Network Transport Layer:
TCP/Reno Analysis, TCP Cubic, TCP/Vegas

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https://qiaoxiang.me/courses/cnns-xmuf21/index.shtml

11/23/2021

This deck of slides are heavily based on CPSC 433/533 at Yale University, by courtesy of Dr. Y. Richard Yang.

Admin.

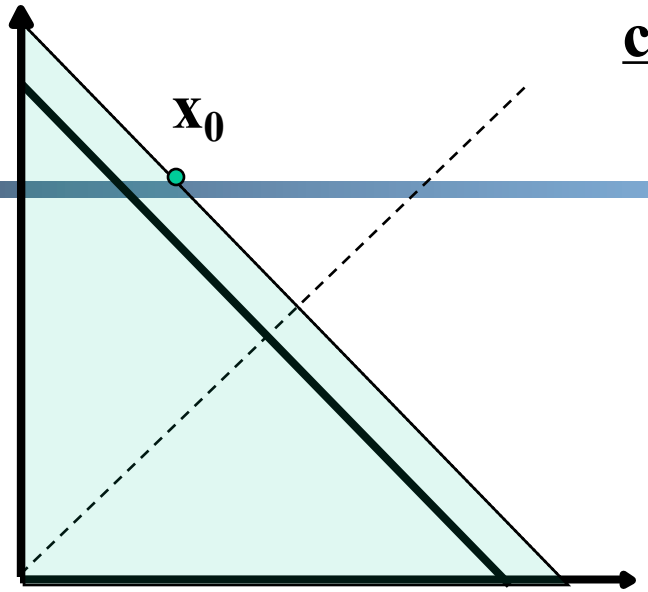
- Lab assignment 5 posted, due on Dec. 31
- Final exam date: 2-4pm, Jan. 5

Recap: Transport Design

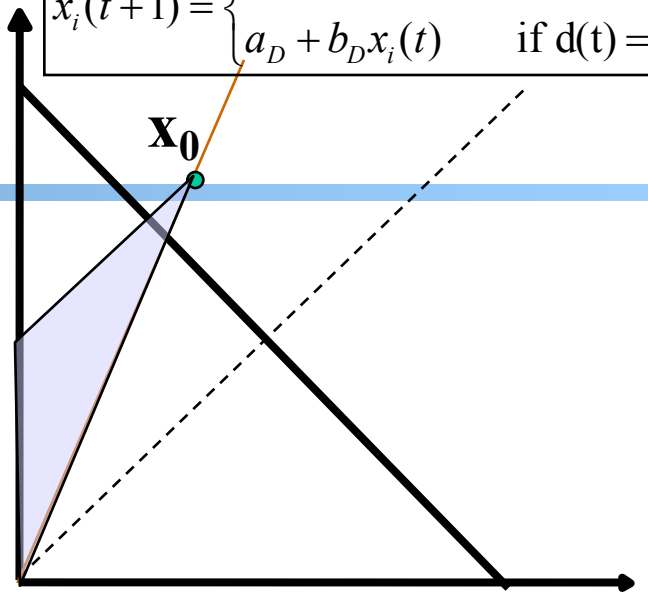
- Basic structure/reliability: sliding window protocols
- Determine the "right" parameters
  - Timeout
    - mean + variation
  - Sliding window size
    - Related w/ congestion control or more generally resource allocation
      - Bad congestion control can lead to congestion collapse (e.g., zombie packets)
    - Goals: distributed algorithm to achieve fairness and efficiency

congestion

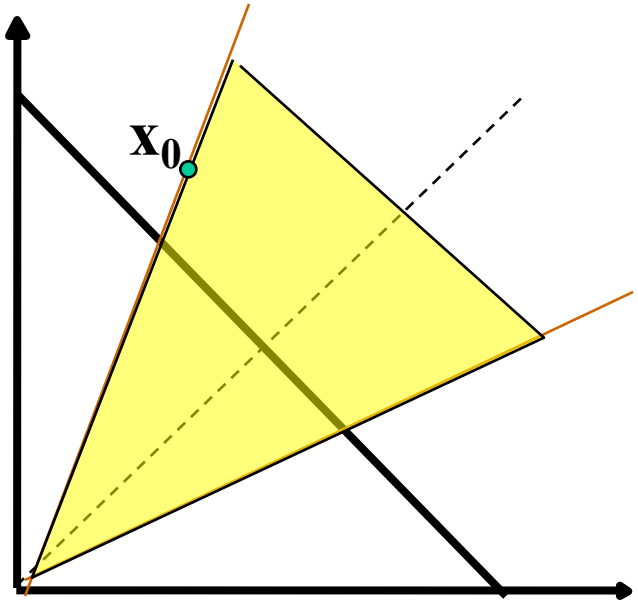
$$x_i(t+1) = \begin{cases} a_I + b_I x_i(t) & \text{if d(t) = no cong.} \\ a_D + b_D x_i(t) & \text{if d(t) = cong.} \end{cases}$$



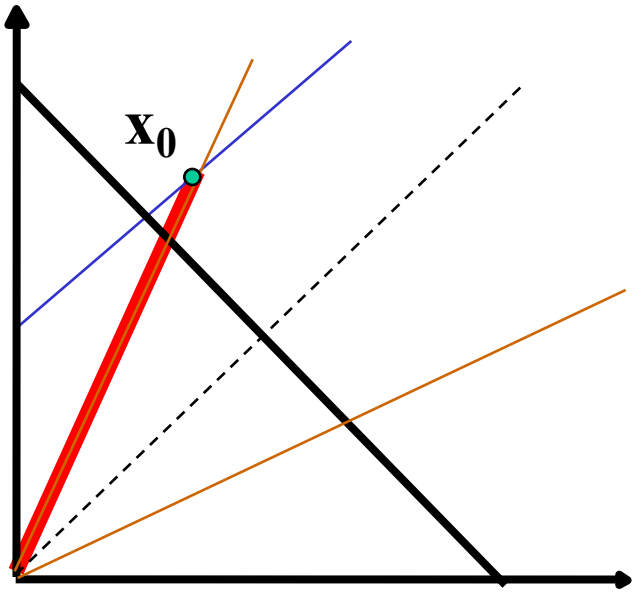
efficiency



efficiency: distributed linear rule



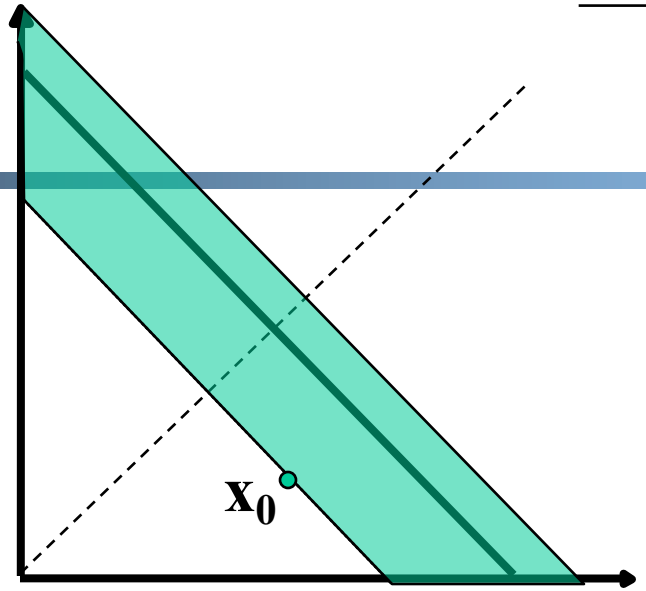
fairness



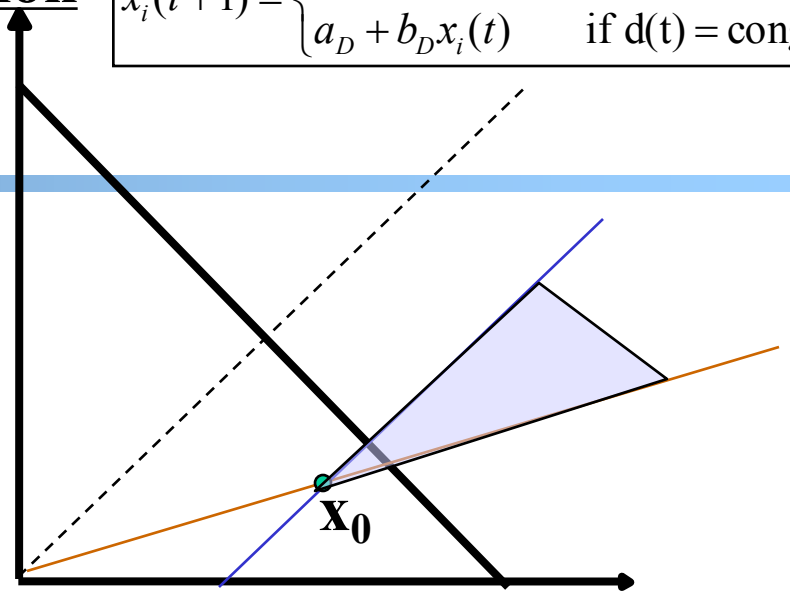
intersection

no-congestion

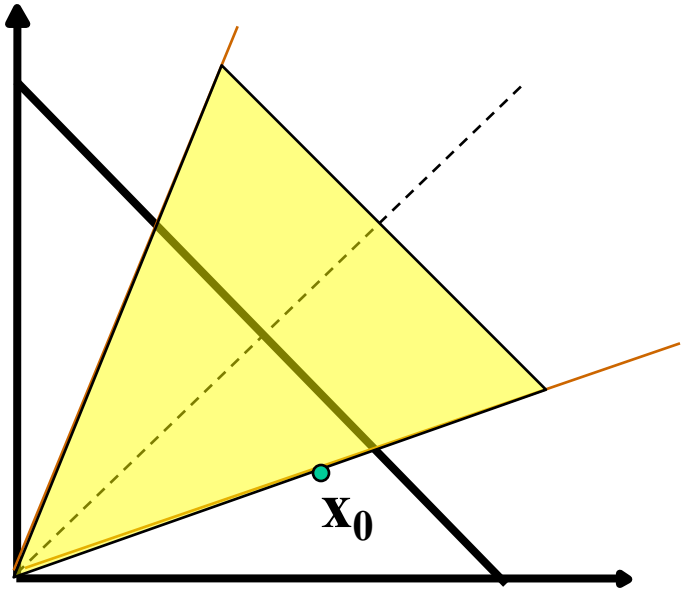
$$x_i(t+1) = \begin{cases} a_I + b_I x_i(t) & \text{if d(t) = no cong.} \\ a_D + b_D x_i(t) & \text{if d(t) = cong.} \end{cases}$$



efficiency



efficiency: distributed linear rule



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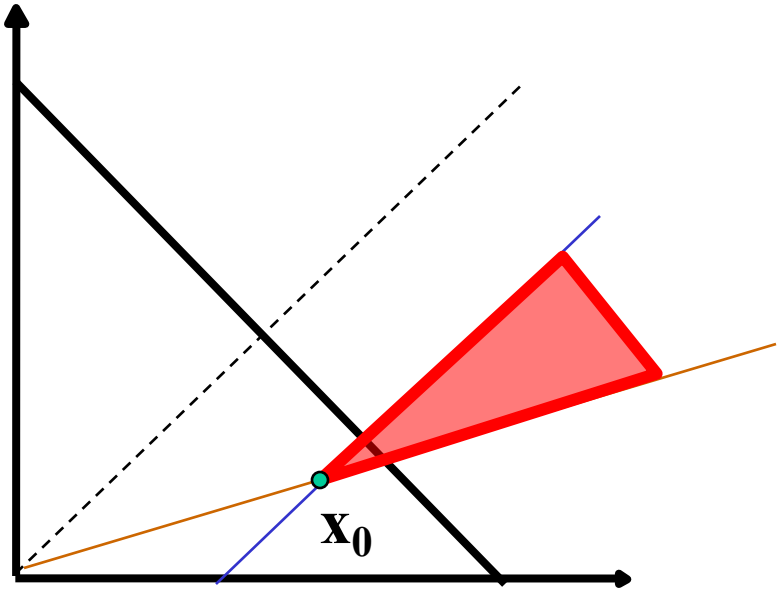


