EECS 489 Computer Networks

Winter 2024

Mosharaf Chowdhury

Material with thanks to Aditya Akella, Sugih Jamin, Philip Levis, Sylvia Ratnasamy, Peter Steenkiste, and many other colleagues.

Agenda

- TCP congestion control wrap-up
- TCP throughput equation
- Problems with congestion control

Recap

Flow Control

Restrict window to RWND to make sure that the receiver isn't overwhelmed

Congestion Control

Restrict window to CWND to make sure that the network isn't overwhelmed

Together

Restrict window to min{RWND, CWND} to make sure that neither the receiver nor the network are overwhelmed

CC Implementation

- States at sender
 - CWND (initialized to a small constant)
 - ssthresh (initialized to a large constant)
 - > Timer
- Events
 - ACK (new data)
 - > Timeout

Event: ACK (new data)

- If CWND < ssthresh
 - > CWND += 1 ____

- CWND packets per RTT
- Hence, after one RTT with no drops:
 CWND = 2xCWND

Event: ACK (new data)

- If CWND < ssthresh
 - > CWND += 1

Slow start phase

- Else
 - CWND = CWND + 1/CWND

Congestion avoidance phase

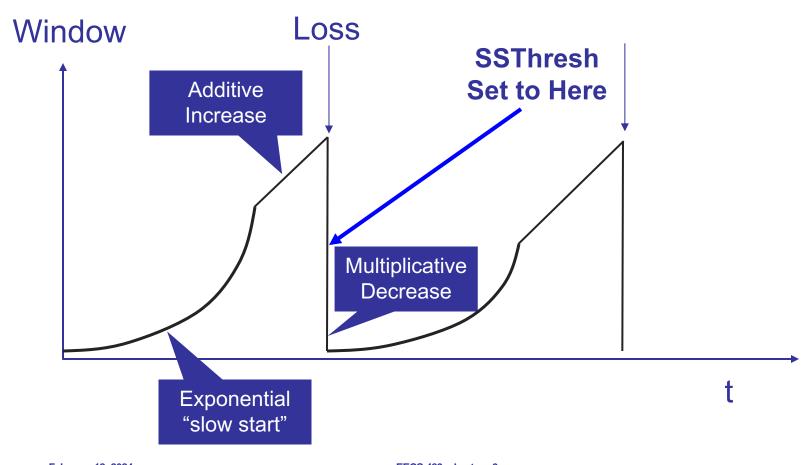
- CWND packets per RTT
- Hence, after one RTT with no drops:

CWND = CWND + 1

Event: TimeOut

- On Timeout
 - > ssthresh ← CWND/2
 - > CWND ← 1

AIMD leads to TCP sawtooth



CC Implementation (cont.)

States at sender

- CWND (initialized to a small constant)
- ssthresh (initialized to a large constant)
- Timer
- dupACKcount

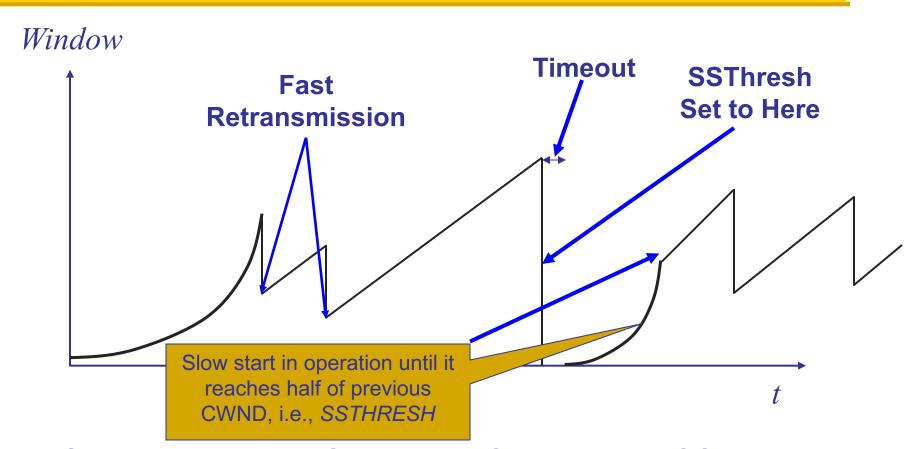
Events

- ACK (new data)
- > Timeout
- dupACK (duplicate ACK for old data)

Event: dupACK

- dupACKcount ++
- If dupACKcount = 3 /* fast retransmit */
 - > ssthresh = CWND/2
 - > CWND = CWND/2

Example



Slow-start restart: Go back to CWND = 1 MSS, but take advantage of knowing the previous value of CWND

Why AIMD?

