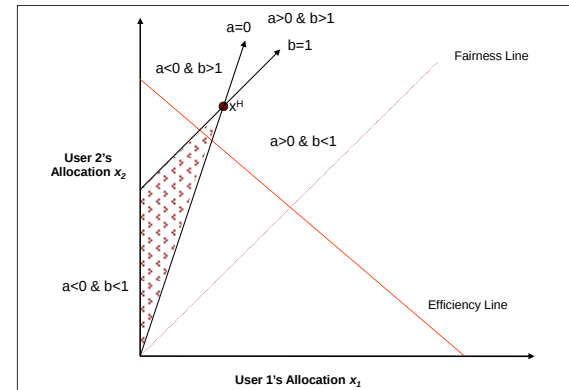
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Convergence to Efficiency



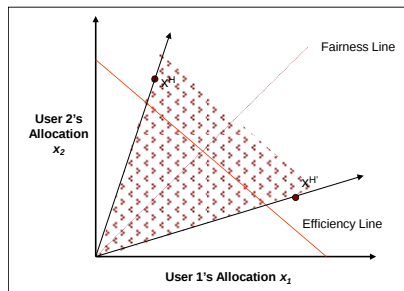
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Distributed Convergence to Efficiency



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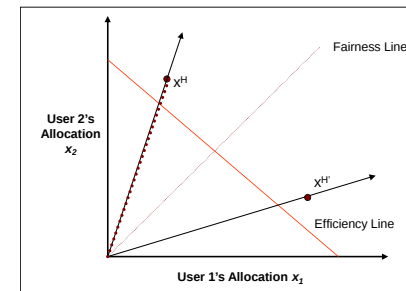
Convergence to Fairness



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Convergence to Efficiency & Fairness

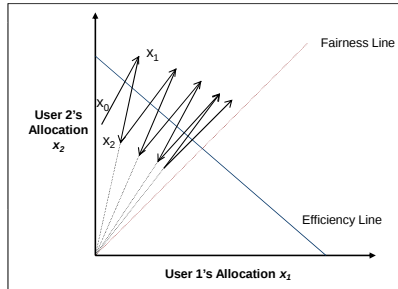
- Intersection of valid regions
- For decrease: $a=0$ & $b < 1$



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What is the Right Choice?

- Constraints limit us to AIMD
 - Can have multiplicative term in increase (MAIMD)
 - AIMD moves towards optimal point



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Important Lessons

- Transport service
 - UDP → mostly just IP service
 - TCP → congestion controlled, reliable, byte stream
- Types of ARQ protocols
 - Stop-and-wait → slow, simple
 - Go-back-n → can keep link utilized (except w/ losses)
 - Selective repeat → efficient loss recovery
- Sliding window flow control
- TCP flow control
 - Sliding window → mapping to packet headers
 - 32bit sequence numbers (bytes)

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Important Lessons

- Why is congestion control needed?
- How to evaluate congestion control algorithms?
 - Why is AIMD the right choice for congestion control?
- TCP flow control
 - Sliding window → mapping to packet headers
 - 32bit sequence numbers (bytes)

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Good Ideas So Far...

- Flow control
 - Stop & wait
 - Parallel stop & wait
 - Sliding window
- Loss recovery
 - Timeouts
 - Acknowledgement-driven recovery
 - Selective repeat versus go-back-N
 - Cumulative acknowledgement
- Congestion control
 - AIMD → fairness and efficiency
- Next Lecture: How does TCP actually implement these?

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