

TCP Performance

- Effective over a wide range of capacities
- A lot of operational experience

$$\lambda_{tcp} = F(p, RTT)$$

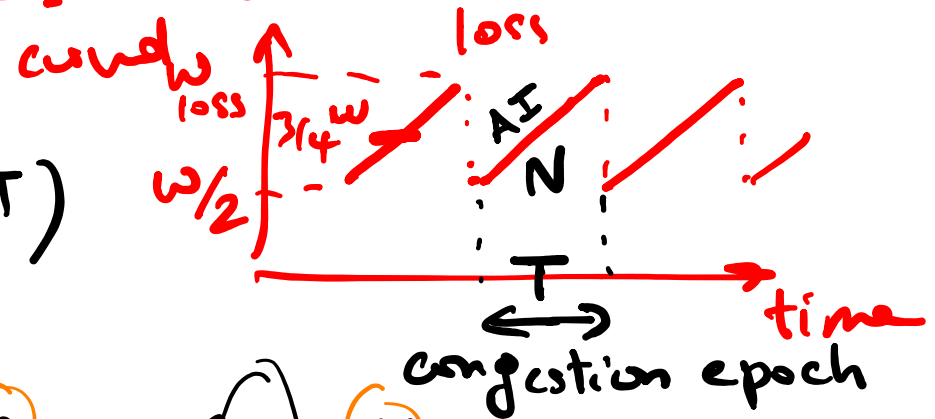
$\frac{w}{RTT}$ sending rate (throughput) \uparrow prob. of segment lost

- Periodic loss (macroscopic) model shows that throughput is inversely proportional to
 - square root of loss probability p
 - RTT
 - average sending rate = $\text{sqrt}(1.5/p) / RTT$

- Reno
- Steady-state, static pRRT
- low/moderate congestion \Rightarrow AIMD

$$\lambda = ? = F(P, R+1)$$

- periodic loss model
(losing one segment every T)



$$\lambda = \frac{N}{L}$$

$$Z = \frac{1}{2} \left(\frac{3}{2} + 1 + \frac{3}{2} + 2 + \dots + \frac{3}{2} + (n+1) \right)$$

$$\frac{1+2+3+\dots+n}{2} = \frac{n(n+1)}{2}$$

$$= \frac{3}{2} + \frac{3}{8} + \frac{3}{2} = \frac{9}{2}$$

$$\underline{\frac{3}{4}\omega^2 + \frac{3}{4}\omega} = \frac{3}{4}\omega\left(\frac{\omega}{2} + 1\right) \quad G_2(p)$$

$$T = \left(\frac{2}{2} + 1 \right) R_{TT} \Rightarrow \lambda = \frac{2}{12} = \frac{\frac{3}{4} \omega \left(\frac{2}{2} + 1 \right)}{\left(\frac{2}{2} + 1 \right) R_{TT}} = \frac{3/4 \omega}{R_{TT}}$$

$$N = \frac{3}{2}\omega^2 + \frac{1}{2}\omega \approx \frac{3}{2}\omega^2 \quad (\omega \gg 1)$$

$$N = \frac{3}{8} \omega^2 + \frac{3}{4} \omega \approx \frac{3}{8} \omega^2 \quad (\omega \gg 1)$$

$$p = \frac{1}{2} = \left(\frac{3}{8} \omega^2 \right) \Rightarrow \omega^2 = \frac{8/3}{p} = \frac{8}{3p}$$

$\omega = \frac{\sqrt{8/3}}{\sqrt{P}}$
 $\lambda = \frac{3/4 \sqrt{8/3}}{\sqrt{P} \cdot RTT}$
 $\lambda = \frac{\sqrt{3/2}}{\sqrt{P} \cdot RTT}$