- Recall the three issues
 - Finding available bottleneck bandwidth
 - Adjusting to bandwidth variations
 - Sharing bandwidth

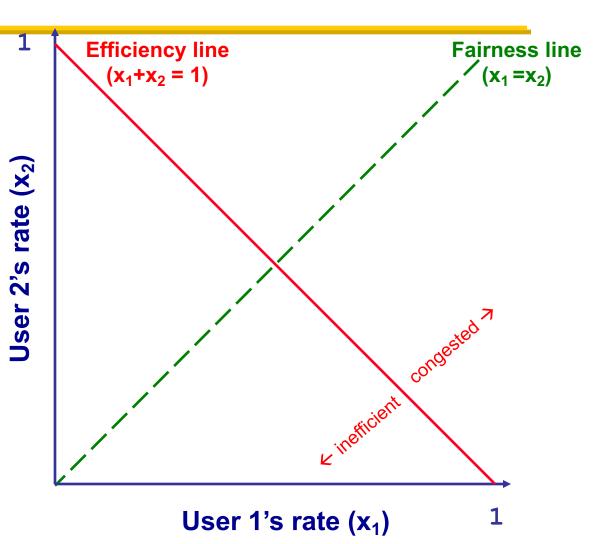
- Two goals for bandwidth sharing
 - Efficiency: High utilization of link bandwidth
 - Fairness: Each flow gets equal share

Why AIMD?

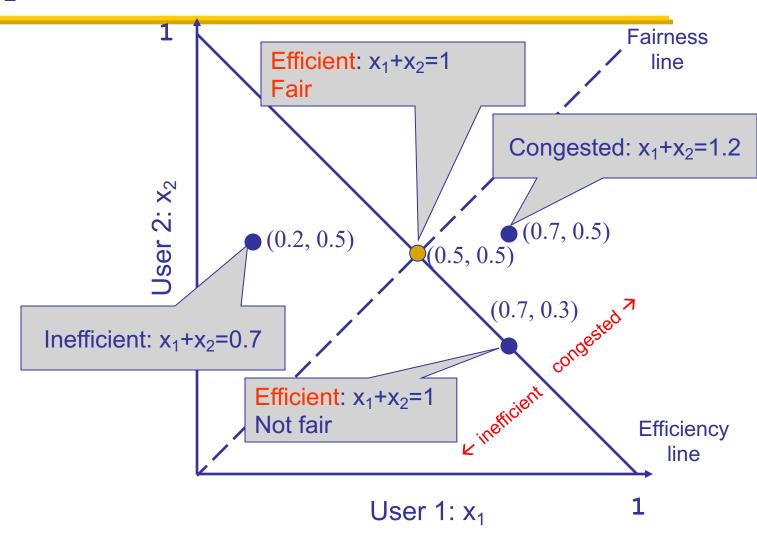
- Every RTT, we can do
 - Multiplicative increase or decrease: CWND→ a*CWND
 - ➤ Additive increase or decrease: CWND→ CWND + b
- Four alternatives:
 - > AIAD: gentle increase, gentle decrease
 - > AIMD: gentle increase, drastic decrease
 - MIAD: drastic increase, gentle decrease
 - MIMD: drastic increase and decrease

Simple model of congestion control

- Two users
 - rates x1 and x2
- Congestion when x1+x2 > 1
- Unused capacity
 when x1+x2 < 1
- Fair when x1 = x2



