Transport Protocols for Networked Games

TCP or UDP?

Why use TCP?

- TCP provides reliable, in-order delivery
- TCP goes through most firewalls, UDP does not
- TCP manages connection for us

Why not to use TCP?

- TCP incur higher latency
- Don't always need reliability and in-order delivery
- High header overhead

position =
$$10 \longrightarrow X$$

position = $13 \longrightarrow X$
position = $15 \longrightarrow X$

Updated position not delivered to application until (outdated) lost packet is received



Gesture from someone far away need not be received reliably.

A study on ShenZhou Online shows that 46% of the bandwidth is occupied by TCP header

enet.cubik.org

A library that provides reliability, sequencing, connection managements over of

Delivery can be streamoriented (like TCP) or message-oriented (like UDP)

Supports partial reliability

```
enet_packet_create ("abc", 4, ENET_PACKET_FLAG_RELIABLE)
```

Retransmission triggered by Timeout based on RTT

Data in queue are bundled into one packet if there is space

enet.cubik.org

Portable, easy to use, but still, most firewalls block UDP traffic

- MMORPG that uses TCP: WoW, Lineage I/II, Guild Wars, Ragnarok Online, Anarchy Online, Mabinogi
- MMORPG that uses **UDP**:
 EverQuest, SW Galaxies, City of Heroes, Ultima Online, Asherons Call, FFXI

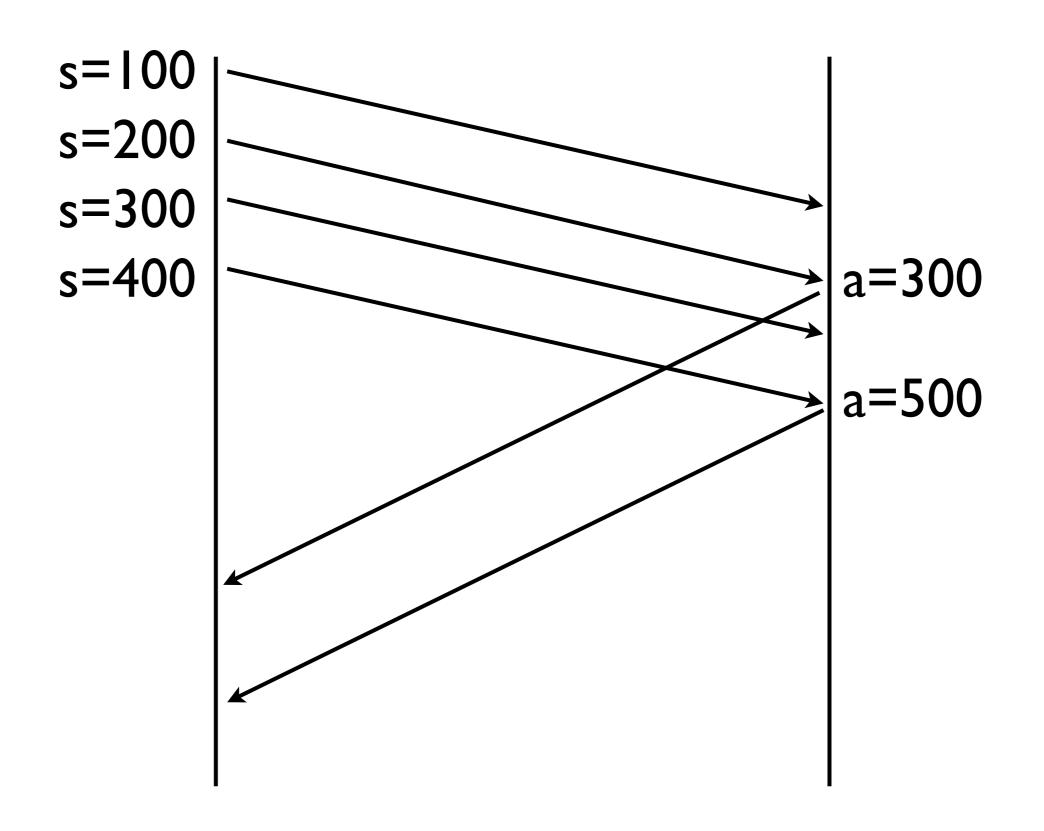
Need to study the use of TCP for networked games

How to provide reliability over UDP?

How slow is TCP, really?

Which part of TCP is the root of slowness?

