L41: Lab 5 TCP Latency and Bandwidth

Lecturelet 5

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L41: Lab 5 – TCP Latency and Bandwidth

- Lab 5 topic and questions
- TCP congestion control
- TCP Protocol Control Block (TCPCB)
- ARM DTrace limitations



Lab 5 – TCP congestion control

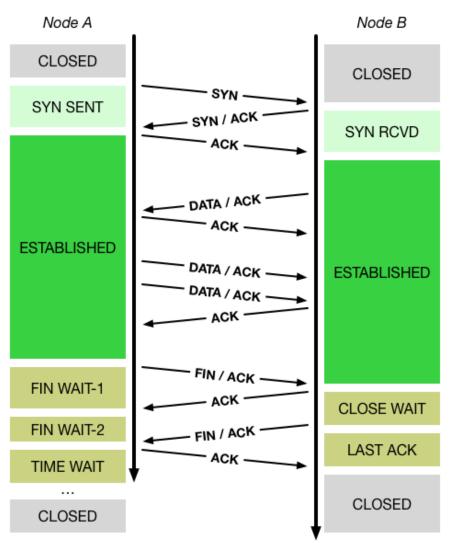
- This lab explores the behavior the TCP implementation and the bandwidth it achieves as latency is varied
 - How does TCP congestion control affect bandwidth at different latencies?
 - What are the impacts of specific implementation choices and policies, such as socket-buffer auto-sizing
- As we are working over the loopback interface, we can instrument both ends of the TCP connection
 - Track packet-level headers on transmit and receive
 - Also track TCP-internal parameters such as whether TCP is in "slow start" or the steady state
- And, of course, we care about the arising probe effect

Experimental questions for the lab report

- 1. Plot DUMMYNET-imposed latency on the X axis and effective bandwidth on the Y axis, considering both the case where the socket-buffer size is set versus allowing it to be auto-resized.
 - Is the relationship between round-trip latency and bandwidth linear?
 - How does socket-buffer auto-resizing help, hurt, or fail to affect performance as latency varies?
- 2. Plot a time-bandwidth graph comparing the effects of setting the socket-buffer size versus allowing it to be auto-resized by the stack. Stack additional graphs showing the sender last received advertised window and congestion window on the same X axis.
 - How does socket-buffer auto-resizing affect overall performance, as explained in terms of the effect of window sizes?
- 3. Be sure to describe any simulation or probe effects.

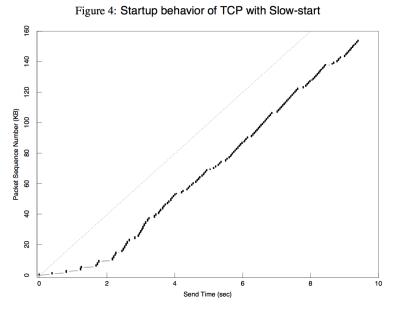


Lecture 6: TCP goals and properties



- Network may delay, (reorder), drop, corrupt packets
- TCP: Reliable, ordered, stream transport protocol over IP
- Three-way handshake: SYN / SYN-ACK / ACK (mostly!)
- Sequence numbers ACK'd; data retransmitted on loss
- Round-Trip Time (RTT) measured to time out loss
- Flow control via advertised window size in ACKs
- Congestion control ('fairness') via packet loss and ECN