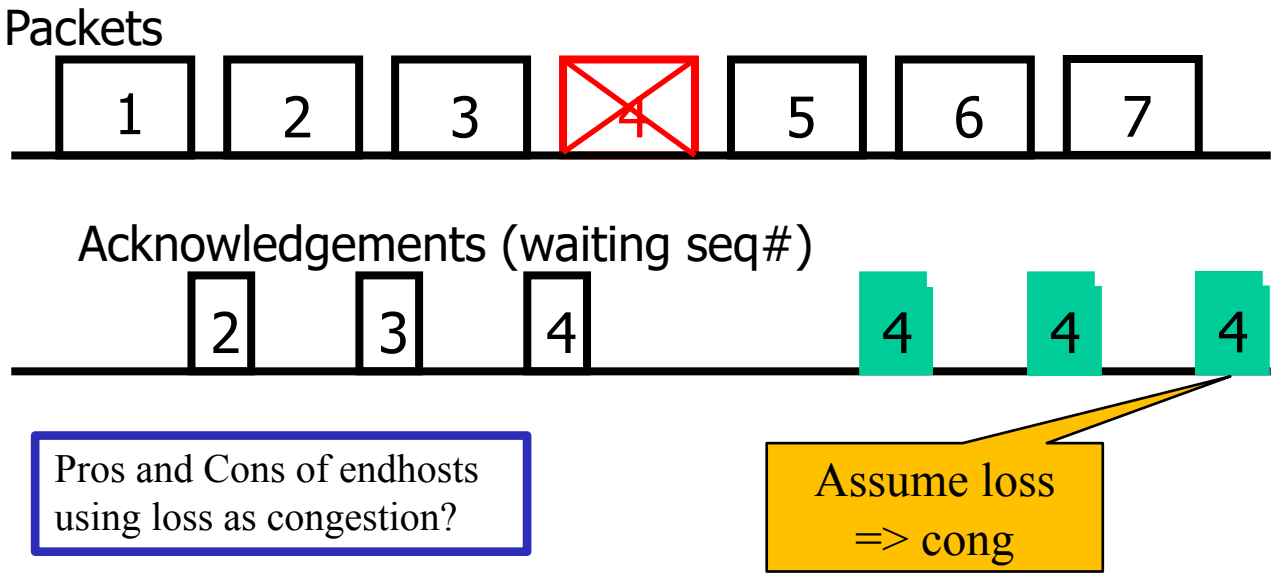
convergence

Mapping A(M)I-MD to Protocol

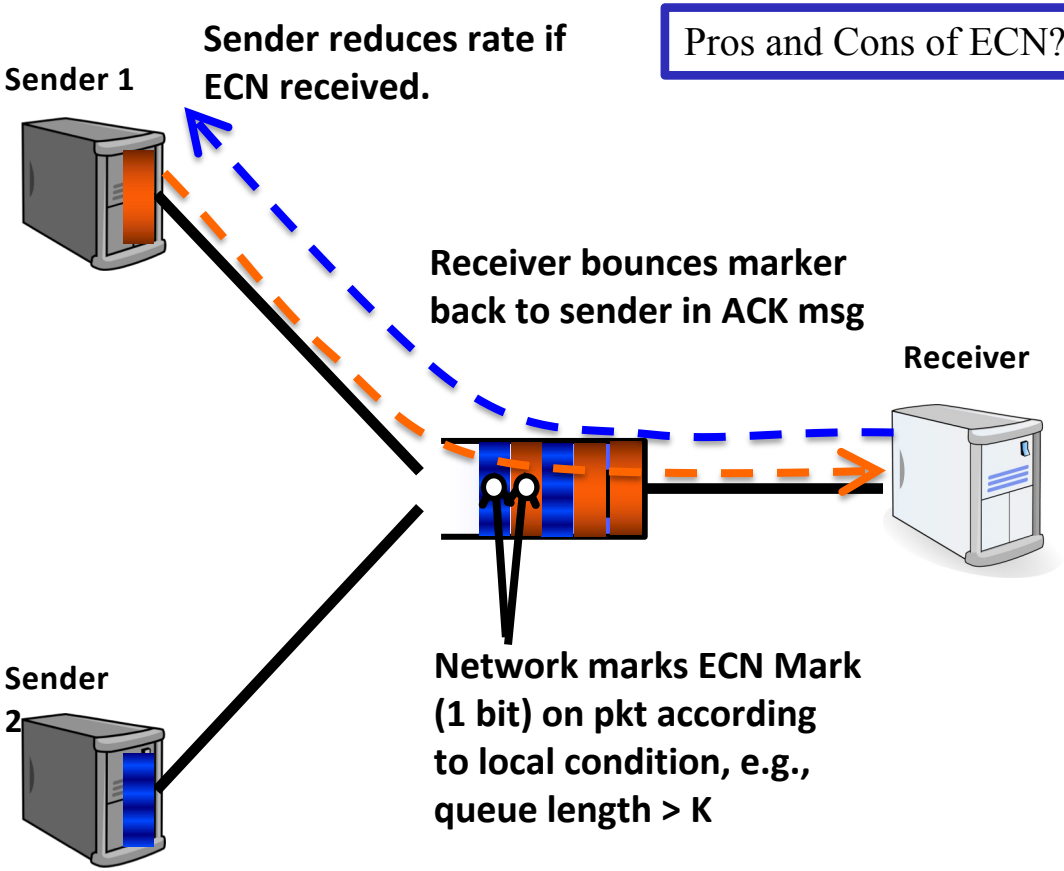
- Basic questions to look at:
  - How to obtain d(t)--the congestion signal?
  - What values do we choose for the formula?
  - How to map formula to code?

$$x_i(t+1) = \begin{cases} a_I + x_i(t) & \text{if } \mathrm{d(t)} = \text{no cong.} \\ b_D x_i(t) & \text{if } \mathrm{d(t)} = \text{cong.} \end{cases}$$

Obtain d(t) Approach 1: End Hosts Consider Loss as Congestion



Obtain d(t) Approach 2: Network Feedback (ECN: Explicit Congestion Notification)



Mapping A(M)I-MD to Protocol

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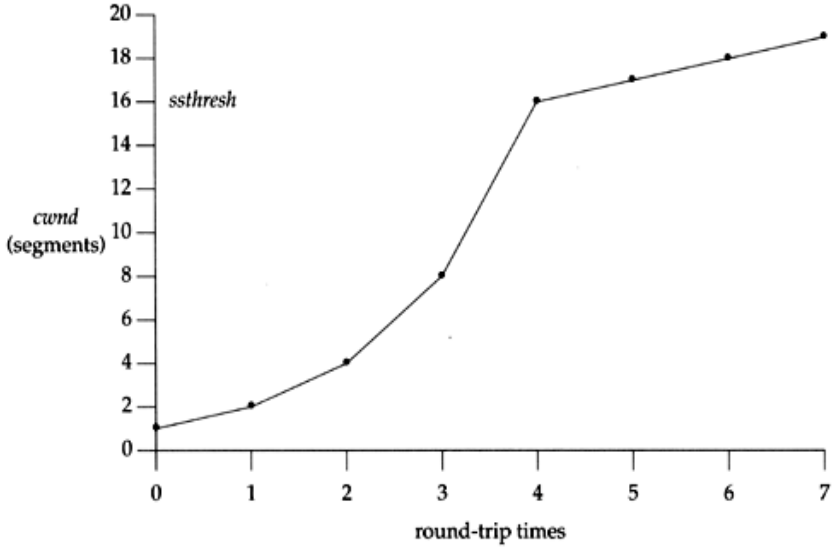
TCP/Reno Formulas

- Multiplicative Increase (MI)
  - double *the rate:* $x(t+1) = 2\ x(t)$
- Additive Increase (AI)
  - Linear increase *the rate:* $x(t+1) = x(t) + 1$
- Multiplicative decrease (MD)
  - half *the rate:* $x(t+1) = 1/2\ x(t)$

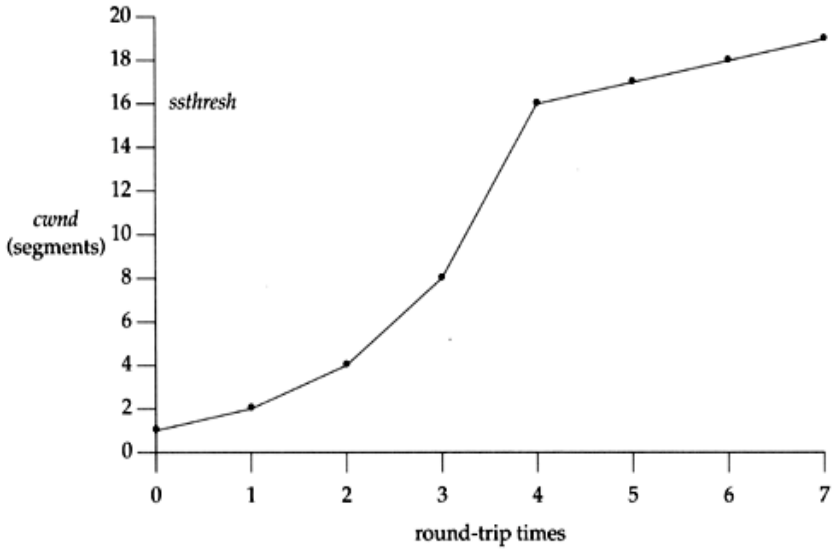
TCP/Reno Formula Switching (Control Structure)

- Two "phases"
  - slow-start
    - Goal: getting to equilibrium gradually but quickly, to get a rough estimate of the optimal of *cwnd*
    - *Formula: MI*
  - congestion avoidance
    - Goal: Maintains equilibrium and reacts around equilibrium
    - Formula: AI MD



TCP/Reno Formula Switching (Control Structure)

- Important variables:
  - **`cwnd:`** congestion window size
  - **`ssthresh:`** threshold between the slow-start phase and the congestion avoidance phase
- If cwnd < ssthresh
  - MI
- Else
  - AIMD



MI: Slow Start

- Algorithm: MI
  - double *cwnd* every RTT until network congested

- Goal: getting to equilibrium gradually but quickly, to get a rough estimate of the optimal of *cwnd*