A Quick Review of TCP

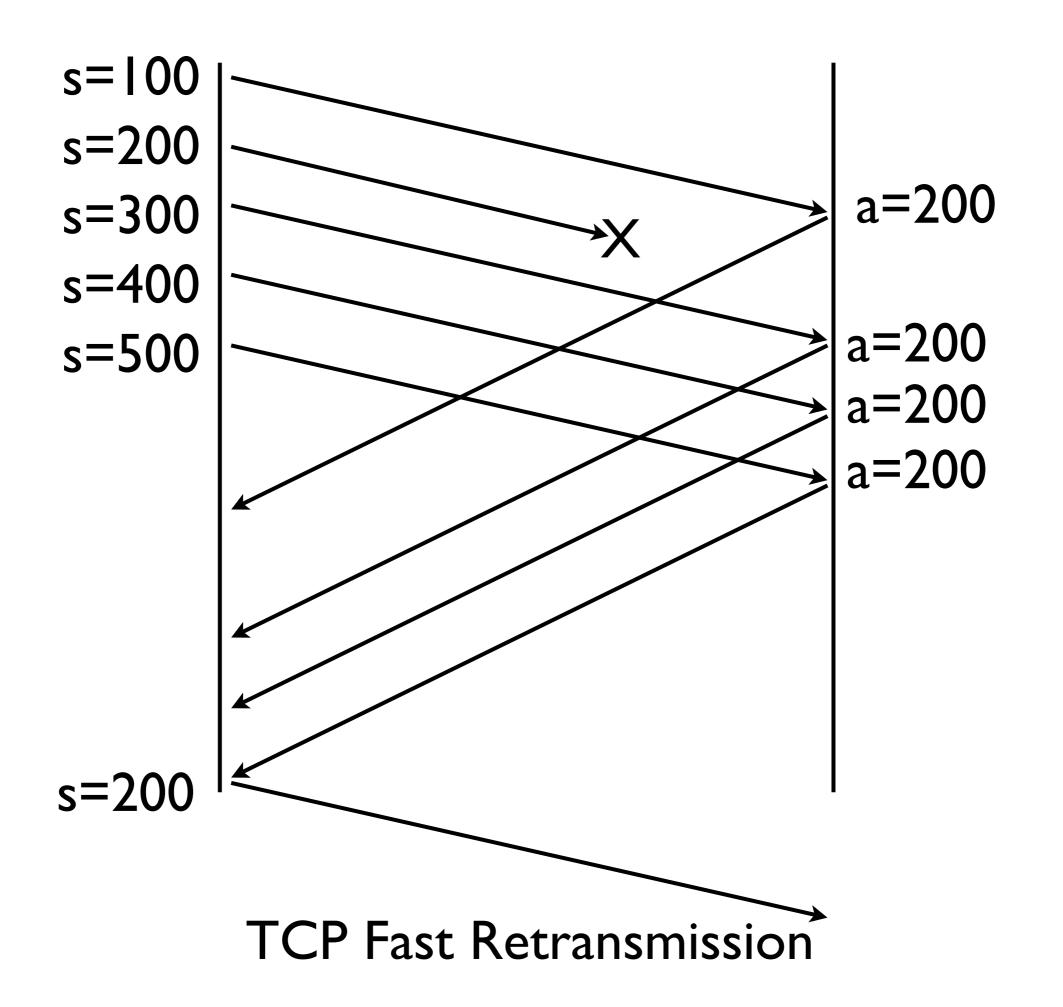


TCP Delayed ACK

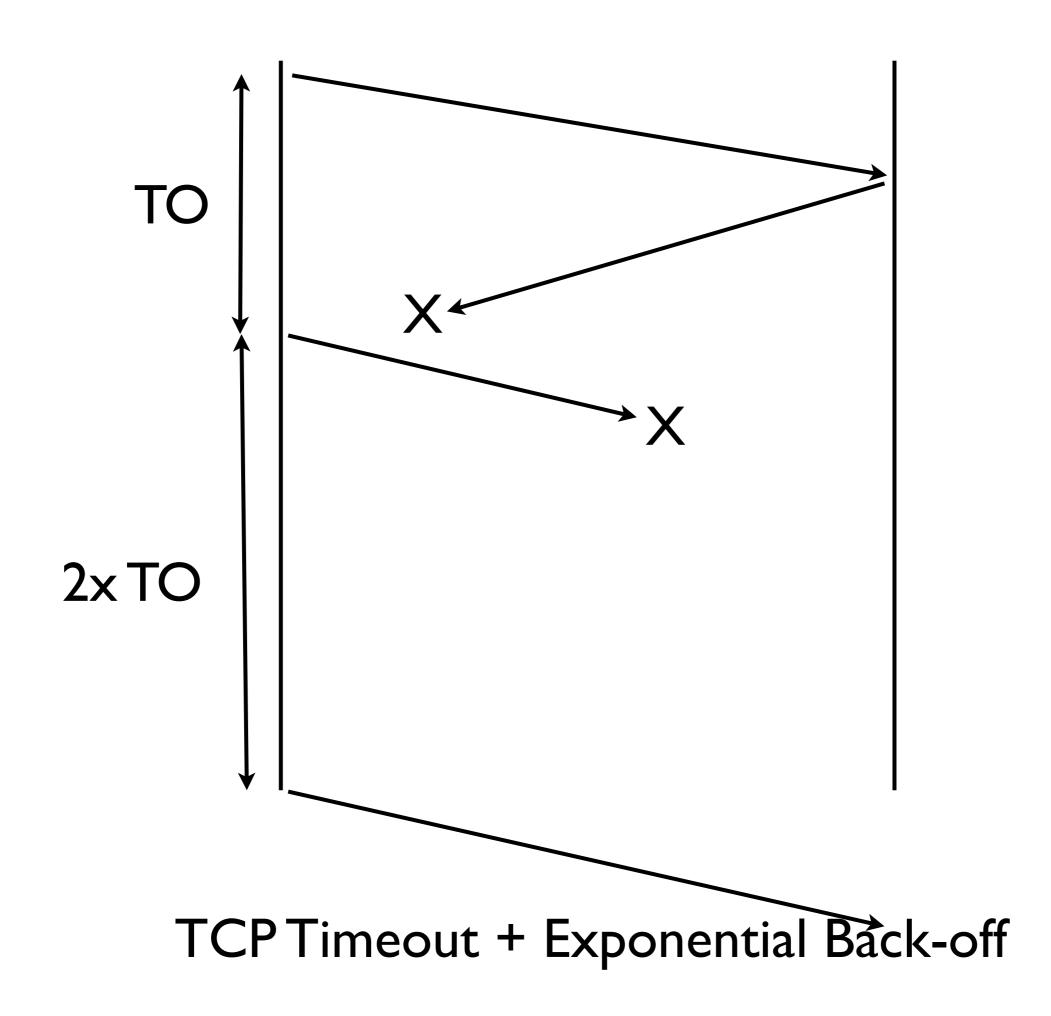
TCP Spec: max **500ms**delay Most implementation: **200ms**