Matta © BUCS - Transport 1-86

TCP Futures: TCP over "long, fat pipes"

$$T = \left(\frac{W}{2} + 1\right) RTT$$

$B \times D \uparrow$

- Example: 10,000-bit segments, 100ms RTT, want 10 Gbps throughput

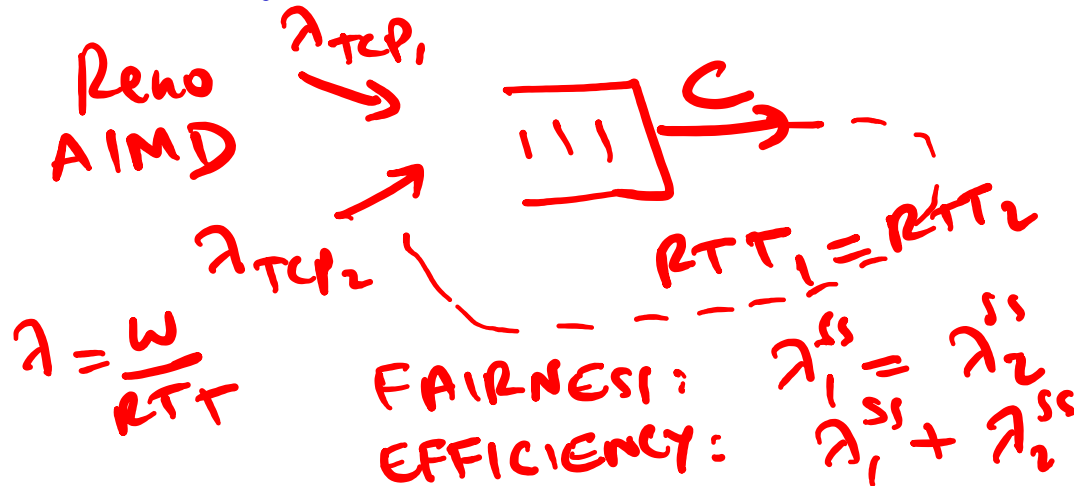
$$W_{loss} \geq W_{ideal} = B \times D = \frac{10 \text{ G}^{10^9}}{10,000 = 10^4} \times 100 \text{ ms} = 10^5 = 100,000 \text{ segments}$$

$$\Rightarrow T = \left(\frac{W_{loss}}{2} + 1\right) RTT \geq \left(\frac{100,000}{2} + 1\right) * 100 \text{ ms} = 5000 \text{ seconds} > 1 \text{ hour}$$

TCP Futures: TCP over “long, fat pipes”

- Example: 10,000-bit segments, 100ms RTT, want 10 Gbps throughput
- Requires window size $W = 100,000$ in-flight segments
- Throughput in terms of loss rate:
$$10 \text{ Gbps} = \lambda = \frac{1.22 \text{ MSS} \leftarrow 10,000}{RTT \sqrt{p} \leftarrow 100\text{ms}} \rightarrow \checkmark$$
- $p = 1.5 \times 10^{-10}$ *Wow*
- New versions of TCP for high-speed

Why is TCP fair?