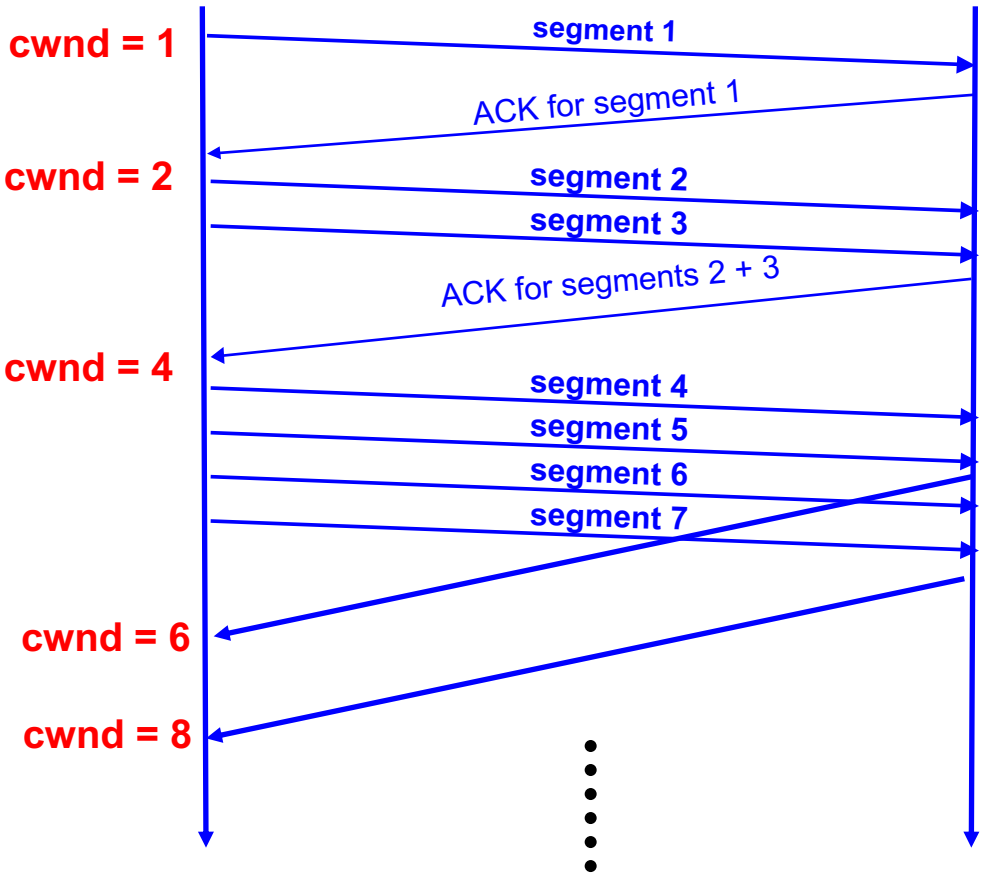
MI: Slow-start

Initially:

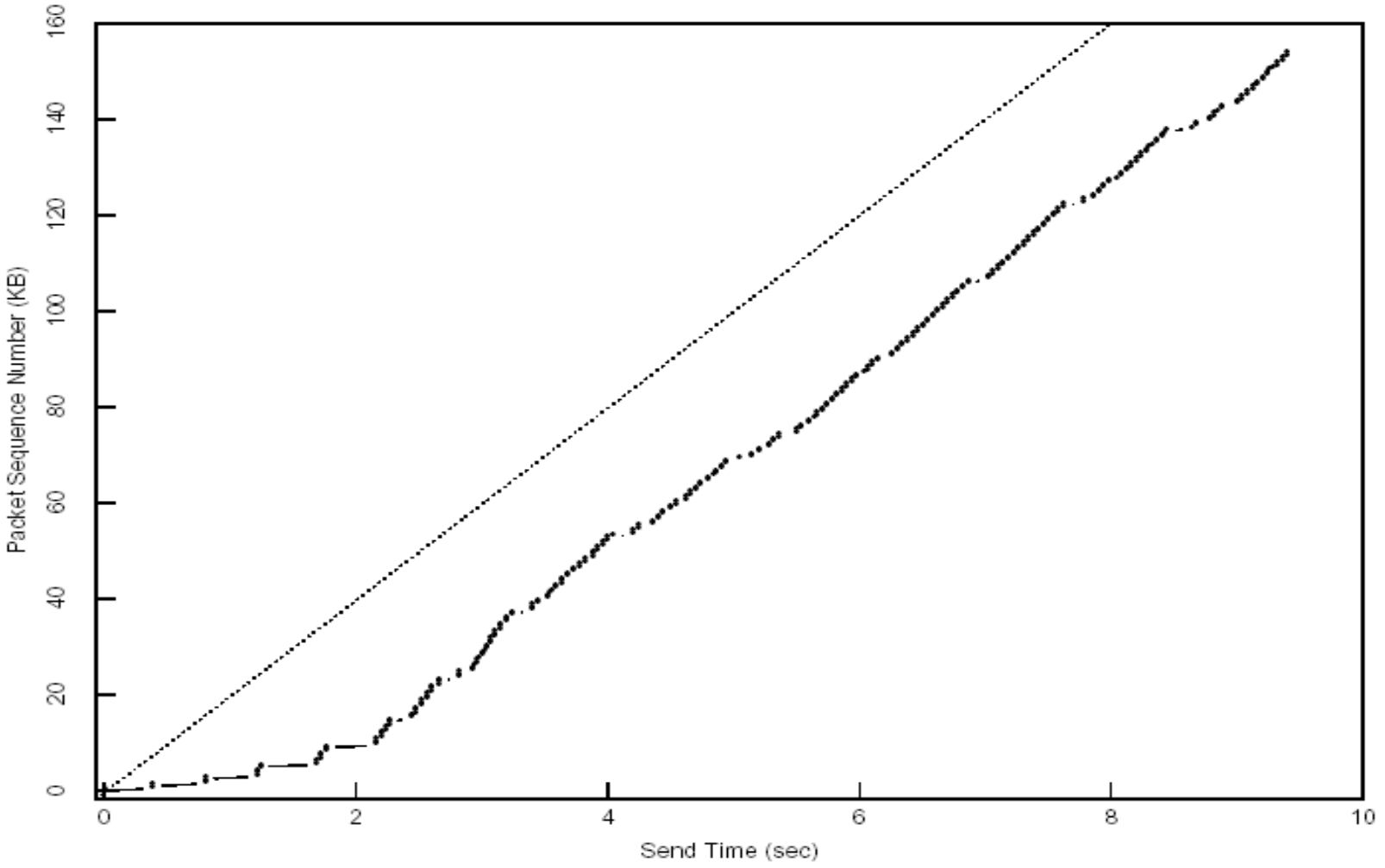
```
cwnd = 1;
ssthresh = infinite (e.g., 64K);
```

For each newly ACKed segment:

```
if (cwnd < ssthresh)
    /* MI: slow start*/
    cwnd = cwnd + 1;
```



Startup Behavior with Slow-start



See [Jac89]

AIMD: Congestion Avoidance

- Algorithm: AIMD
  - increases window by 1 per round-trip time (how?)
  - cuts window size
    - to half when detecting congestion by 3DUP
    - to 1 if timeout
    - if already timeout, doubles timeout
- Goal: Maintains equilibrium and reacts around equilibrium

TCP/Reno Full Alg

Initially:

```
    cwnd = 1;
    ssthresh = infinite (e.g., 64K);
```

For each newly ACKed segment:

```
    if (cwnd < ssthresh)        // slow start: MI
        cwnd = cwnd + 1;
    else
                                // congestion avoidance; AI
        cwnd += 1/cwnd;
```

Triple-duplicate ACKs:

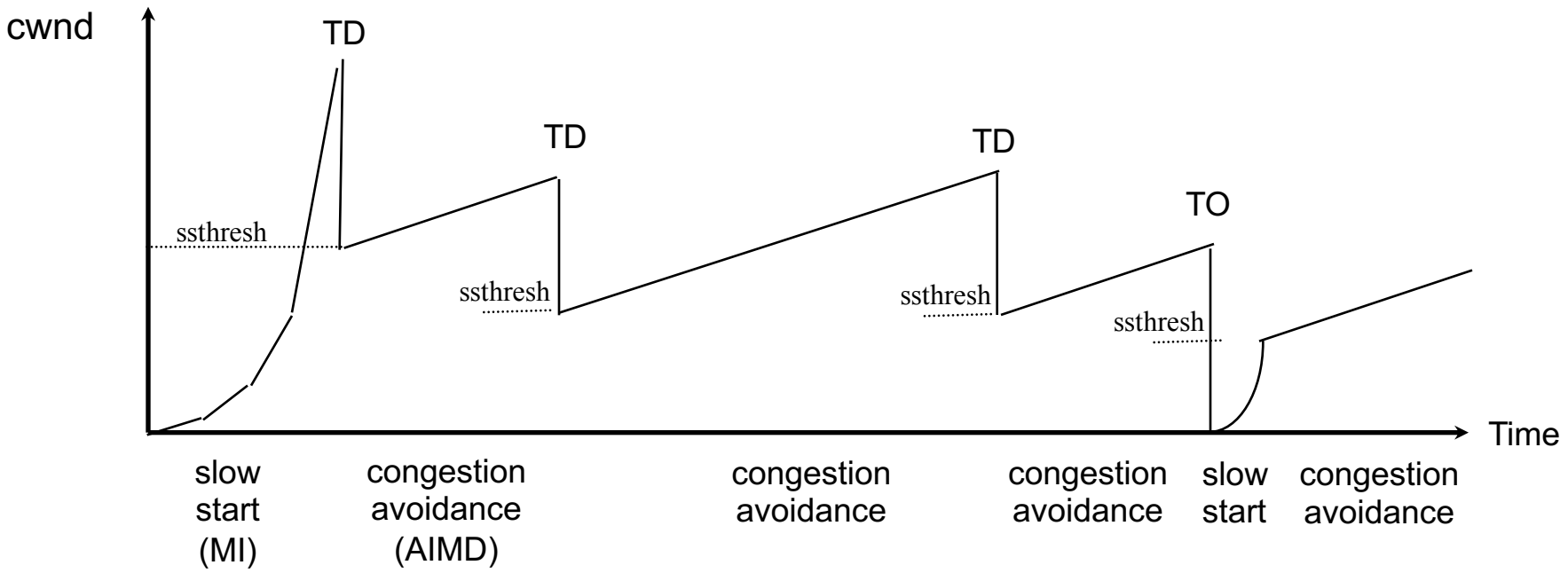
```
                                // MD
    cwnd = ssthresh = cwnd/2;
```

Timeout:

```
    ssthresh = cwnd/2;          // reset
    cwnd = 1;
```

(if already timed out, double timeout value; this is called exponential backoff)

TCP/Reno: Big Picture



TD: Triple duplicate acknowledgements
TO: Timeout

Outline

- Admin and recap
- Transport congestion control
  - what is congestion (cost of congestion)
  - basic congestion control alg.
  - TCP/Reno congestion control
    - design
    - *analysis*

Objective

- To understand
  - the throughput of TCP/Reno as a function of RTT (RTT), loss rate (p) and packet size
  - the underlying queue dynamics

- We will analyze TCP/Reno under two different setups

TCP/Reno Throughput Analysis

- Given mean packet loss rate p, mean round-trip time RTT, packet size S
- Consider only the congestion avoidance mode (long flows such as large files)
- Assume no timeout
- Assume mean window size is W_m segments, each with S bytes sent in one RTT:

$$\text{Throughput} = \frac{W_m * S}{RTT} \text{ bytes/sec}$$

Outline

- Admin and recap
- Transport congestion control
  - what is congestion (cost of congestion)
  - basic congestion control alg.
  - TCP/Reno congestion control
    - design
    - analysis
      - small fish in a big pond
        - loss rate given from the environment

TCP/Reno Throughput Modeling (Fixed, Given Loss Rate)

$$\Delta W = \begin{cases} \frac{1}{W} & \text{if the packet is not lost} \\ -\frac{W}{2} & \text{if packet is lost} \end{cases}$$

$$mean\ of\ \Delta W = (1-p)\frac{1}{W} + p(-\frac{W}{2}) = 0$$

$$\Rightarrow \quad mean\ of\ W = \sqrt{\frac{2(1-p)}{p}} \approx \frac{1.4}{\sqrt{p}}, \text{ when } p \text{ is small}$$

$$\Rightarrow \quad \boxed{throughput \approx \frac{1.4S}{RTT\sqrt{p}}, \text{ when } p \text{ is small}}$$

This is called the TCP throughput sqrt of loss rate law.