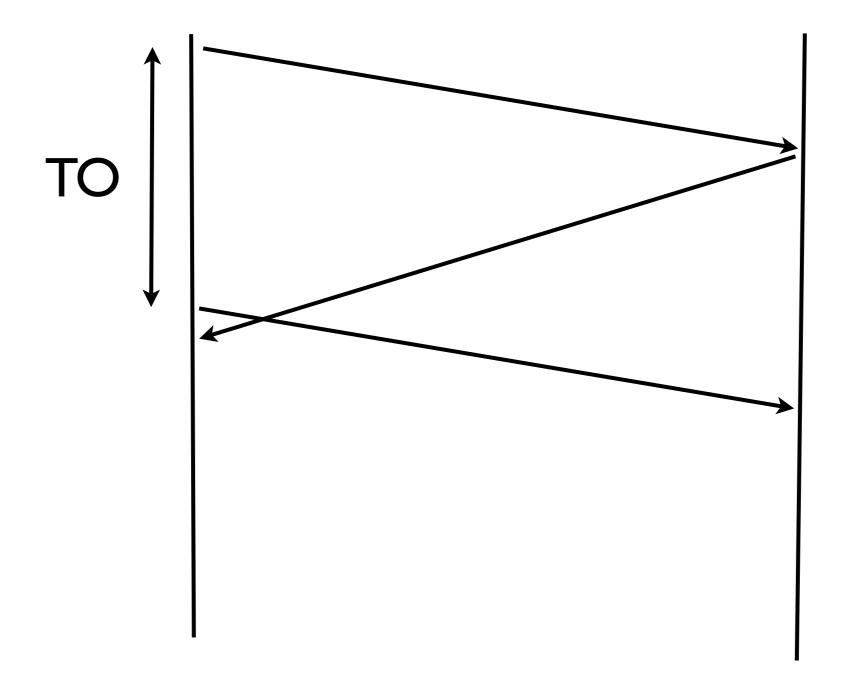
Why delay ACK?

- reduce num of ACKs
- in case receiver wants to send data within 200ms (in which case it can piggyback the ACK with data)
- give sender time to buffer more data for sending (avoid silly window syndrome)



Definition of Dup ACKs in 4.4BSD and Al Stevens: "pure ACK, cannot piggyback with data"





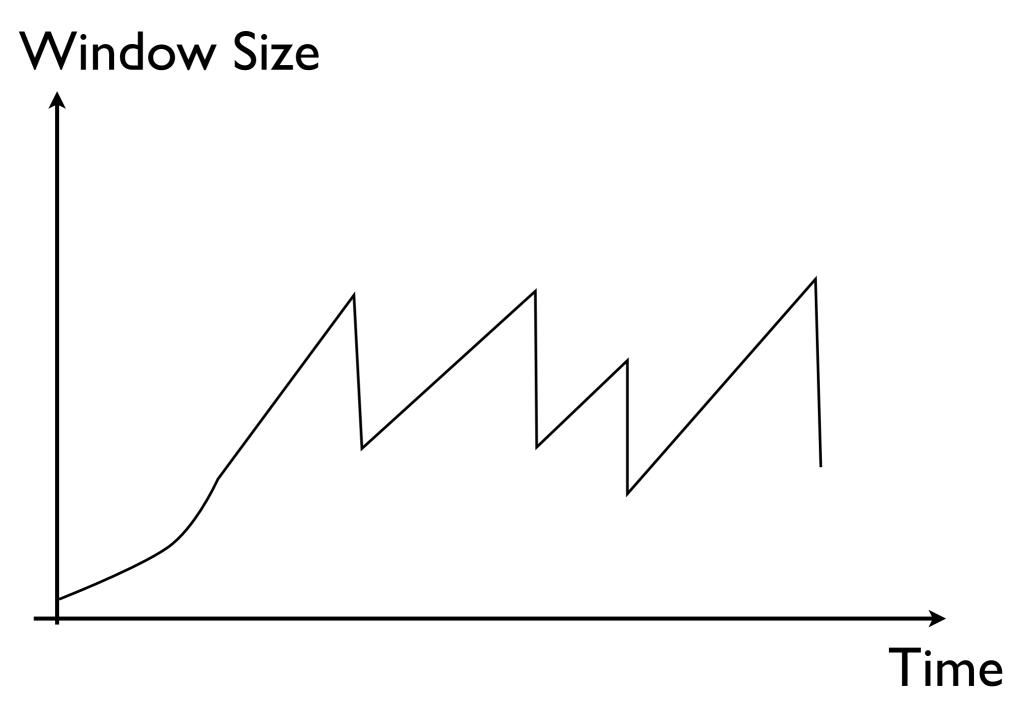
Spurious Retransmission

RTO estimation $E_i = 7E_{i-1}/8 + RTT/8$ $V_i = 3V_{i-1}/4 + |RTT-E_{i-1}|/4$ RTO = max(E_i + 4V_i, Is)

Linux's RTO estimation $E_i = 7E_{i-1}/8 + RTT/8$ $V_i = 3V_{i-1}/4 + |RTT-E_{i-1}|/4$ RTO = max(200ms, $E_i + min(V_i, 50ms)$