Example

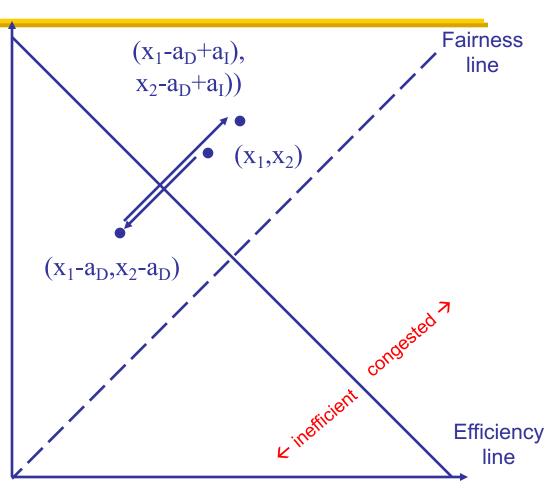


AIAD

Increase: x + a_I

Decrease: x - a_D

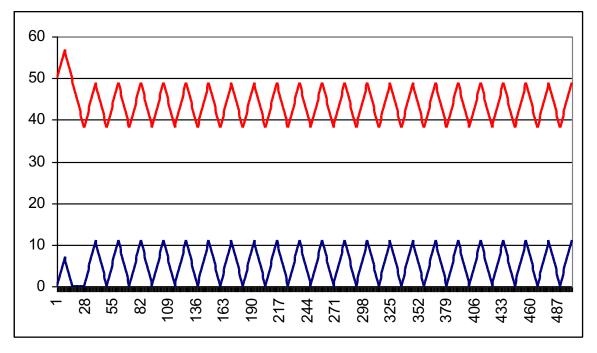
Does not converge to fairness



User 1: x₁

AIAD Sharing Dynamics





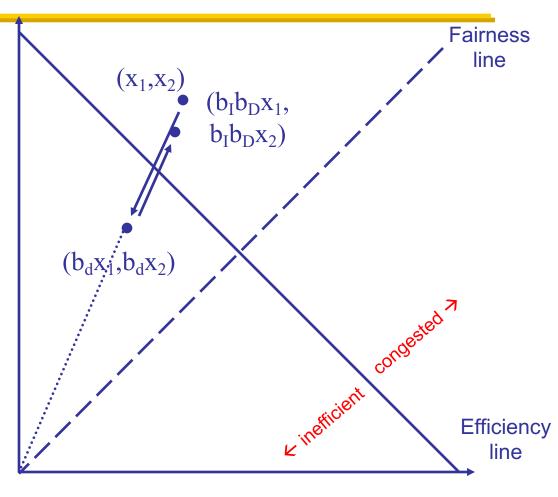
MIMD

Increase: x*b_I

Decrease: x*b_D

 Does not converge to fairness

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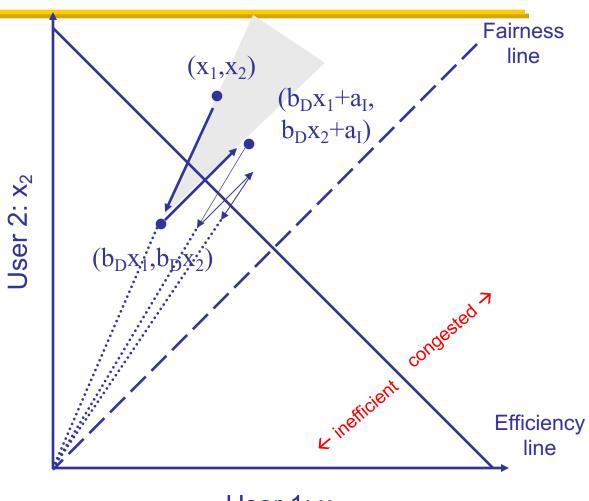
User 1: x₁