AI: $w \leftarrow w + 1$ every RTT

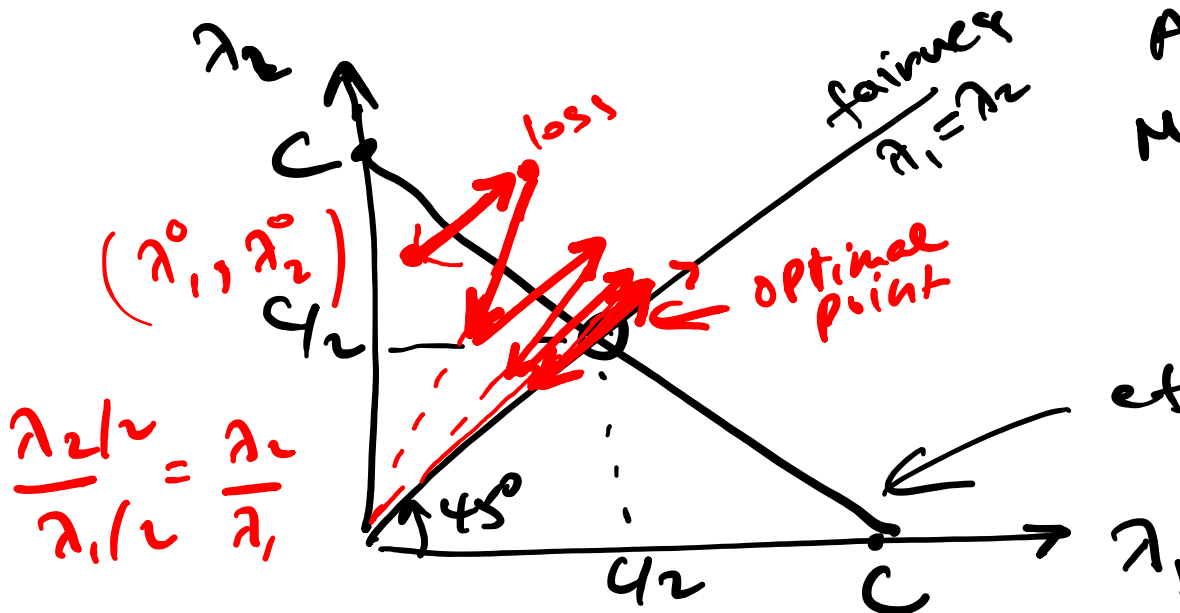
MD: $w \leftarrow \frac{w}{2}$ upon loss

OPTIMAL
OPERATING
POINT

AI: $\lambda \leftarrow \lambda + \frac{1}{RTT}$

MD: $\lambda \leftarrow \lambda/2$

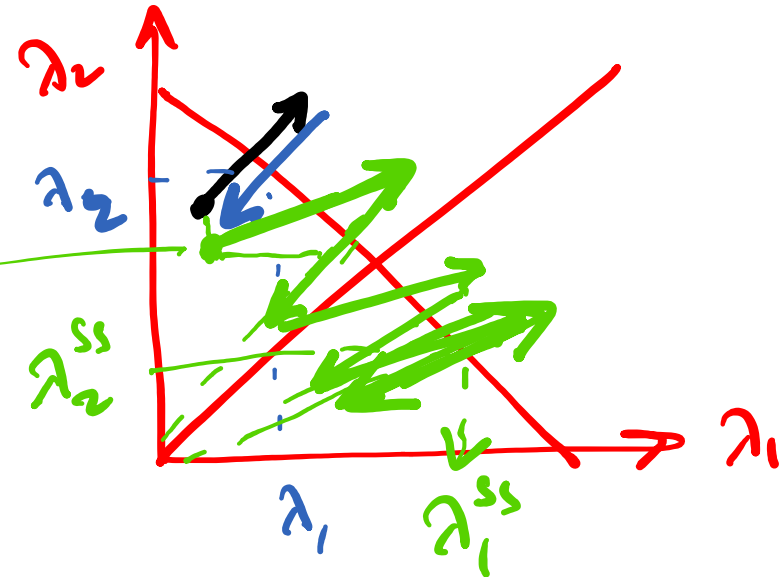
synchronized
update model



Why is TCP fair?