Note: Delayed ACK => increase RTT => increase RTO

Congestion Control

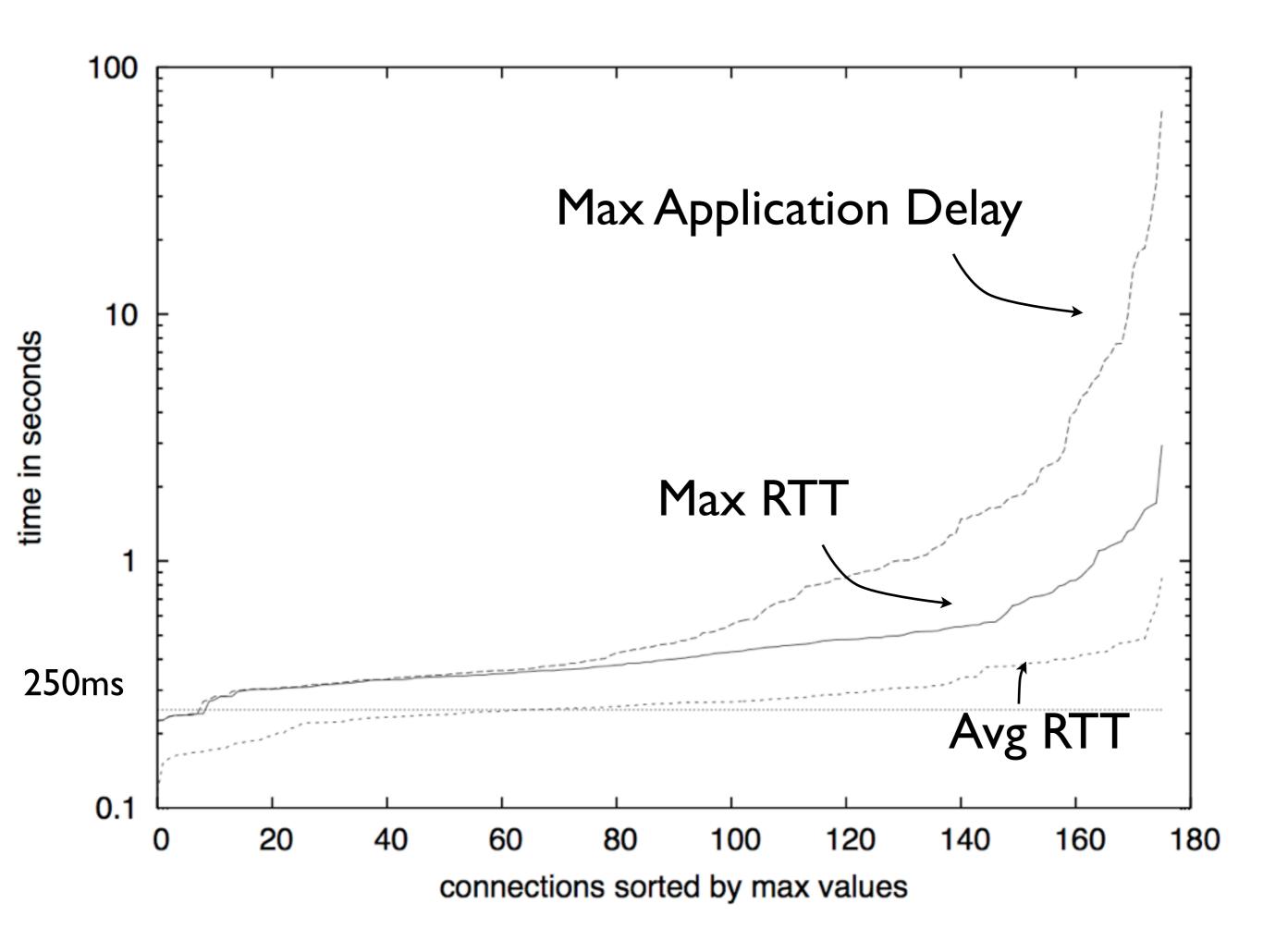


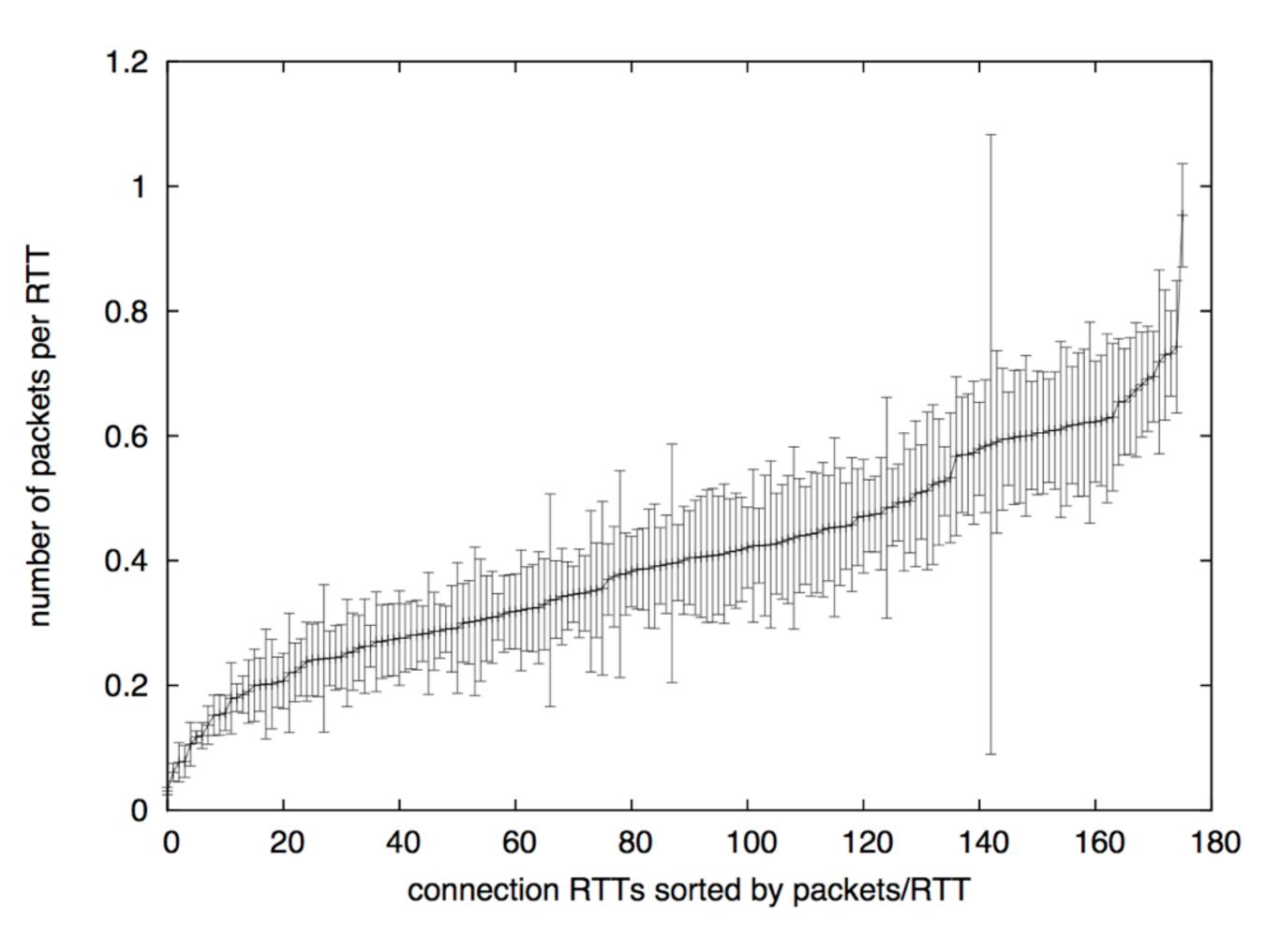
TCP Congestion Control

Congestion Window reset to 2 after an idle period (> RTO)

What does real game traffic look like?