AIMD

• Increase: x+a₁

Decrease: x*b_D

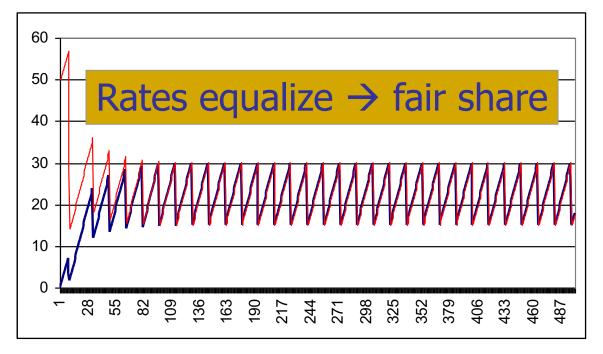
Converges to fairness



User 1: x₁

AIMD Sharing Dynamics



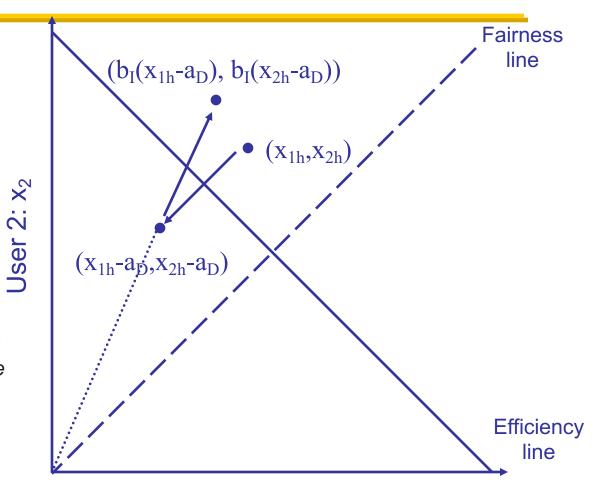


MIAD

Increase: x*b_I

Decrease: x - a_D

- Does not converge to fairness
- Does not converge to efficiency
- "Analysis of the Increase and Decrease Algorithms for Congestion Avoidance in Computer Networks"
 - -- Chiu and Jain



User 1: x₁

Not done yet!

 Problem: congestion avoidance too slow in recovering from an isolated loss

Example

- Consider a TCP connection with:
 - CWND=10 packets
 - Last ACK was for packet # 101
 - »i.e., receiver expecting next packet to have seq. no. 101
- 10 packets [101, 102, 103,..., 110] are in flight
 - Packet 101 is dropped

Timeline: [1X1, 102, ..., 110]

- ACK 101 (due to 102) cwnd=10 dupACK#1 (no xmit)
- ACK 101 (due to 103) cwnd=10 dupACK#2 (no xmit)
- ACK 101 (due to 104) cwnd=10 dupACK#3 (no xmit)
- RETRANSMIT 101 ssthresh=5 cwnd= 5
- ACK 101 (due to 105) cwnd=5 + 1/5 (no xmit)
- ACK 101 (due to 106) cwnd=5 + 2/5 (no xmit)
- ACK 101 (due to 107) cwnd=5 + 3/5 (no xmit)
- ACK 101 (due to 108) cwnd=5 + 4/5 (no xmit)
- ACK 101 (due to 109) cwnd=5 + 5/5 (no xmit)
- ACK 101 (due to 110) cwnd=6 + 1/6 (no xmit)
- ◆ ACK 111 (due to 101)
 ← only now can we transmit new packets
- Plus no packets in flight so ACK "clocking" (to increase CWND) stalls for another RTT

Solution: Fast recovery

- Idea: Grant the sender temporary "credit" for each dupACK so as to keep packets in flight
- If dupACKcount = 3
 - ssthresh = CWND/2
 - CWND = ssthresh + 3
- While in fast recovery
 - CWND = CWND + 1 for each additional dupACK
- Exit fast recovery after receiving new ACK
 - set CWND = ssthresh