AIAD

$$\lambda_1 < \lambda_2$$



AIMD

$$RTT_1 < RTT_2$$

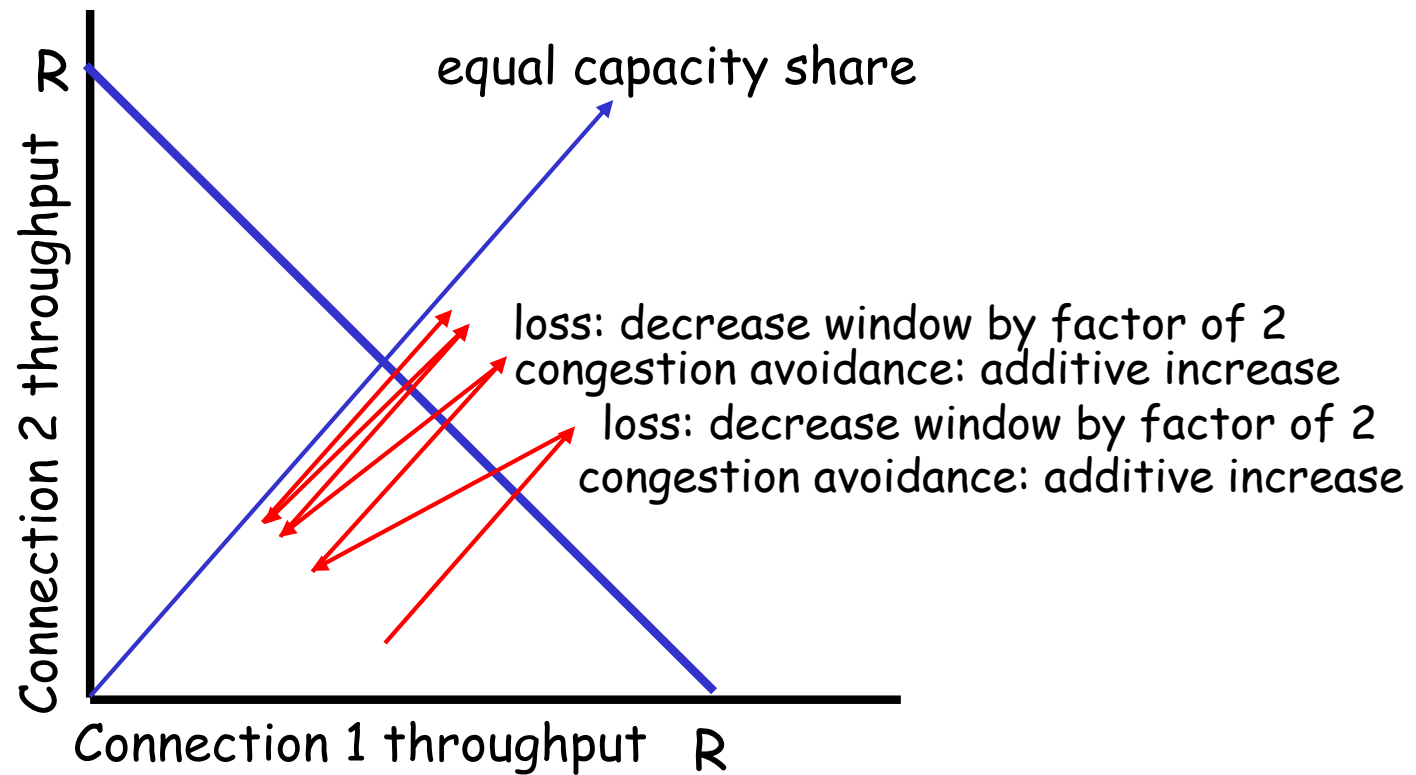
$$AI: \lambda \leftarrow \lambda + \left(\frac{1}{RTT} \right) \Rightarrow \lambda_1^{ss} > \lambda_2^{ss}$$

$$\lambda \propto \frac{1}{\sqrt{p} \cdot RTT}$$

Why is TCP fair?

Two competing sessions (with same RTT):

- Additive increase gives slope of 1, as throughput increases
- multiplicative decrease decreases throughput proportionally



Delay modeling

Q: How long does it take to receive an object from a Web server after sending a request?

Ignoring congestion, delay is influenced by:

- ❑ TCP connection establishment
- ❑ data transmission delay
- ❑ slow start

Notation, assumptions:

- ❑ Assume one link between client and server of rate C
- ❑ S : MSS (bits)
- ❑ O : object size (bits)
- ❑ no retransmissions (no loss, no corruption)

TCP Delay Modeling: Slow Start

Delay components:

