L4 – practical examples



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Ports

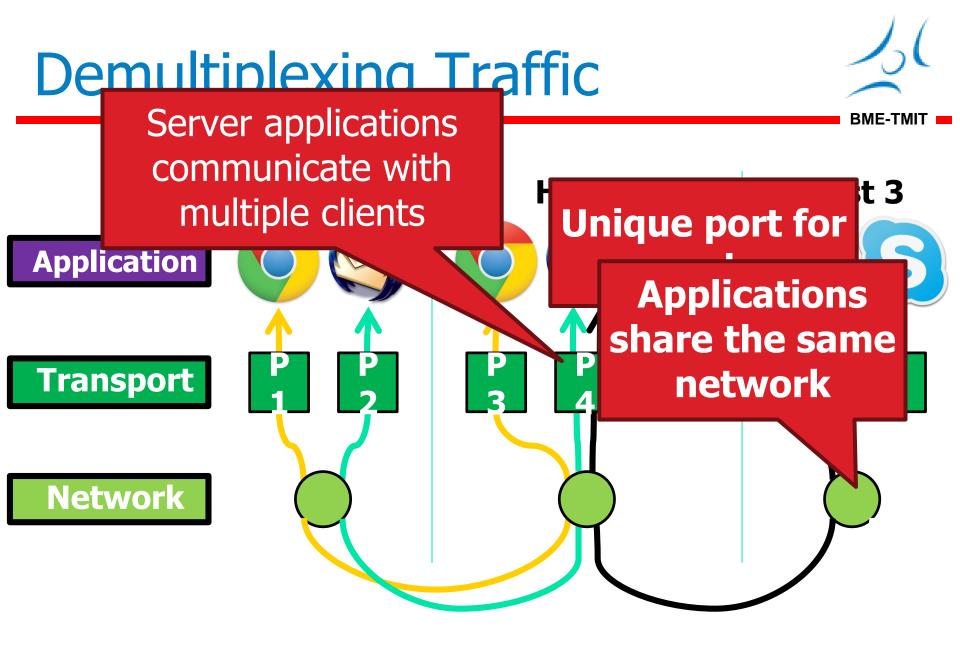


A connection 5-tuple

- Setup a connection
 - UDP no setup
 - TCP
 - -3 way handshake
 - Why it is needed?

Netstat command

TCP 127.0.0.1:1906 localhost:1907 **ESTABLISHED** TCP 192.168.1.147:53699 13.77.87.52:https **ESTABLISHED** TCP 192.168.1.147:53703 91.190.216.57:12350 **FSTABLISHED** TCP 192.168.1.147:53737 64.4.23.152:40008 **ESTABLISHED** TCP 192.168.1.147:53759 108.177.96.188:5228 **ESTABLISHED** TCP 192.168.1.147:53772 40.77.226.192:https **ESTABLISHED** TCP 192.168.1.147:54512 a104-96-129-73:https **CLOSE WAIT** TCP 192.168.1.147:54513 a104-96-129-73:https CLOSE_WAIT TCP 192.168.1.147:54514 a104-96-129-73:https **CLOSE WAIT**



Endpoints identified by <src_ip, src_port, dest_ip, dest_port>

UDP



Data segments

Error handling

- ICMP: port unreachable
- Loss: no feedback

• Delay, bandwidth

Multicast!

The Evolution of TCP



TCP Tahoe

Initial version

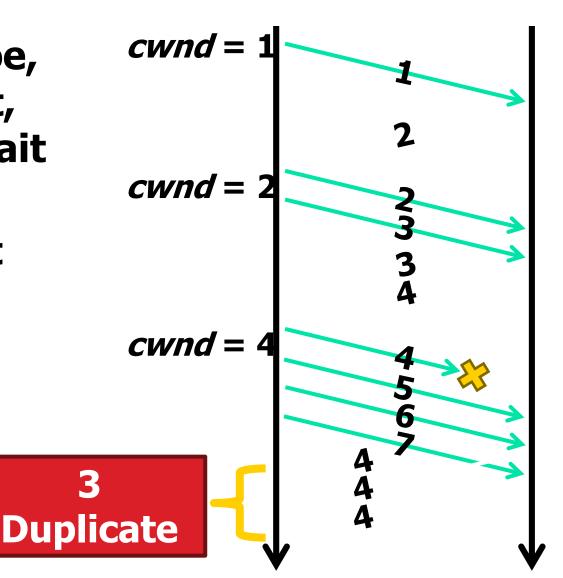
• The TCP was developed in 1974!