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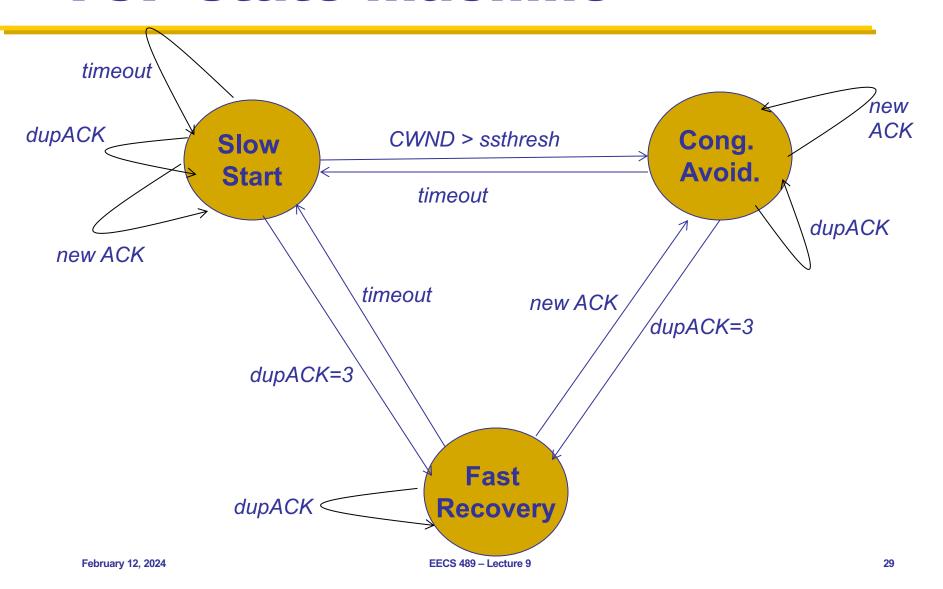
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- ACK 101 (due to 104) cwnd=10 dup#3
- RETRANSMIT 101 ssthresh=5 cwnd= 8 (5+3)
- ACK 101 (due to 105) cwnd= 9 (no xmit)
- ACK 101 (due to 106) cwnd=10 (no xmit)
- ACK 101 (due to 107) cwnd=11 (xmit 111)
- ACK 101 (due to 108) cwnd=12 (xmit 112)
- ACK 101 (due to 109) cwnd=13 (xmit 113)
- ACK 101 (due to 110) cwnd=14 (xmit 114)
- ACK 111 (due to 101) cwnd = 5 (xmit 115) ← exiting fast recovery
- Packets 111-114 already in flight
- ACK 112 (due to 111) cwnd = $5 + 1/5 \leftarrow$ back in cong. avoidance

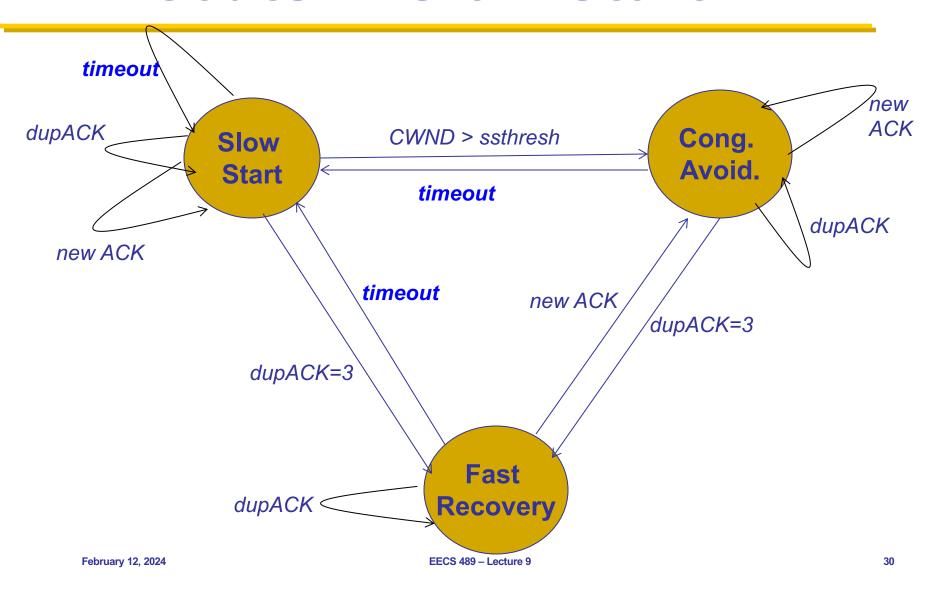
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5-MINUTE BREAK!