Exercise: Application of Analysis

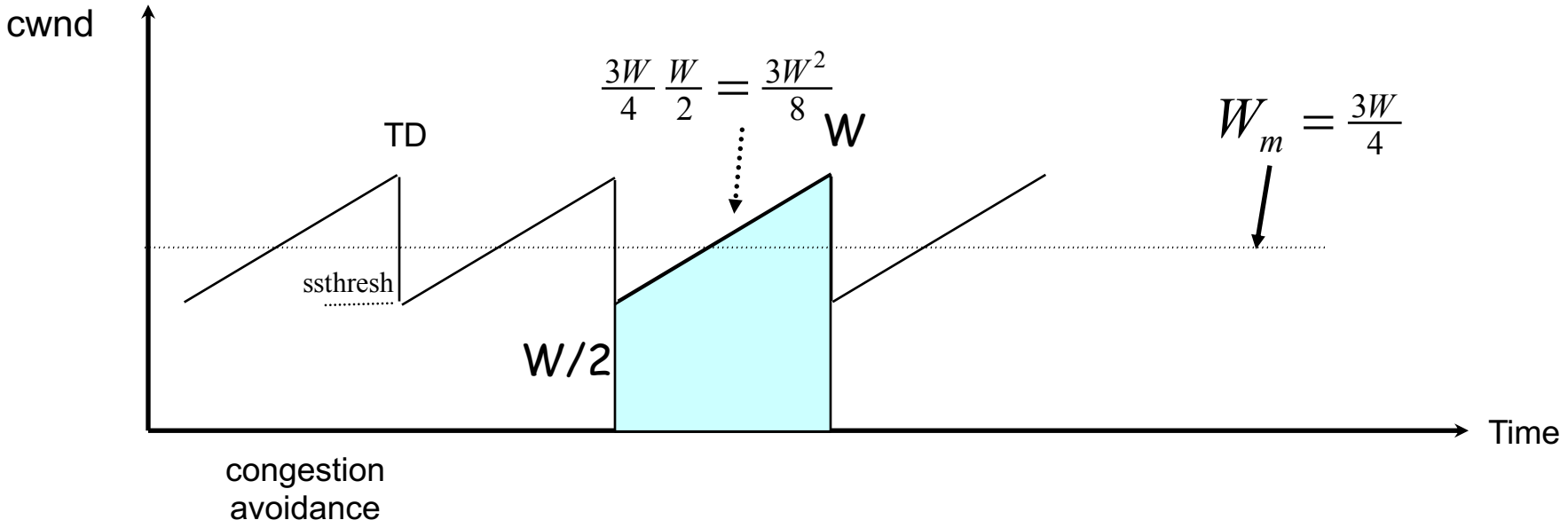
- State of art network link can reach 100 Gbps. Assume packet size 1250 bytes, RTT 100 ms, what is the highest packet loss rate to still reach 100 Gbps?

tcp-reno-tput.xlsx

Outline

- Admin and recap
- Transport congestion control
  - what is congestion (cost of congestion)
  - basic congestion control alg.
  - TCP/Reno congestion control
    - design
    - analysis
      - small fish in a big pond
      - big fish in small pond
        - growth causes losses

TCP/Reno Throughput Modeling: Relating W with Loss Rate p



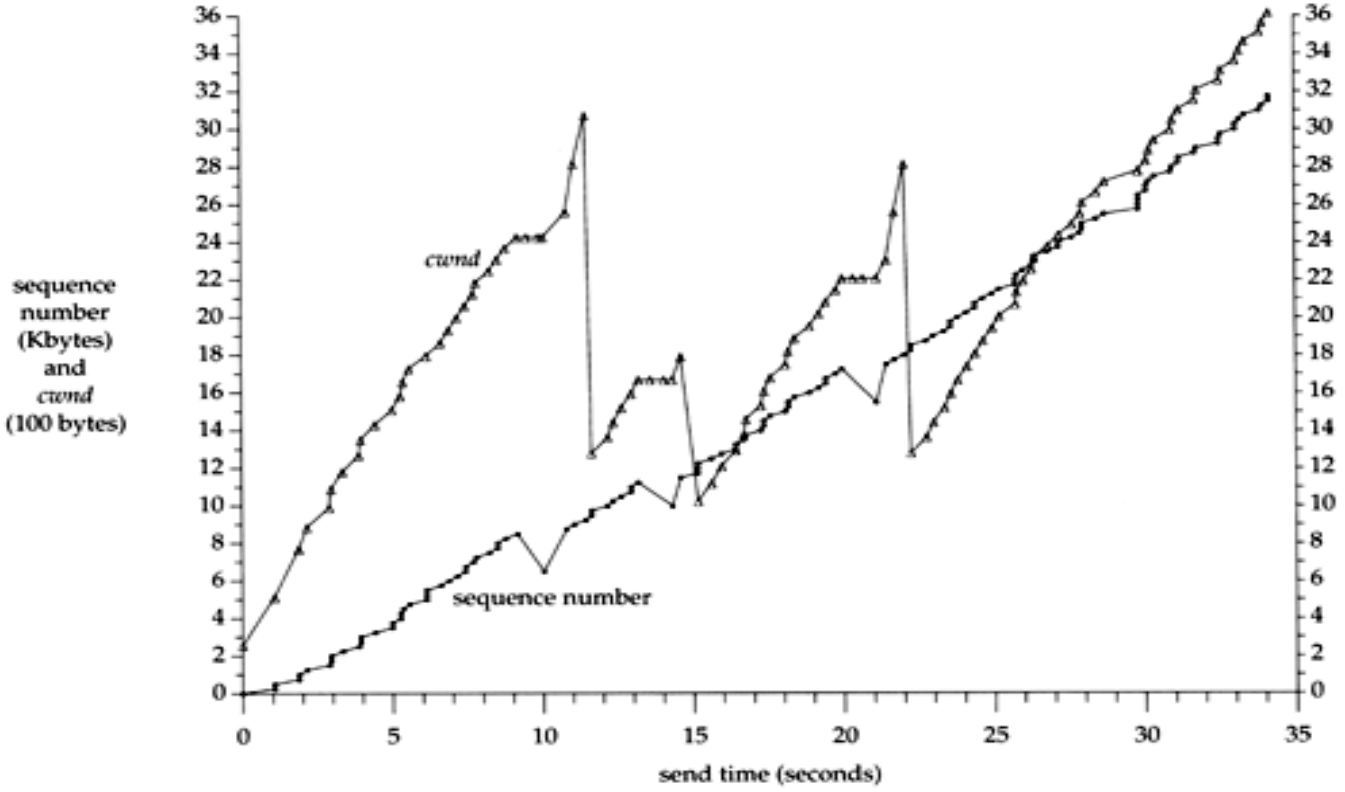
Total packets sent per cycle = (W/2 + W)/2 * W/2 = $3W^2/8$

Assume one loss per cycle => p = $1/(3W^2/8) = 8/(3W^2)$

$$\Rightarrow \quad W = \frac{\sqrt{8/3}}{\sqrt{p}} = \frac{1.6}{\sqrt{p}}$$

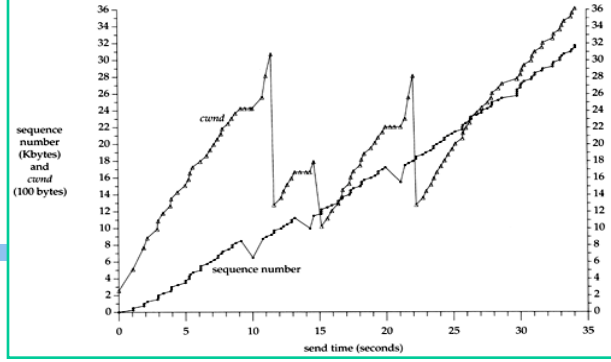
$$\Rightarrow throughput = \frac{S}{RTT}\ \frac{3}{4}\ \frac{1.6}{\sqrt{p}} = \boxed{\frac{1.2S}{RTT\sqrt{p}}}$$

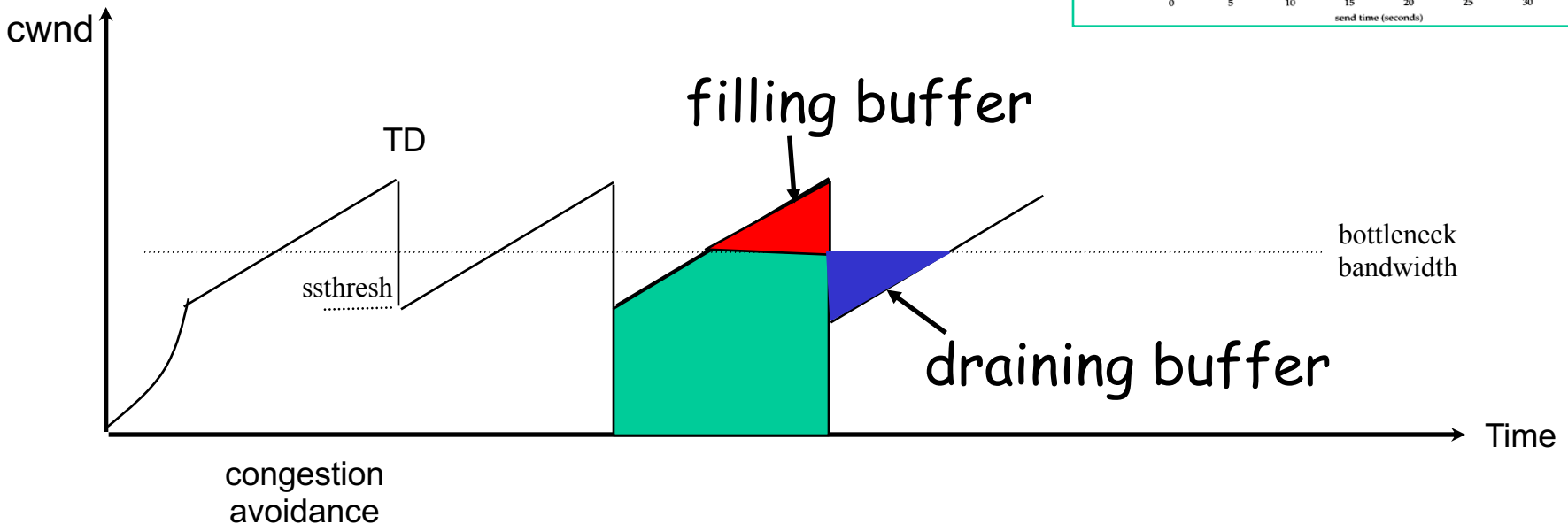
A Puzzle: cwnd and Rate of a TCP Session



Question: although cwnd fluctuates widely (i.e., cut to half), why can the sending rate stay relatively smooth?