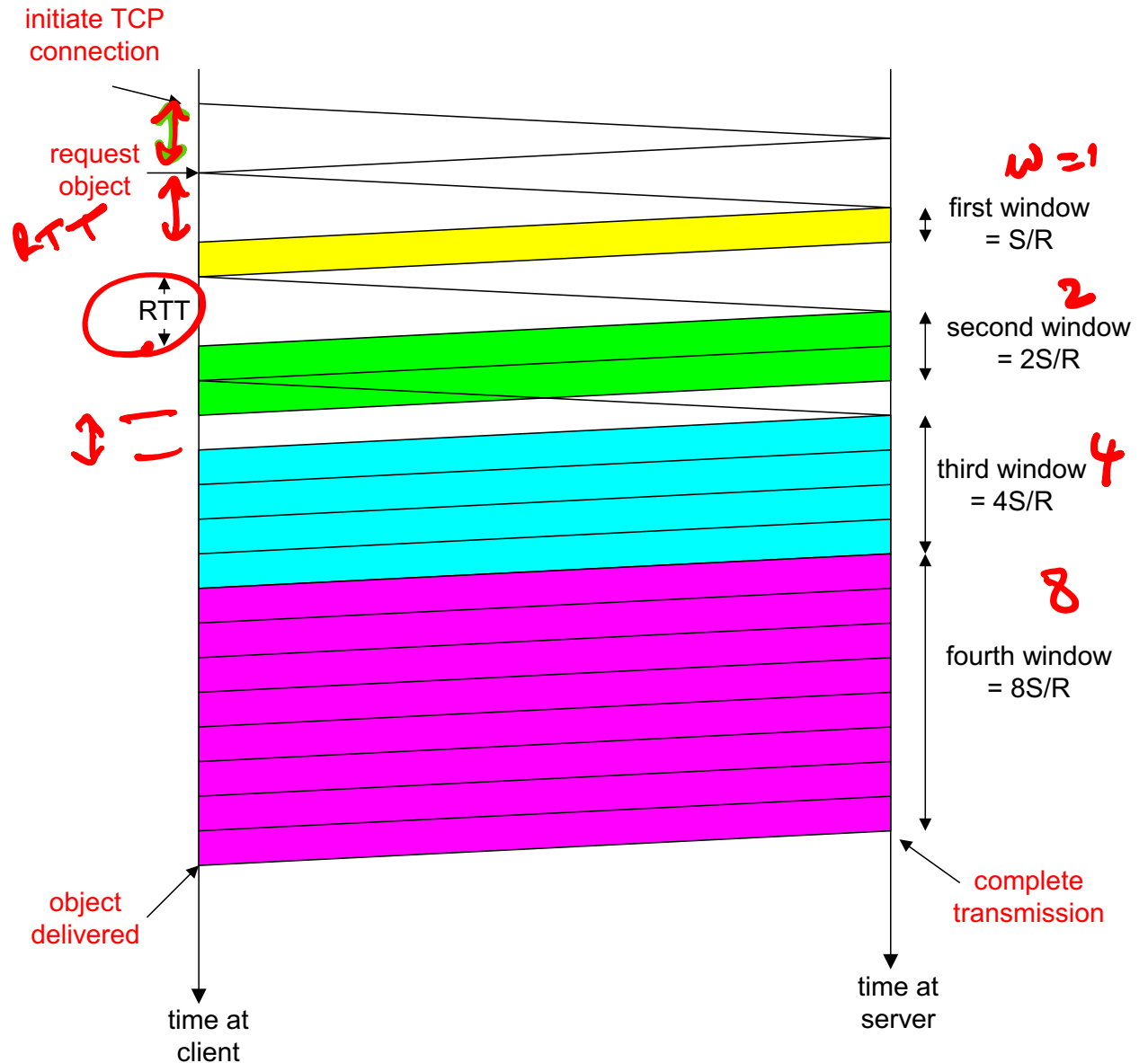
- 2 RTT for connection estab and request
- O/C to transmit object

$$[RTT - 2 \frac{S}{C}]$$

Example:

- $O/S = 15$ segments

Server idles 2 times
due to slow start



Chapter 3: Summary

- ❑ principles behind transport layer services:
 - ❑ multiplexing, demultiplexing
 - ❑ reliable data transfer
 - ❑ flow control
 - ❑ congestion control
- ❑ instantiation and implementation in the Internet
 - ❑ UDP
 - ❑ TCP

Next:

- ❑ leaving the network “edge” (application, transport layers)
- ❑ into the network “core”