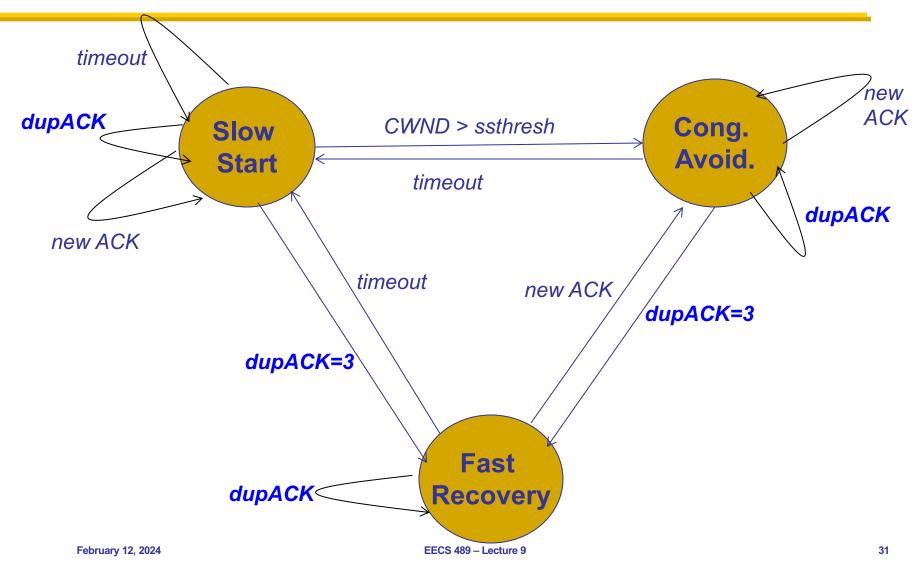
TCP state machine



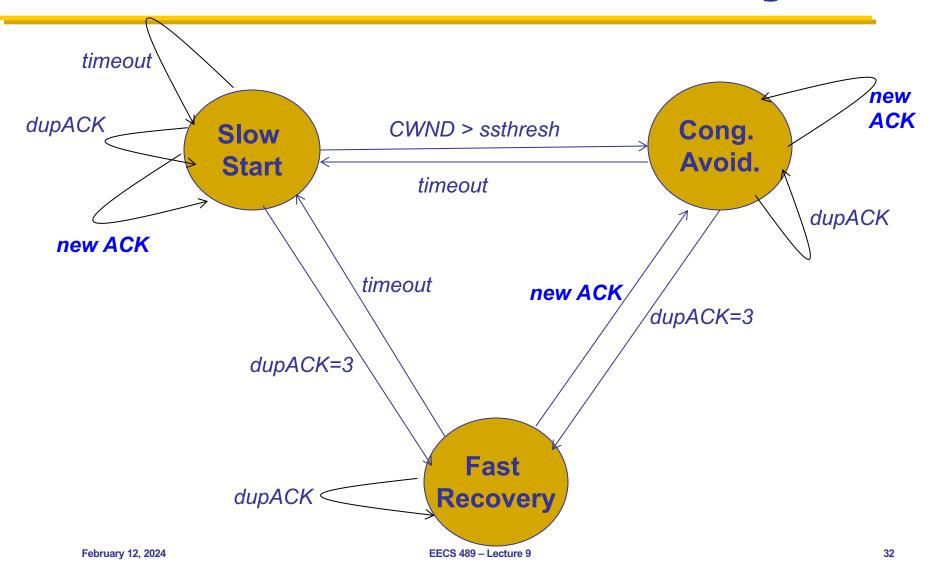
Timeouts → **Slow Start**



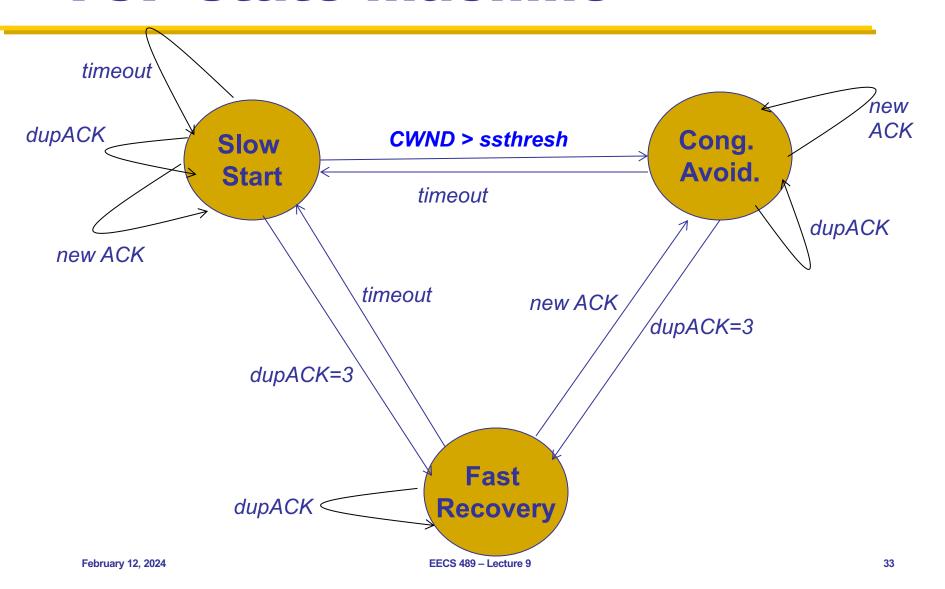
dupACKs → Fast Recovery



New ACK changes state ONLY from Fast Recovery



TCP state machine



TCP flavors

- TCP-Tahoe
 - > CWND =1 on 3 dupACKs
- TCP-Reno
 - CWND =1 on timeout