TCP/Reno Queueing Dynamics





If the buffer at the bottleneck is large enough, the buffer is never empty (not idle), during the cut-to-half to "grow-back" process.

Exercise: How big should the buffer be to achieve full utilization?

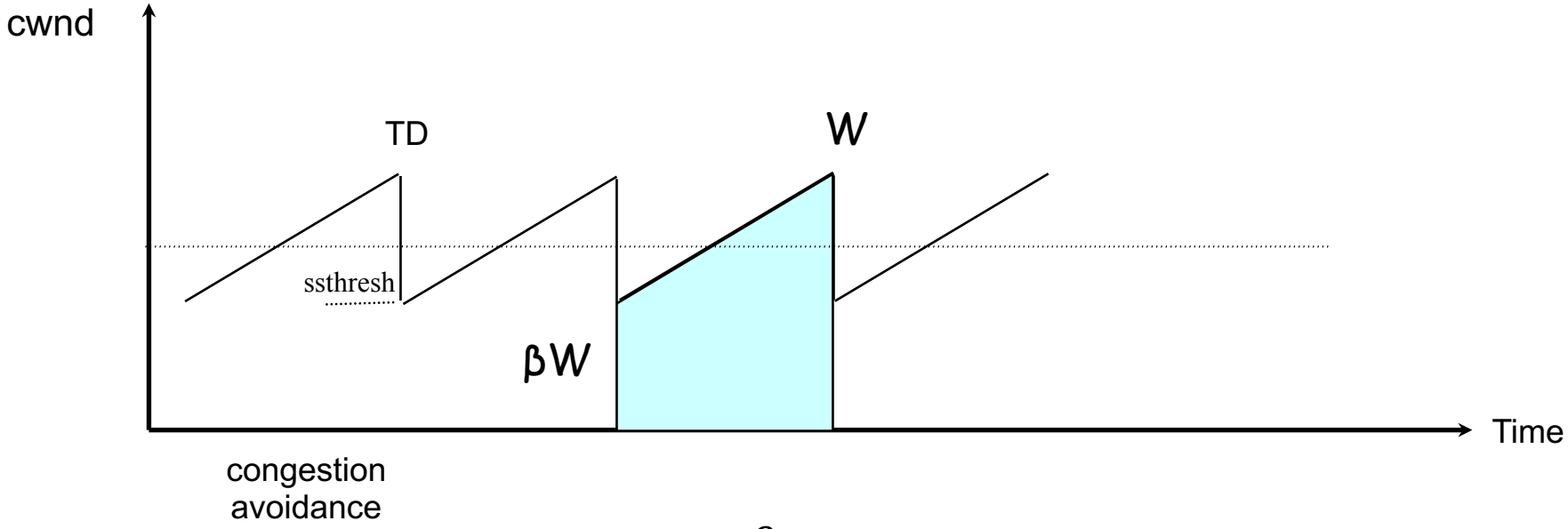
Extreme: Zero Buffer

- Assume
  - BW: 10 G
  - RTT: 100 ms
  - Packet: 1250 bytes
  - BDP (full window size): 100,000 packets
- A loss can cut window size from 100,000 to 50,000 packets
- To fully grow back
  - Need 50,000 RTTs => 5000 seconds, 1.4 hours
- Q: What is the link utilization in one cycle?

Design

- Assume a generic AIMD alg:
  - reduce to β W after each loss event
- Q: What value β gives higher utilization (assume small/zero buffer)?
- Q: Assume picking a high value β, how to make the alg TCP friendly?

Generic AIMD and TCP Friendliness



Total packets sent per cycle = $\frac{\beta W+W}{2}\frac{(1-\beta)W}{\alpha} = \frac{(1-\beta)(1+\beta)}{2\alpha}\mathrm{w}^2$

Assume one loss per cycle $\mathrm{p} = \frac{2\alpha}{(1-\beta)(1+\beta)\mathrm{w}^2}$ $\mathrm{w} = \sqrt{\frac{2\alpha}{(1-\beta)(1+\beta)\mathrm{p}}}$

$$\mathrm{tput} = \frac{W_m S}{RTT} = \frac{S}{RTT}\frac{(1+\beta)W}{2} = \frac{S}{RTT}\sqrt{\frac{\alpha(1+\beta)}{2(1-\beta)\mathrm{p}}}$$

TCP friendly => $\alpha = 3\frac{1-\beta}{1+\beta}$

Outline

- Admin and recap
- Transport congestion control
  - what is congestion (cost of congestion)
  - basic congestion control alg.
  - TCP/Reno congestion control
  - TCP Cubic

TCP Cubic

- Designed in 2008
- Default for Linux
- Most sockets in MAC appear to use cubic as well
  - sw_vers
  - sysctl –a

TCP Cubic Goals

- Improve TCP efficiency over fast, long-distance links

  Smaller reduction, longer stay at BDP, faster than linear increase---cubic function

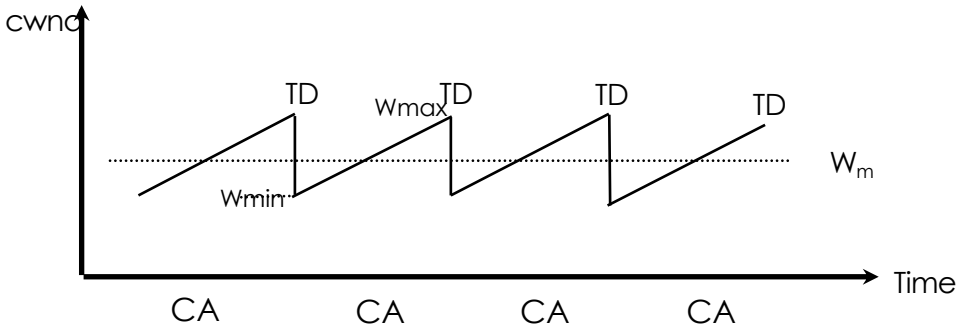
- TCP friendliness

  Follows TCP if TCP gives higher rate

- Fairness of flows w/ different RTTs

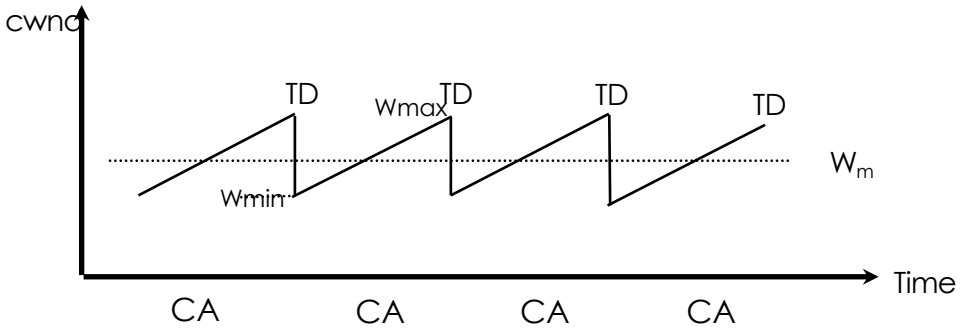
  Window growth depends on real-time (from congestion-epoch through synchronized losses)

TCP BIC Algorithm



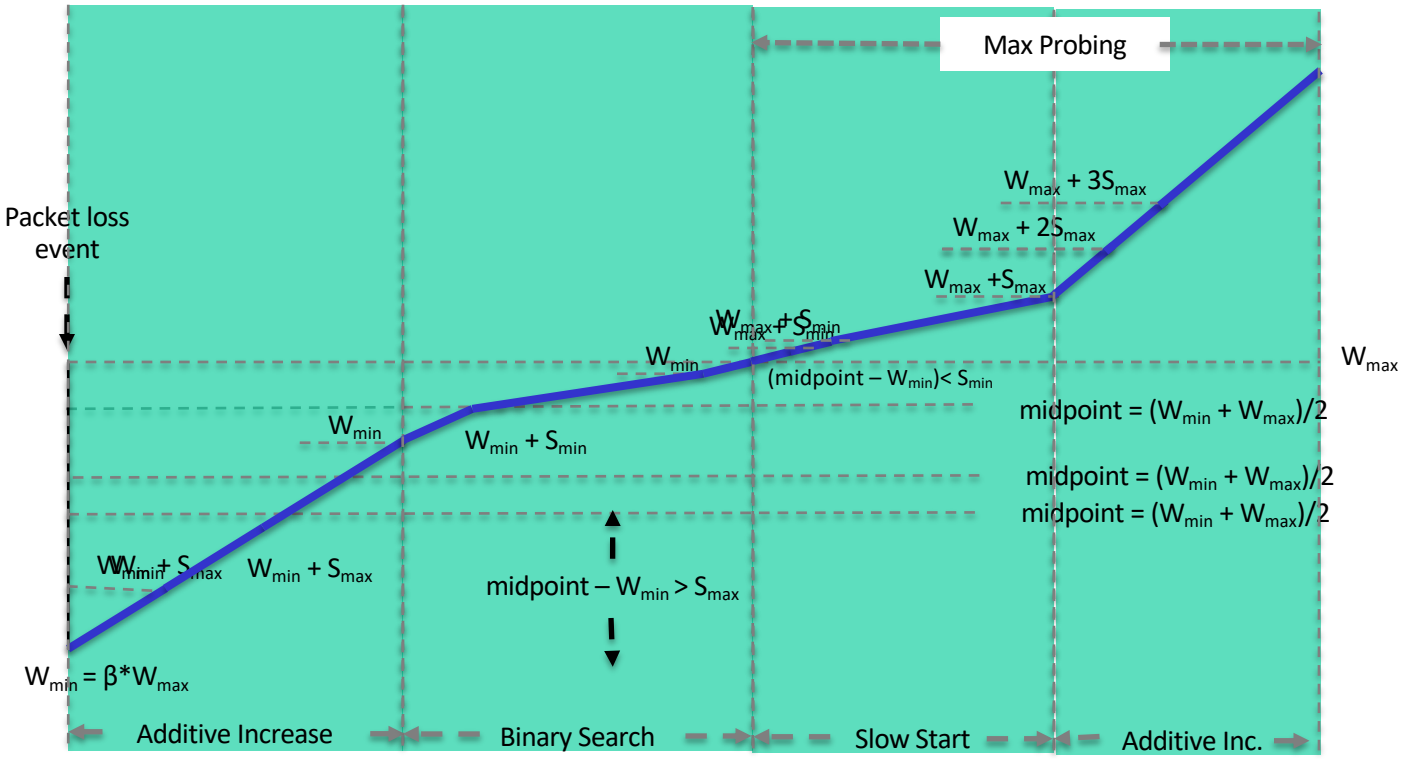
- Setting
  - W_{max} = cwnd size before reduction
    - Too big
  - W_{min} = β*W_{max} – just after reduction, where β is multiplicative decrease factor
    - Small
- Basic idea
  - binary search between W_{max} and W_{min}