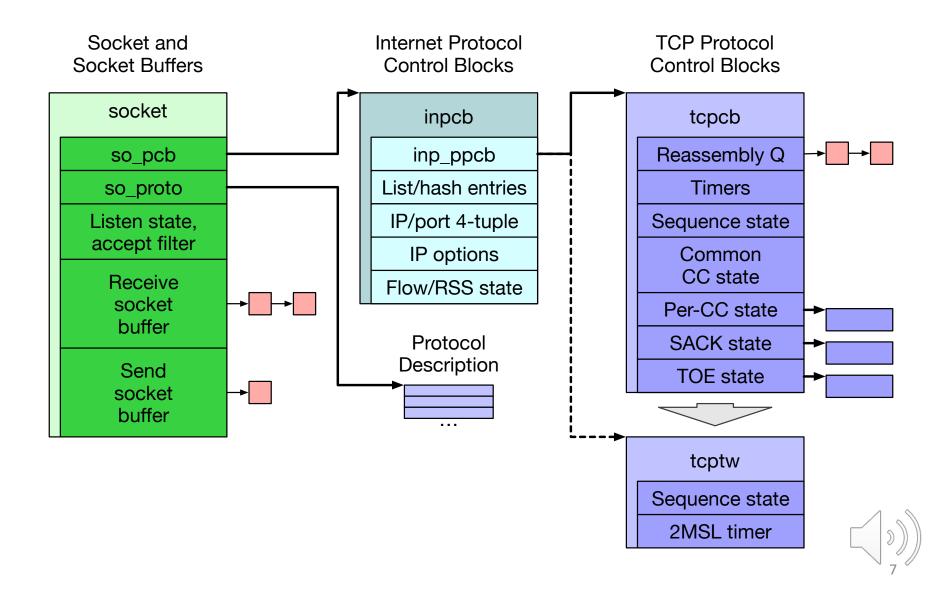
Lecture 6: TCP congestion control and avoidance



Same conditions as the previous figure (same time of day, same Suns, same network path, same buffer and window sizes), except the machines were running the 4.3^{+} TCP with slow-start. No bandwidth is wasted on retransmits but two seconds is spent on the slow-start so the effective bandwidth of this part of the trace is 16 KBps — two times better than figure 3. (This is slightly misleading: Unlike the previous figure, the slope of the trace is 20 KBps and the effect of the 2 second offset decreases as the trace lengthens. E.g., if this trace had run a minute, the effective bandwidth would have been 19 KBps. The effective bandwidth without slow-start stays at 7 KBps no matter how long the trace.)

- 1986 Internet CC collapse
 - 32Kbps → 40bps
- Van Jacobson, SIGCOMM 1988
 - Don't send more data than the network can handle!
 - Conservation of packets via ACK clocking
 - Exponential retransmit timer, slow start, aggressive receiver ACK, and dynamic window sizing on congestion
- ECN (RFC 3168), ABC (RFC 3465), Compound (Tan, et al, INFOCOM 2006), Cubic (Rhee and Xu, ACM OSR 2008), BBR (Cardwell et al, ACM Queue 2016)

Lecture 6: Data structures – sockets, control blocks



tcpcb sender-side data-structure fields

- In this lab, there are two parties with **tcpcb**s as we run:
 - The 'client' is receiving data
- For the purposes of classical TCP congestion control, only the sender retains congestion-control state
- Described in more detail in the lab assignment:

snd_wndLast received advertised flow-control window.snd_cwndCurrent calculated congestion-control window.snd_ssthreshCurrent slow-start threshold:

if (snd_cwnd <= snd_ssthresh), then TCP is in slowstart; otherwise, it is in congestion avoidance

- Instrument tcp_do_segment using DTrace to inspect TCP header fields and tcpcb state for only the server
 - Inspect port number to decide which way the packet is going
- NB: Flush the TCP host cache between benchmark runs



ARM DTrace limitations (1/2)

In previous years, we had suggested that the TCP segment length can be computed as follows:

```
tdatalen = ntohs(((struct ip *)args[0]->m_data)-
>ip_len) - ((((struct ip *)args[0]->m_data)->ip_hl <<
2) + (args[1]->th_off << 2));</pre>
```

However, a bug in ARM DTrace resulted in the indexing of the m_data field dereferencing NULL

The TCP segment length should instead by measured directly from the ip_length field in the struct ipinfo_t structure (accessible via args[2] in the tcp:::send probe)



ARM DTrace limitations (2/2)

- FreeBSD's DTrace implementation restricts the creation of trace buffer sizes that exceed a fixed percentage of the available kernel memory
- Unfortunately, for small memory boards such as the BBB this is overly restrictive and prevents the allocation of trace buffers greater than 3MB:

#pragma D option bufsize=3M
#pragma D option bufresize=manual

 When running the benchmark, it is acceptable to limit your experiment to a total buffer sizes that does not result in drops (exceeding space in the trace buffer)

This lab session

- Ensure that you are able to properly extract both TCP header and tcpcb fields from the tcp_do_segment FBT probe.
- Generate the data for a time–bandwidth graph.
- Ask us if you have any questions or need help.