 - CWND = CWND/2 on 3 dupACKs
- TCP-newReno__
 - > TCP-Reno + improved fast recovery
- TCP-SACK
 - > Incorporates selective acknowledgements

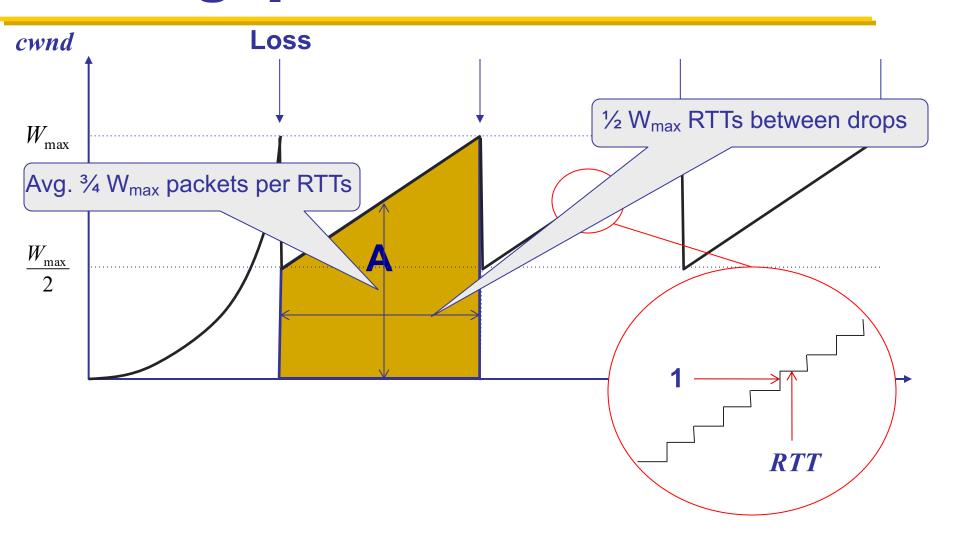
Our default assumption

How can they coexist?

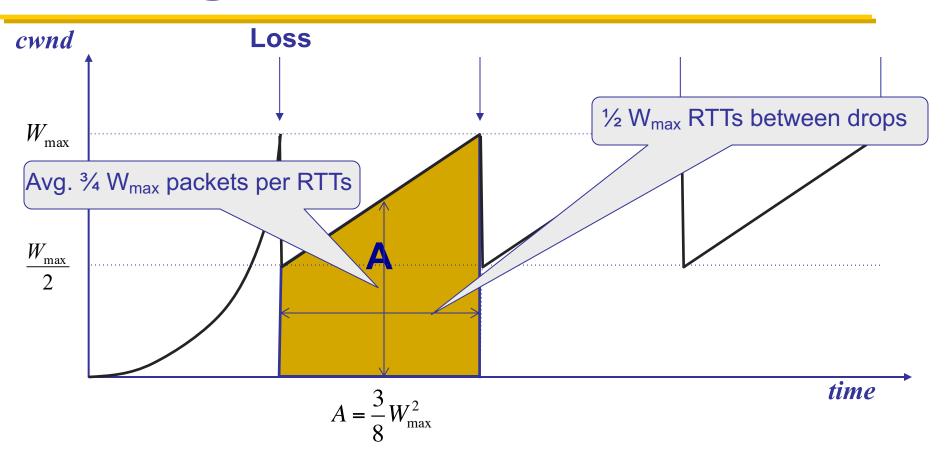
- All follow the same principle
 - Increase CWND on good news
 - Decrease CWND on bad news