TCP BIC Algorithm: Issues



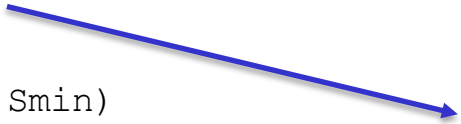
- Pure binary search (jump from W_{min} to (W_{max} and W_{min})/2) may be too aggressive
  - Use a large step size Smax
- What if you grow above W_{max}?
  - Use binary growth (slow start) to probe more

TCP BIC Algorithm

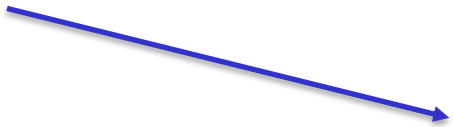


TCP BIC Algorithm

```
while (cwnd < Wmax) {
     if ( (midpoint - Wmin) > Smax )
          cwnd = cwnd + Smax
      else
          if ((midpoint - Wmin) < Smin)
               cwnd = Wmax
          else
               cwnd = midpoint
     if (no packet loss)
          Wmin = cwnd
     else
          Wmin = β*cwnd
          Wmax = cwnd
     midpoint = (Wmax + Wmin)/2
 }
```



Additive Increase



Binary Search

TCP BIC Algorithm: Probe

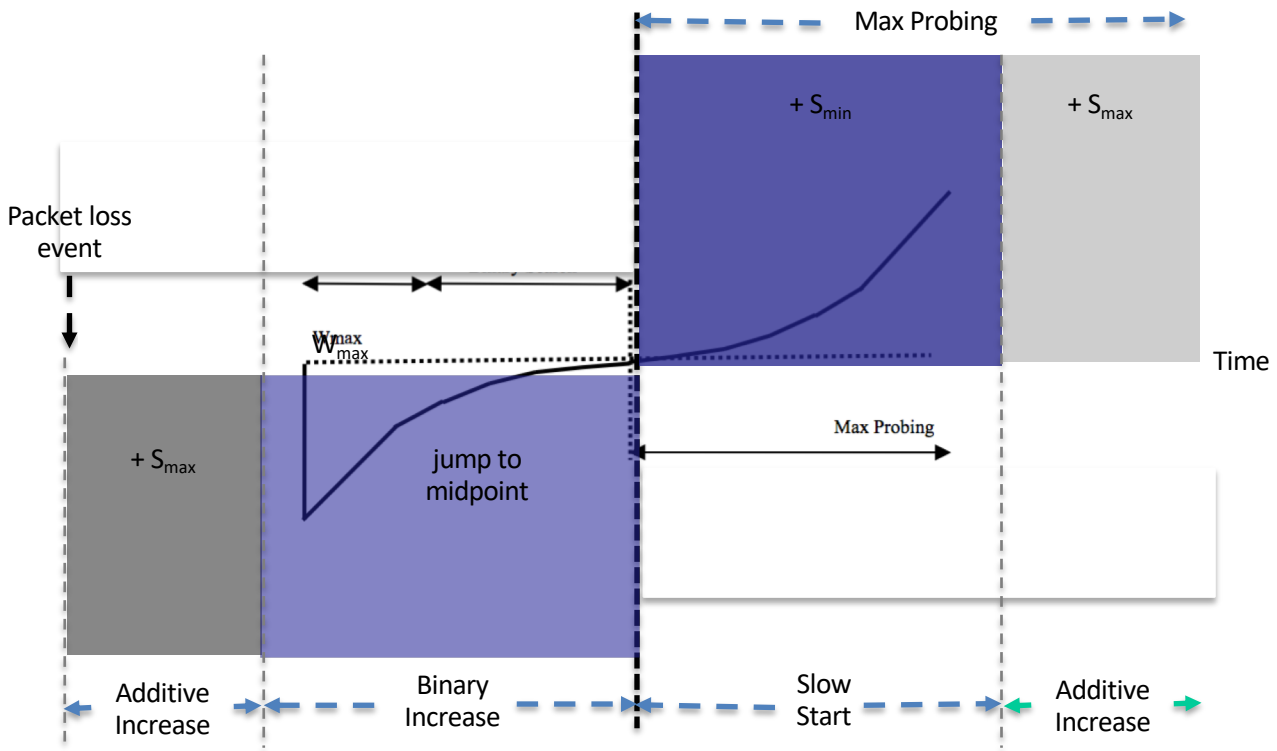
```
while (cwnd >= W_max){
  if (cwnd < W_max + S_max)
       cwnd = cwnd + S_min
  else
       cwnd = cwnd + S_max

  if (packet loss)
       W_min = β*cwnd
       W_max = cwnd
}
```

Slow growth

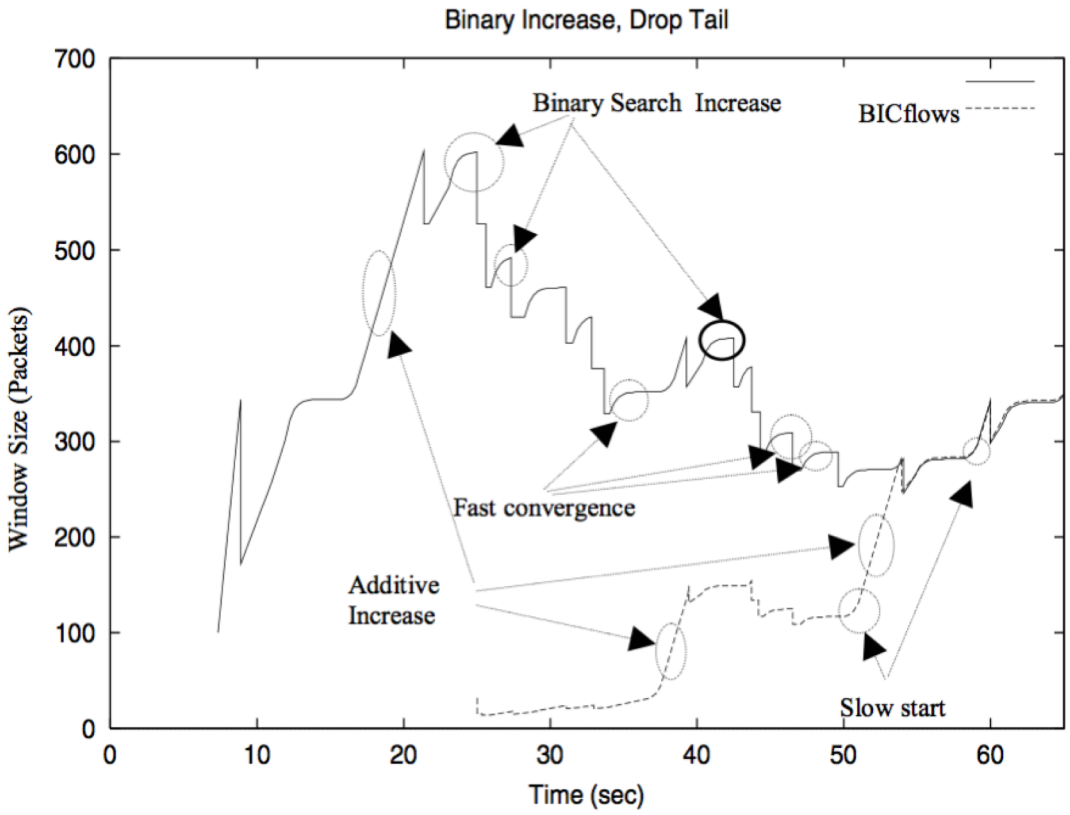
Fast growth

Max Probing

TCP BIC - Summary



TCP BIC in Action



TCP BIC Analysis

- Advantages
  - *Faster convergence at large gap*
  - *Slower growth at convergence to avoid timeout*

- *Issues*
  - Still depend on RTT
  - Complex growth function

More details: http://www.land.ufrj.br/~classes/coppe-redes-2007/projeto/BIC-TCP-infocom-04.pdf

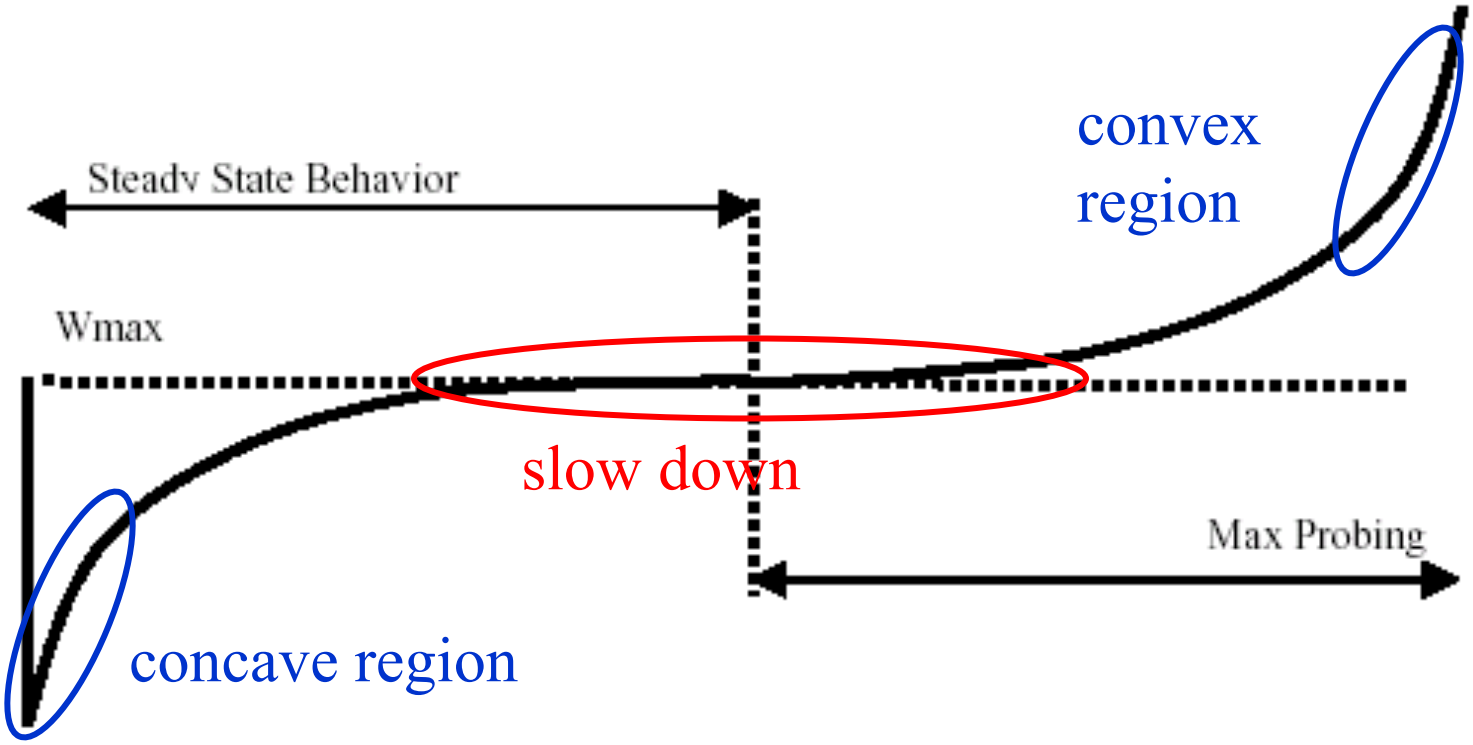
Cubic High-Level Structure

- If (received ACK && state == cong avoid)
  - Compute $W_{cubic}(t+RTT)$.
  - If cwnd < W_{TCP}
    - Cubic in TCP mode
  - If cwnd < Wmax
    - Cubic in concave region
  - If cwnd > Wmax
    - Cubic in convex region