- Today there are many versions of TCP
- A widely spread initial version: TCP Reno
 - Tahoe, plus...
 - Fast retransmit
 - Fast recovery





 Reno: retransmit after 3 duplicate ACKs



TCP Reno: Fast Recovery

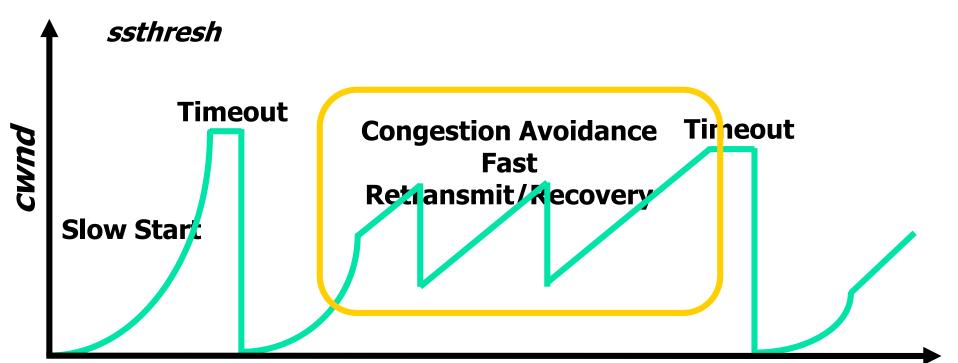
- After a fast-retransmit set *cwnd* to *ssthresh/2*
 - i.e. don't reset cwnd to 1

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- Avoid unnecessary return to slow start
- Prevents expensive timeouts
- But when RTO expires still do cwnd = 1
 - Return to slow start, same as Tahoe
 - Indicates packets aren't being delivered at all
 - i.e. congestion must be really bad

Fast Retransmit and Fast Recovery





Time

- At steady state, *cwnd* oscillates around the optimal window size
- TCP always forces packet drops

人(

- Tahoe: the original
 - Slow start with AIMD
 - Dynamic RTO based on RTT estimate
- Reno: fast retransmit and fast recovery
- NewReno: improved fast retransmit
 - Each duplicate ACK triggers a retransmission
 - Problem: >3 out-of-order packets causes pathological retransmissions
- Vegas: delay-based congestion avoidance
- And many, many, many more...



- What are the most popular variants today?
 - Key problem: TCP performs poorly on high bandwidthdelay product networks (like the modern Internet)
 - Compound TCP (Windows)
 - Based on Reno
 - Uses two congestion windows: delay based and loss based
 - Thus, it uses a *compound* congestion controller
 - TCP CUBIC (Linux)
 - Enhancement of BIC (Binary Increase Congestion Control)
 - Window size controlled by cubic function
 - Parameterized by the time T since the last dropped packet

Things you should keep in mind about TCP...



- When you are a programmer
- When you are a network operator
- Why the difference?
 - As programmer you don't care about the network between the endpoints
 - As operator, you don't care about application needs



- There is traffic even before sending the first byte of data
 - The three-way handshake, blocking
- 2. Data arrives as in blocks
 - Unless push/urgent is used (rarely is used)
- 3. TCP throughput is affected by application buffer reads
 - Sometimes this is a good thing



- Bandwidth delay product
- 8k bandwidth is limited
 - Window is a limiting factor when delay is high
- 64K maximum possible without options
 better
 - Window scale option scale up the window field

Bandwidth delay product



Optimal value for window can be calculated
Win = RTT * Bandwidth

- Similarly,
 - Bandwidth = win / RTT
- Assuming that there is no other bottleneck in the network