TCP THROUGHPUT EQUATION

A simple model for TCP throughput



A simple model for TCP throughput

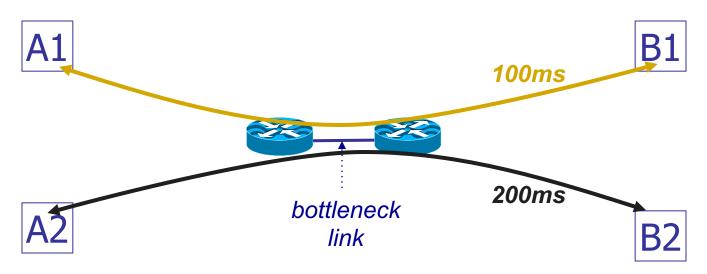


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Implications (1): Different RTTs

Throughput =
$$\sqrt{\frac{3}{2}} \frac{1}{RTT\sqrt{p}}$$

- Flows get throughput inversely proportional to RTT
- TCP unfair in the face of heterogeneous RTTs!



Implications (2): High-speed TCP

Throughput =
$$\sqrt{\frac{3}{2}} \frac{1}{RTT\sqrt{p}}$$

- Assume RTT = 100ms, MSS=1500bytes, BW=100Gbps
- What value of p is required to reach 100Gbps throughput?
 - $> \sim 2 \times 10^{-12}$
- How long between drops?
 - > ~ 16.6 hours
- How much data has been sent in this time?
 - > ~ 6 petabits

Implications (3): Rate-based CC

Throughput =
$$\sqrt{\frac{3}{2}} \frac{1}{RTT\sqrt{p}}$$

- TCP throughput is swings between W/2 to W
- Apps may prefer steady rates (e.g., streaming)
- "Equation-Based Congestion Control"
 - Just follow the equation
 - Measure drop percentage p, and set rate accordingly
- Following the TCP equation ensures "TCP friendliness"
 - > i.e., use no more than TCP does in similar setting

Implications (4): Loss not due to congestion?

- TCP will confuse corruption with congestion
- Flow will cut its rate
 - Throughput ~ 1/sqrt(p) where p is loss prob.
 - Applies even for non-congestion losses!

Implications (5): Short flows cannot ramp up

- 50% of flows have < 1500B to send; 80% < 100KB
- Implications
 - Short flows never leave slow start!
 - »They never attain their fair share
 - > Too few packets to trigger dupACKs
 - »Isolated loss may lead to timeouts
 - »At typical timeout values of ~500ms, might severely impact flow completion time

Implications (6): Short flows share long delays

- A flow deliberately overshoots capacity, until it experiences a drop
- Means that delays are large, and are large for everyone
 - Consider a flow transferring a 10GB file sharing a bottleneck link with 10 flows transferring 100B
 - Larger flows dominate smaller ones

Implications (7): Cheating

- Three easy ways to cheat
 - Increasing CWND faster than +1 MSS per RTT

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 - Increasing CWND faster than +1 MSS per RTT
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 - »Common practice by many companies
 - Opening many connections

Open many connections



Assume

- A starts 10 connections to B
- D starts 1 connection to E
- Each connection gets about the same throughput

Then A gets 10 times more throughput than D

Implications (8): CC intertwined with reliability

- CWND adjusted based on ACKs and timeouts
- Cumulative ACKs and fast retransmit/recovery rules
- Complicates evolution
 - Changing from cumulative to selective ACKs is hard
- Sometimes we want CC but not reliability
 - > e.g., real-time applications
- We may also want reliability without CC

Recap: TCP problems

- Miśled by non-congestion losses
- Fills up queues leading to high delays
- Short flows complete before discovering available capacity
- AIMD impractical for high speed links
- Saw tooth discovery too choppy for some apps
- Unfair under heterogeneous RTTs
- Tight coupling with reliability-mechanisms
- End hosts can cheat

Routers tell endpoints if they're congested

Routers tell endpoints what rate to send at

Routers enforce fair sharing

Could fix many of these with some help from routers!

Summary

- TCP works even though it has many flaws
- Many of them can be fixed via assistance from the network

Next: The Network Layer

Adapting TCP to high speed

- Once past a threshold speed, increase CWND faster
 - A proposed standard [Floyd'03]: once speed is past some threshold, change equation to p^{-.8} rather than p^{-.5}
 - Let the additive constant in AIMD depend on CWND
- Other approaches?
 - Multiple simultaneous connections (hack but works today)
 - Router-assisted approaches