About 4 packets / sec

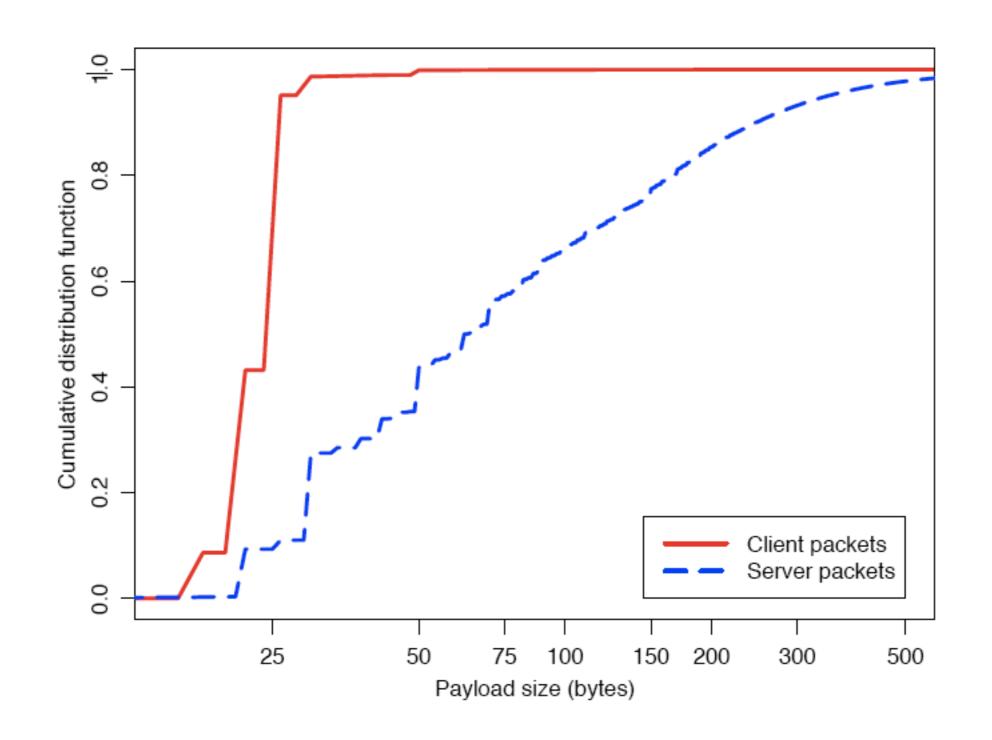
Average Payload: I 00 Bytes

Loss Rate 1%

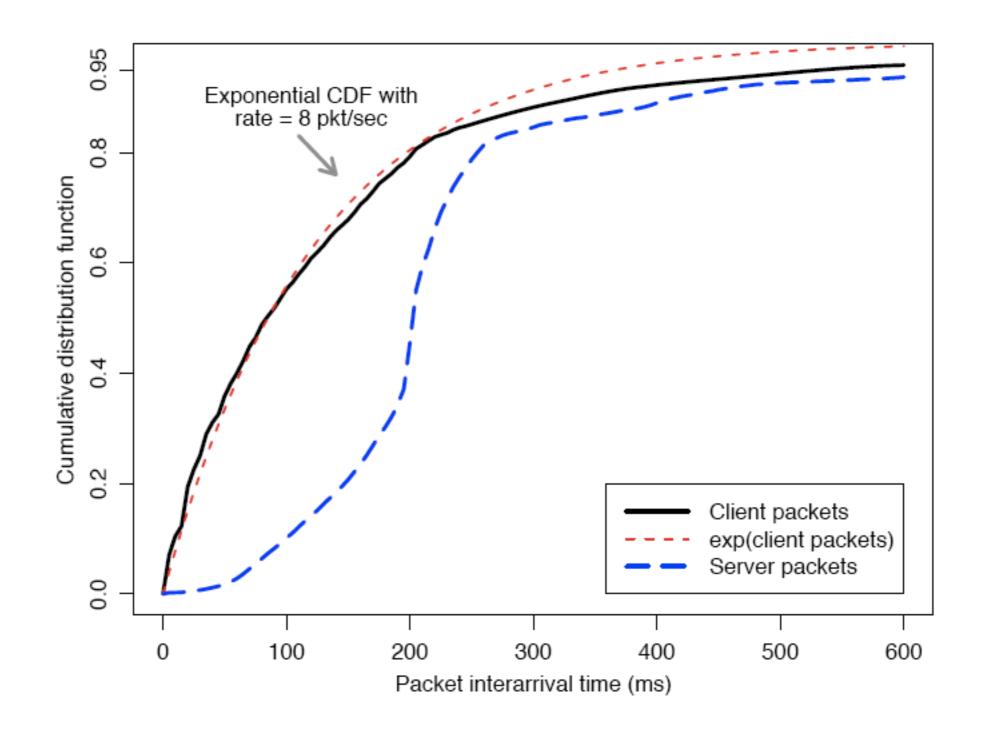
But some experience 6 retransmissions

ShenZhou Online

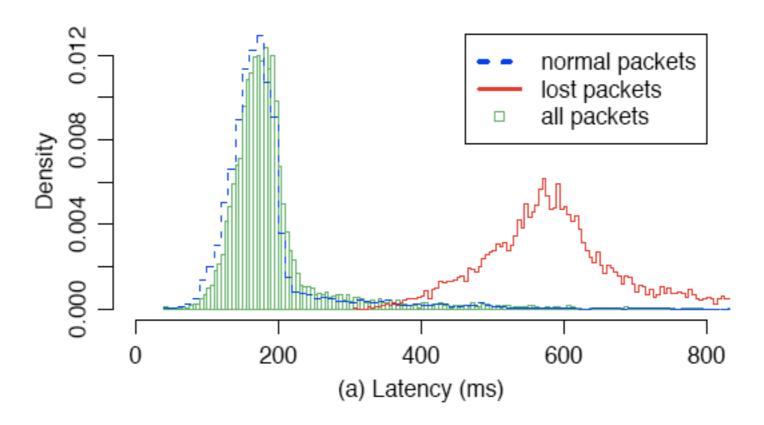




Payload size (bytes)







Similar stats for other games

	payload			
application			_	
(platform)	average	min	max	ave
Anarchy Online (PC) [‡]	98	8	1333	
World of Warcraft (PC)	26	6	1228	
Counter Strike (PC) [‡]	36	25	1342	
Halo 3 (Xbox 360) ^{†‡}	247	32	1264	
Halo 3 (Xbox 360) ^{†‡}	270	32	280	
Gears of War (Xbox 360) [‡]	66	32	705	
Tony Hawk's Project 8 (Xbox 360) [‡]	90	32	576	
Test Drive Umlimited (Xbox 360) [‡]	80	34	104	

[†] For Halo 3 (beta version), we also show differences between intens

Table 1: Examples of game stream packet st

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[‡] The presented values are average values over all players (sending :

packet interarrival time (ms)			avg. bandwidth				
		percentiles		requirement			
average	median	min	max	1%	99%	(pps)	(bps)
632	449	7	17032	83	4195	1.582	2168
314	133	0	14855	0	3785	3.185	2046
124	65	0	66354	34	575	8.064	19604
36	33	0	1403	32	182	27.778	60223
67	66	32	716	64	69	14.925	35888
457	113	3	10155	14	8953	2.188	10264
308	163	0	4070	53	2332	3.247	5812
40	33	0	298	0	158	25.000	22912

intensive (the upper row) and moderate (the lower row) action. ding minimum 1000 packets) within the period of the trace.

et statistics per stream based on packet traces

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"Thin Streams"

Findings I: Fast retransmission rarely triggered

In ShenZhou Online traces, fail to trigger fast retransmission because insufficient dup ACK (50%) interrupted by data (50%)

Findings 2: Delay due mostly to timeout

Retrx latency

RTO + RTT

RTO

0.006

0.005

Figure 9: Average latency of dropped packets

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Findings 3: Congestion Window reset is frequent