The Cubic function

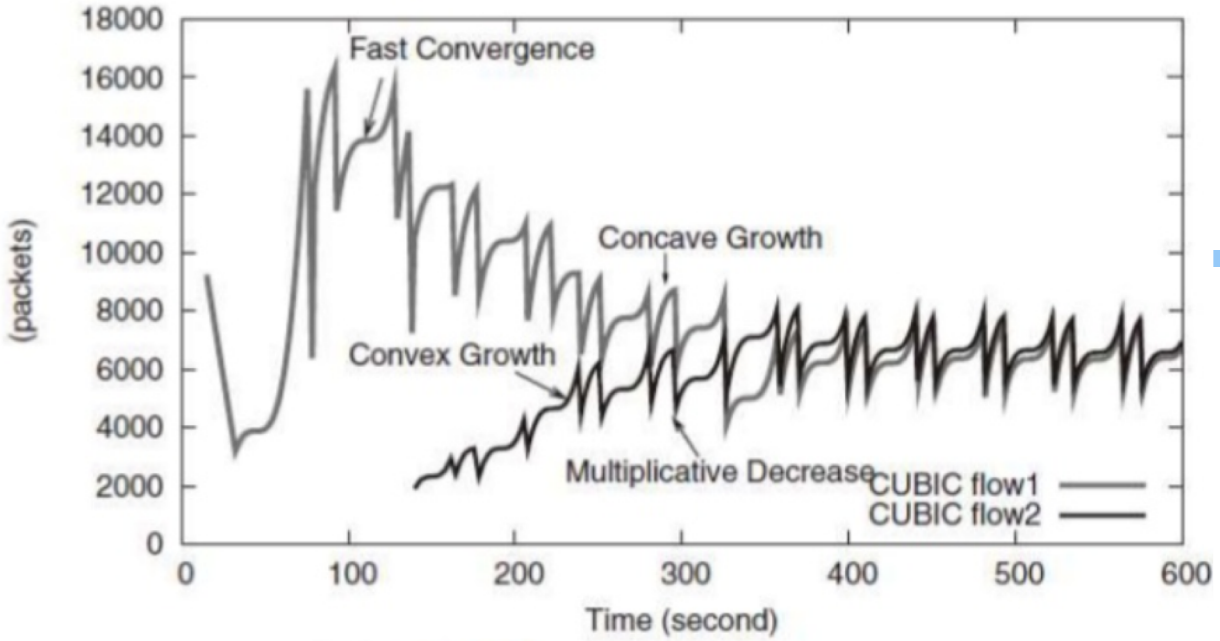
$$\beta' = 1 - \beta$$

$$W_{tcp(t)} = Wmax\,\beta' + 3\,\frac{1-\beta'}{1+\beta'}\,\frac{t}{RTT}$$

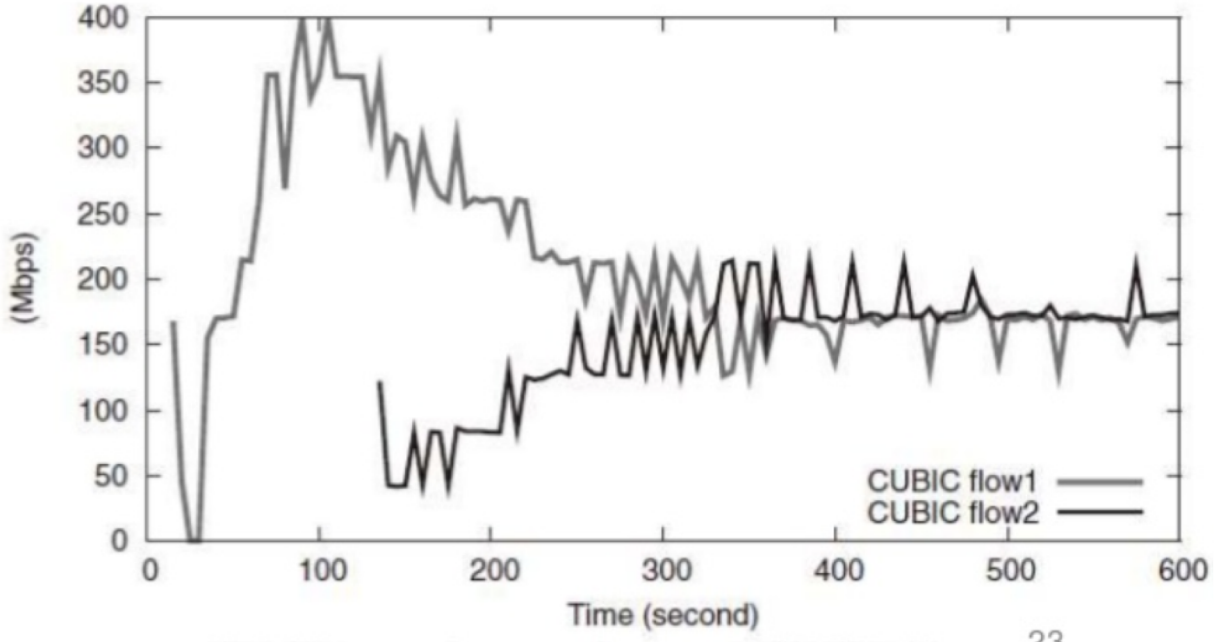


$$W_{cubic} = C(t-K)^3 + W_{\max} \qquad K = \sqrt[3]{W_{\max}\beta/C}$$

where ***C*** is a scaling factor, ***t*** is the elapsed time from the last window reduction, and ***β*** is a constant multiplication decrease factor



(a) CUBIC window curves.

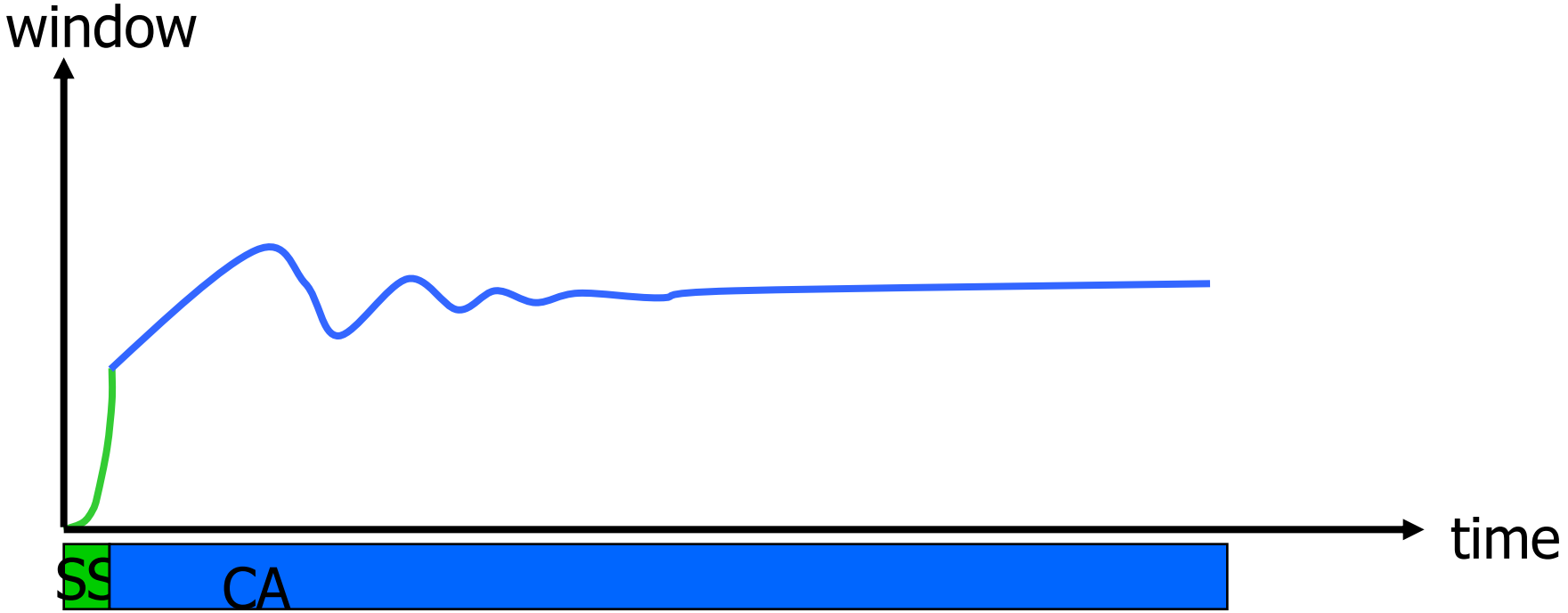


(b) Throughput of two CUBIC flows.[23]

Outline

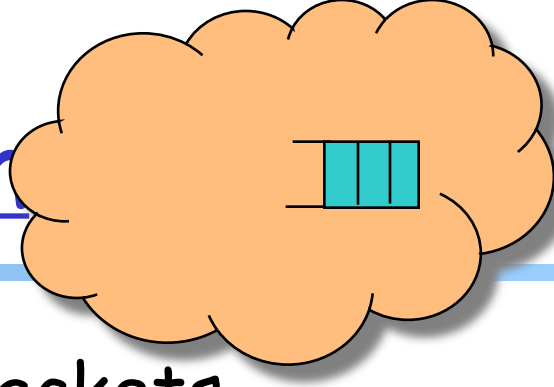
- Admin and recap
- Transport congestion control
  - what is congestion (cost of congestion)
  - basic congestion control alg.
  - TCP/Reno congestion control
  - TCP Cubic
  - TCP/Vegas

TCP/Vegas (Brakmo & Peterson 1994)