TCP options example



No. Time Source Destination Protocol Length Info 4 0.168986 192.168.0.11 239.255.255.250 SSDP 175 M-SEARCH * HTTP/1.1 5 0.221892 fe80::d0f9:8c1:d62f:eb63 ff02::1:3 86 Standard query 0x7e01 A isatap LLMNR 6 0.000117 192.168.0.11 224.0.0.252 LLMNR 66 Standard guery 0x7e01 A isatap < | 111 Frame 12: 66 bytes on wire (528 bits), 66 bytes captured (528 bits) on interface 0 Ethernet II, Src: Wistron 2d:ab:ba (00:1f:16:2d:ab:ba), Dst: 3Com 03:04:05 (00:01:02:03:04:05) Internet Protocol Version 4, Src: 192.168.0.11, Dst: 192.168.0.168 Iransmission Control Protocol, Src Port: 29385, Dst Port: 22, Seq: 0, Len: 0 Source Port: 29385 Destination Port: 22 [Stream index: 0] [TCP Segment Len: 0] Sequence number: 0 (relative sequence number) [Next sequence number: 0 (relative sequence number)] Acknowledgment number: 0 1000 = Header Length: 32 bytes (8) Flags: 0x002 (SYN) Window size value: 8192 [Calculated window size: 8192] Checksum: 0x822a [unverified] [Checksum Status: Unverified] Urgent pointer: 0 Options: (12 bytes), Maximum segment size, No-Operation (NOP), Window scale, No-Operation (NOP), No-Operation (NOP), SACK permitted TCP Option - Maximum segment size: 1460 bytes TCP Option - No-Operation (NOP) TCP Option - Window scale: 2 (multiply by 4) TCP Option - No-Operation (NOP) TCP Option - No-Operation (NOP) TCP Option - SACK permitted 4 [Timestamps] [Time since first frame in this TCP stream: 0.000000000 seconds]

[Time since previous frame in this TCP stream: 0.000000000 seconds]



- 4. Bandwidth may be limited by the network
 - Problem when a required bandwidth is not avail
- 5. Socket options
 - Some TCP algorithm behavior can be changed
- 6. Reuse of addresses
 - Fin bit close of connection still lingering

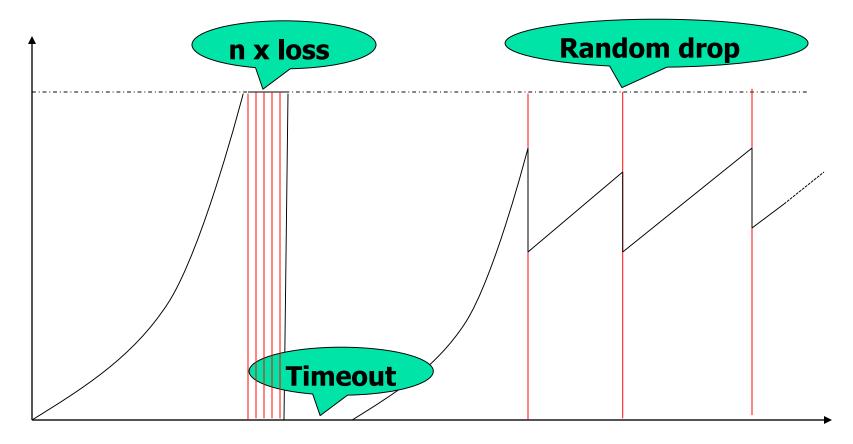


- 1. Type of traffic generated
 - Bursty by nature
- 2. Bottleneck detection
 - Traffic never goes over ~80% (aggregate)
- 3. Congestion control algorithms in routers
 - To handle congestion in advance
 - Achieve fairness
- 4. Lossy channels- radio
 - Misinterpreted as bottleneck

RED – Random Early Drop



• A single loss is better than a timeout



TCP - Wireless



• Problem:

- Wireless loss due the radio
 - no bottleneck!
 - TCP misunderstands, reduces the cwnd

Solutions

- WTCP proxy
- SACK selective acknowledgements

Thank You!

- End -



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