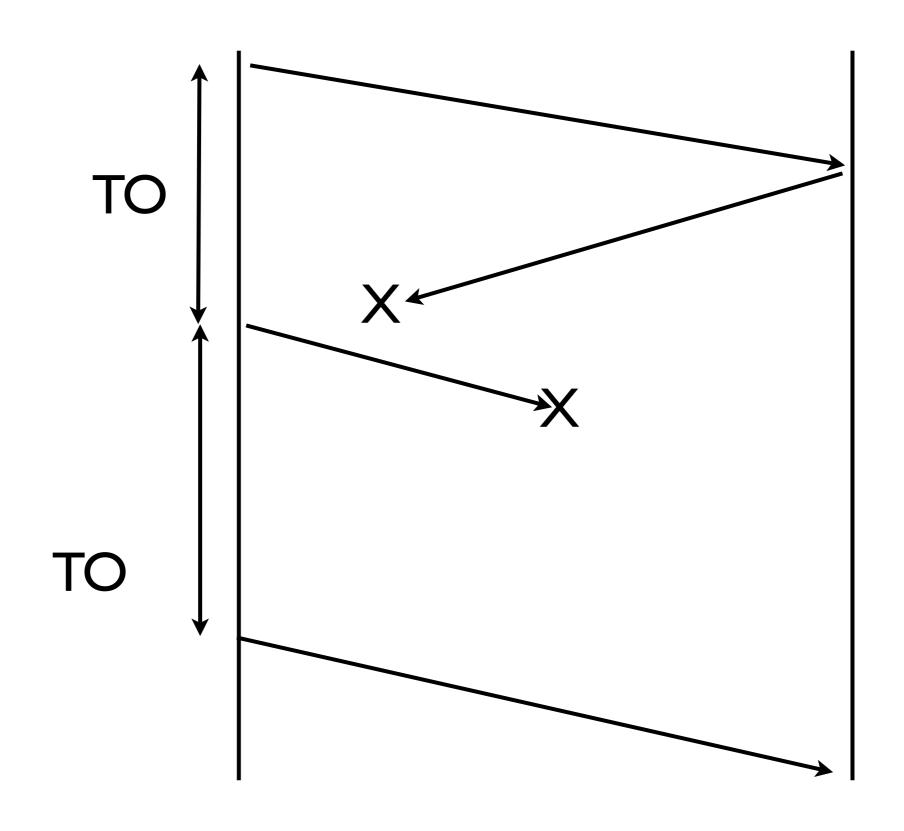
12% - 18% of packets faces window reset

```
think..
think..
think..
click (tank attack here)
click (missle launch there)
click (charge soldiers)
```

The last command is delayed as congestion window = 2

How to make TCP (or, transport protocol) go faster in these games?