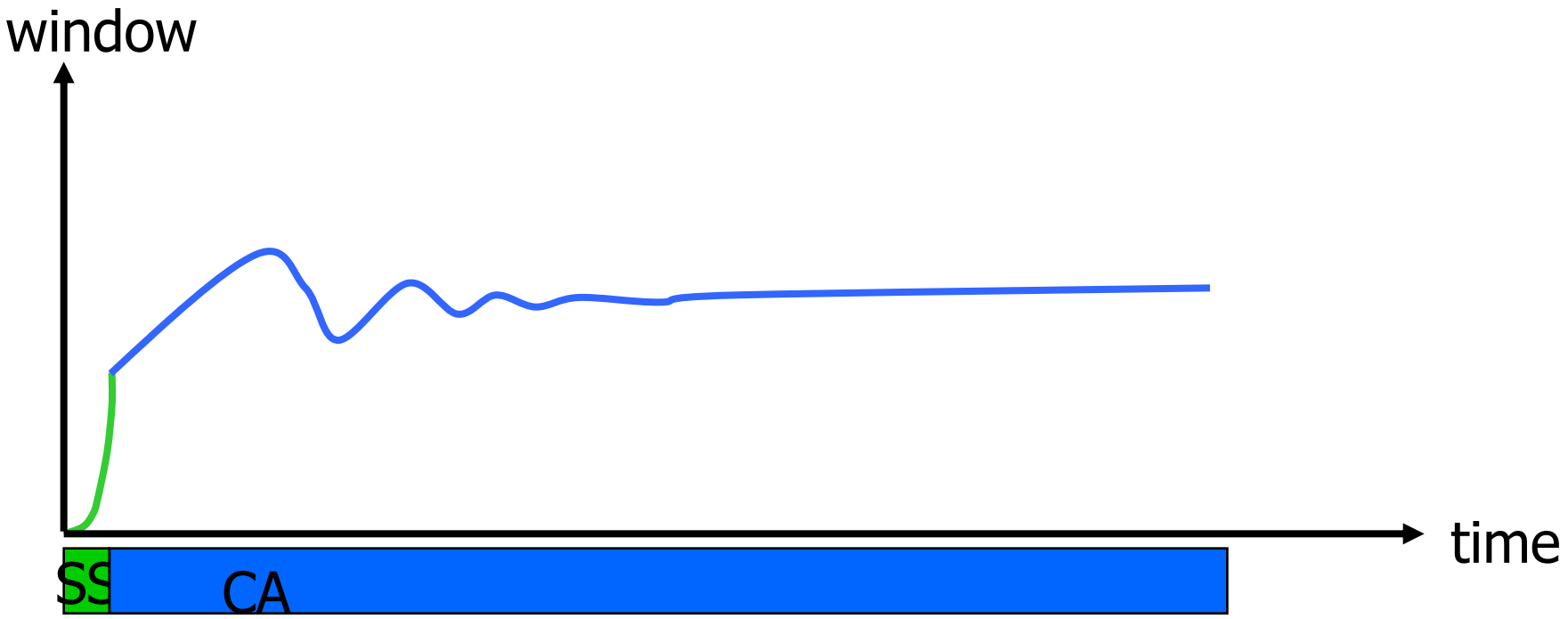


- Idea: try to detect congestion by delay before loss
- Objective: not to overflow the buffer; instead, try to maintain a *constant* number of packets in the bottleneck queue

TCP/Vegas: Key Question

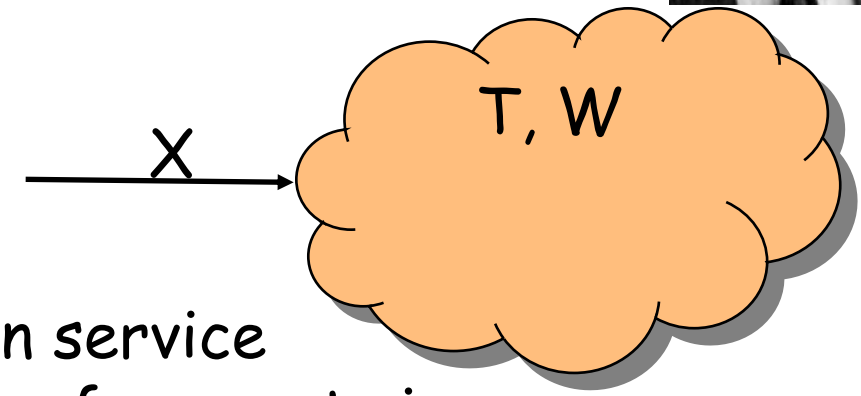


- How to estimate the number of packets queued in the bottleneck queue?



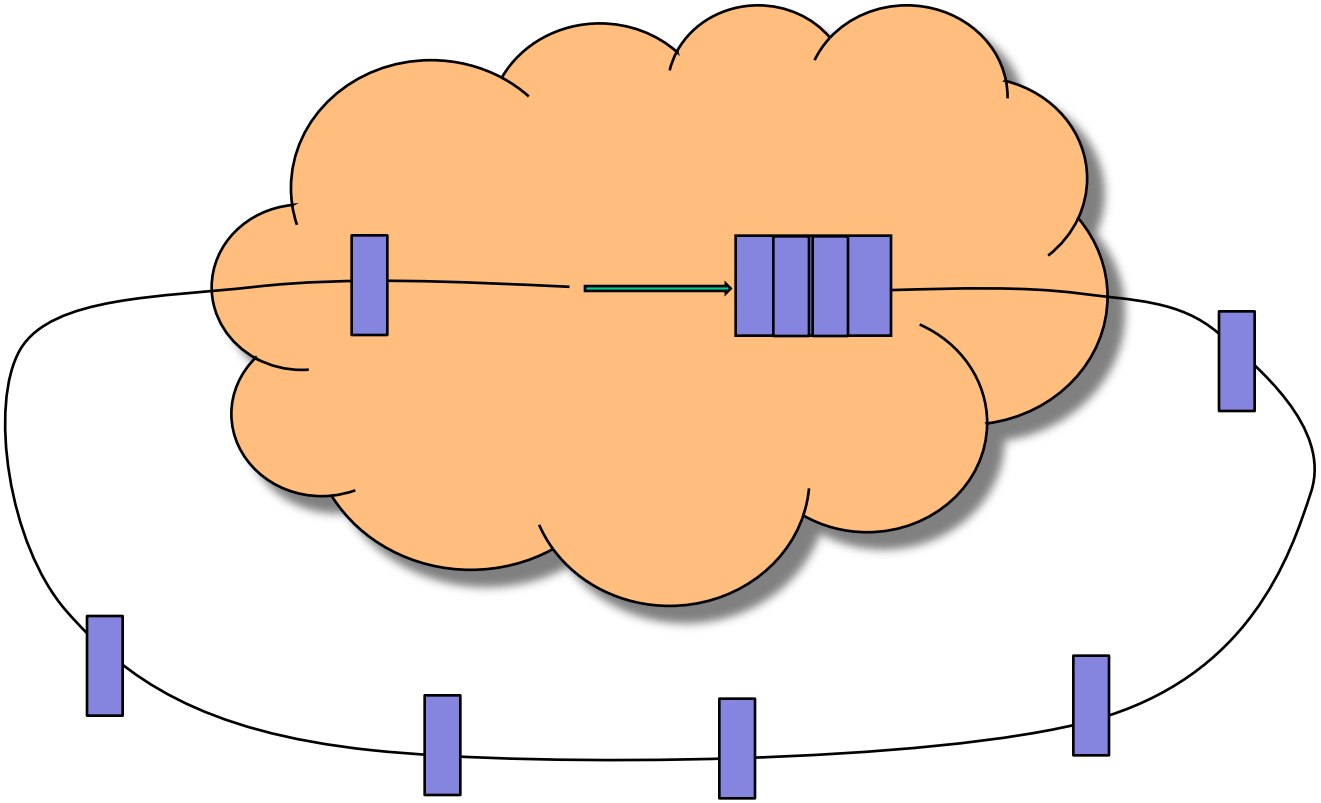
Recall: Little's Law

- For any system with no or (low) loss.
- Assume
  - mean arrival rate X, mean service time T, and mean number of requests in the system W
- Then relationship between W, X, and T:

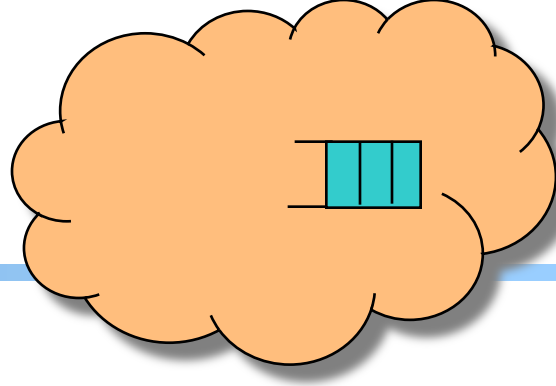


$$W = XT$$

Estimating Number of Packets in the Queue



TCP/Vegas CA algorithm



$$T = T_{prop} + T_{queueing}$$

Applying Little's Law:

$$x_{vegas}\, T = x_{vegas}\, T_{prop} + x_{vegas}\, T_{queueing},$$

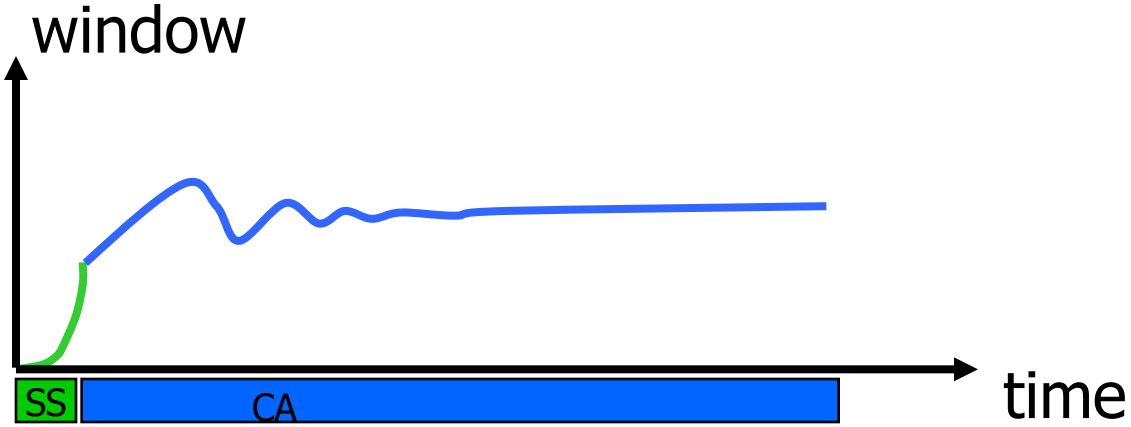
where $x_{vegas} = W / T$ is the sending rate

Then number of packets in the queue is

$$x_{vegas}\, T_{queueing} = x_{vegas}\, T - x_{vegas}\, T_{prop}$$
$$= W - W/T\, T_{prop}$$

TCP/Vegas CA algorithm

maintain a *constant* number of packets in the bottleneck buffer




```
for every RTT
{    if W - W/RTT RTT_min < α  then W ++
     if W - W/RTT RTT_min > α  then W --
}
for every loss
        W := W/2
```

queue size

Discussions

- If two flows, one TCP Vegas and one TCP reno run together, how may bandwidth partitioned among them?
- Issues that limit Vegas deployment?