I. Remove exponential backoff



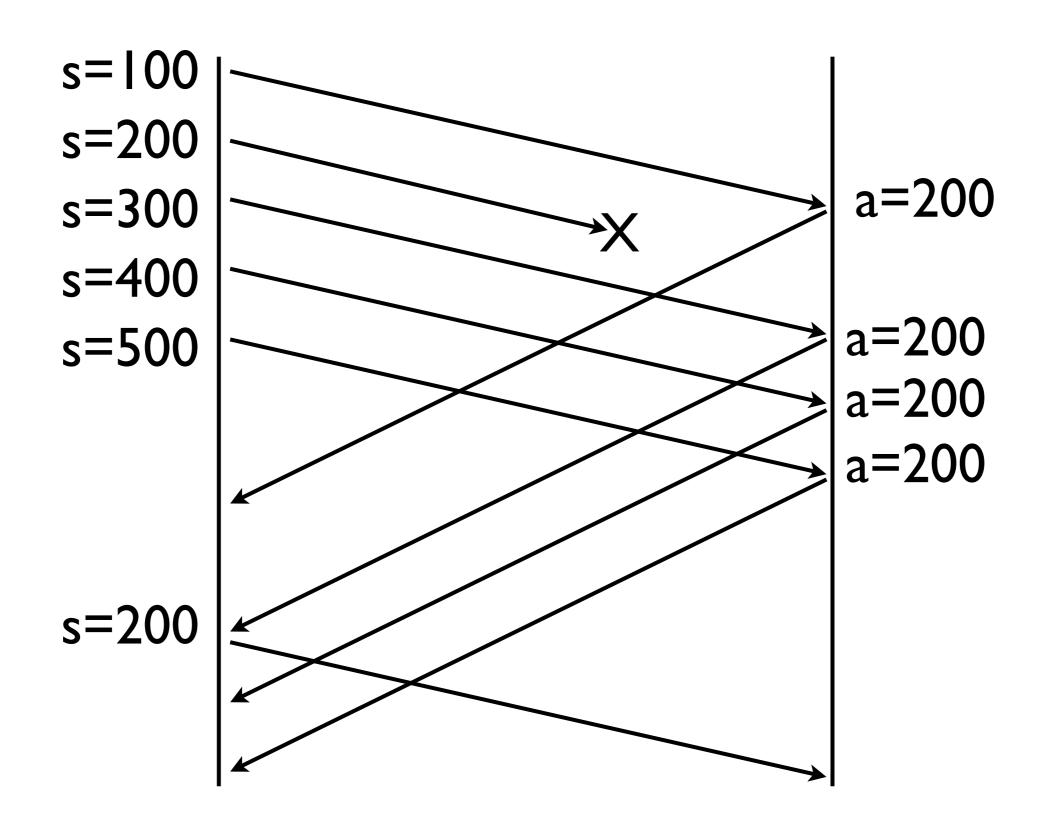
TCP Timeout

2. Make RTO Smaller

make sure minimum RTO is not Is

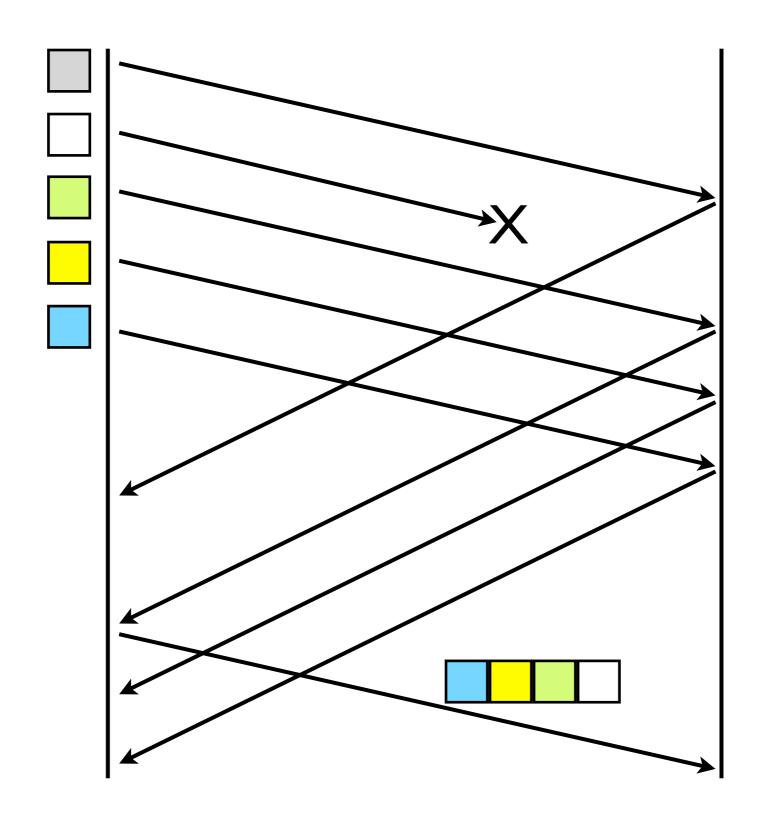
spurious retransmission is not disastrous

3. Make Fast Retransmit Faster



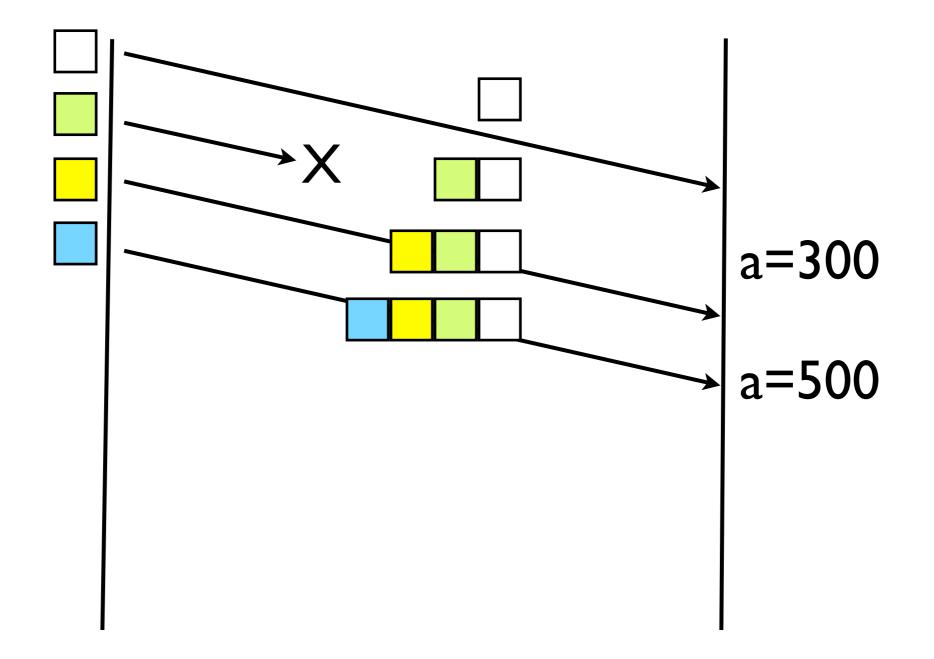
Retransmit after one duplicate ACK

4. Retransmission Bundling



Retransmit all unacknowledge data in queue

5. Redundant Data Bundling

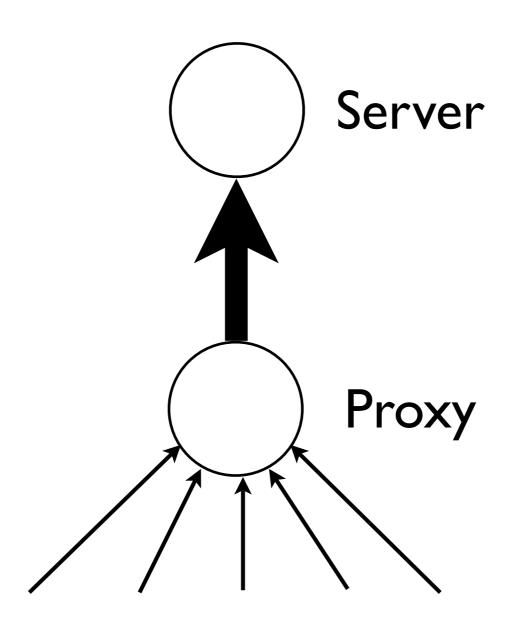


Send any unacknowledged segment in queue as long as there is space. Lost data gets recovered in the next transmission before retransmission.

6. Turn off or reduce Delayed ACKs

Packet interarrival time on average > 200ms (can't combine two ACKs into one)

7. Combine Thin Streams into Thicker Stream



remove exponential backoff

- remove exponential backoff
- reduce RTO

- remove exponential backoff
- reduce RTO
- make fast retransmit faster

- remove exponential backoff
- reduce RTO
- make fast retransmit faster
- retransmit agressively

- remove exponential backoff
- reduce RTO
- make fast retransmit faster
- retransmit agressively
- don't delay ACK

- remove exponential backoff
- reduce RTO
- make fast retransmit faster
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- don't delay ACK
- combine into thick streams