CSCI-1680 Transport Layer II

Data over TCP: Flow Control

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Based partly on lecture notes by Rodrigo Fonseca, David Mazières, Phil Levis, John Jannotti

"Hi, I'd like to hear a TCP joke." "Hello, would you like to hear a TCP joke?" "Yes, I'd like to hear a TCP joke." "OK, I'll tell you a TCP joke." "Ok, I will hear a TCP joke." "Are you ready to hear a TCP joke?" "Yes, I am ready to hear a TCP joke." "Ok, I am about to send the TCP joke. It will last 10 seconds, it has two characters, it does not have a setting, it ends with a punchline." "Ok, I am ready to get your TCP joke that will last 10 seconds, has two characters, does not have an explicit setting, and ends with a punchline." "I'm sorry, your connection has timed out. ... Hello, would you like to hear a TCP joke?"

Administrivia

- IP project grading: happening now! Sign up for a meeting if you haven't already
- TCP assignment: out now—<u>start early</u>!
 - Gearup I: Thursday 10/26 5-7pm
 - Milestone 1: schedule meeting on/before Thursday, November 2

TCP: The story so far

Last lecture

- Sockets
- TCP: connection setup

Today

- Basic flow control: How to send data
- Connection teardown

TCP – Transmission Control Protocol



TCP provides a "reliable, connection oriented, full duplex ordered byte stream"

TCP Header

0	D 15_16 31					
	Source Po	ort	Destination Port			
Sequence Number						
Acknowledgement Number						
Data Offset	Data Offset Reserved R C S S Y I Offset Reserved R C S S Y I G K H T N N					
Checksum Urgent Pointer						
Coptions 2						
Data 2						

Important Header Fields

- Ports: multiplexing
- Sequence number
 - Where segment is in the stream (in bytes)
- Acknowledgment Number
 - Next expected sequence number
- Window
 - How much data you're willing to receive
- Flags...

Important Header Fields: Flags

- SYN: establishes connection ("synchronize")
- ACK: this segment ACKs some data (all packets except first)
- FIN: close connection (gracefully)
- RST: reset connection (used for errors)
- PSH: push data to the application immediately
- URG: whether there is urgent data

Less important header fields

- Checksum: Very weak, like IP
 - Has weird semantics ("pseudo header"), more on this later...
- Data Offset: used to indicate TCP options (mostly unused)
- Urgent Pointer

Establishing a Connection



- Three-way handshake
 - Two sides agree on respective initial sequence nums
- If no one is listening on port: OS may send RST
- If server is overloaded: ignore SYN
- If no SYN-ACK: retry, timeout

Summary of TCP States



TCP State Diagram



We are now here



State for a TCP connection kept in <u>Transmission Control Buffer (TCB)</u>

 Keeps initial sequence numbers, connection state, send/recv buffers, status of unACK'd segments, ...

When to allocate?

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<u>When to allocate?</u>

- Server: listening on a connection*
- Client: Initiating a connection (sending a SYN)
- Server: accepting a new connection (receiving SYN)

 \Rightarrow When to deallocate?

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When to deallocate?

Only after connection termination is fully completed (CLOSED state) => If no state, can't meaningfully respond to packet!

NOTA BENE: This diagram is only a summary and must not be taken as the total specification. Many details are not included.



RFC 9293, Sec 3.3.2

Recall: the socket table

deemer@vesta ~ % netstat -anl Active Internet connections (including servers)					
Proto Recv-Q Send-Q		nd-Q	Local Address	Foreign Address	(state)
tcp4	0	0	172.17.48.121.56915	192.168.1.58.7000	SYN_SENT
tcp4	0	0	172.17.48.121.56908	142.250.80.35.443	ESTABLISHED
tcp4	0	0	172.17.48.121.56887	13.225.231.50.80	ESTABLISHED
· · ·					
tcp4	0	0	*.22	* *	LISTEN

- Each connection has an associated TCB in the kernel
- Depending on socket type, socket contains TCB

Recall: the socket table



- Each connection has an associated TCB in the kernel
- Depending on socket type, socket contains TCB

⇒For each packet, kernel maps the 5-tuple (tcp/udp, local IP, local port, remote IP, remote port) => socket 5-tuple (proto., source IP, source port, dest IP, dest port) => 1 Conn – Kernel maintains socket table: maps (5-tuple) => Socket

deemer@vesta ~ % netstat -anl Active Internet connections (including servers)					
Proto Recv	-Q Send	l-Q	Local Address	Foreign Address	(state)
		Ŭ			ΥΥΥΥΥ ΥΥΥΥΥΥΥΥΥΥΥΥΥΥΥΥΥΥΥΥΥΥΥΥΥΥΥΥΥΥΥ
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 If a 5-tuple is reused => new ISN, so sequence numbers likely out of range from past connection 5-tuple (proto., source IP, source port, dest IP, dest port) => 1 Conn – Kernel maintains socket table: maps (5-tuple) => Socket

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Proto Recv-Q Send-Q Local Address Foreign Address (state)					(state)
tcp4	0	0	172.17.48.121:22	192.168.1.58:34452	SYN_SENT
tcp4	0	0	*.22	* *	LISTEN

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tcp4	0	0	172.17.48.121:22	192.168.1.58:34452	SYN_SENT	
tcp4	0	0	172.17.48.121:22	142.250.80.35:11435	ESTABLISHED	
tcp4	0	0	172.17.48.121:22	13.225.231.50:12345	ESTABLISHED	
• • •						
tcp4	0	0	*.22	* *	LISTEN	

 If a 5-tuple is reused => new ISN, so sequence numbers likely out of range from past connection



Two "types" of sockets:

• "Normal" sockets

• Listen sockets

Proto Recv	-Q Send	-Q	Local Address	Foreign Address	(state)
tcp4	0	Ø	172.17.48.121.56887	13.225.231.50.80	ESTABLISHED
tcp4	0	0	*.22	*.*	LISTEN

<u>"Normal" sockets</u>

- Connection between two specific endpoints
- Can send/recv data

<u>Listen sockets</u>

- Created by receiver to accept new connections
- When a client connects, client info gets queued by kernel
- When server process calls accept(), <u>a new ("normal") socket is created</u> between the server and that client

How to pick the initial sequence number?

- Protocols based on *relative* seq. numbers based on starting value
- Why not start at 0?

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=> RFC9293, Sec 3.4.1: Procedure for picking ISN, based on timer and cryptographic hash => For project, just pick a random integer :)

Relative Sequence Numbering

> Ethernet II, Src: Apple_cd:6a:23 (c8:89:f3:cd:6a:23), Dst: IntelCor_63:c4:45 (0	0000 00 1b 21 63 c4 45 c8 89 f3 cd 6a 23 08 00 45 00
> Internet Protocol Version 4, Src: 172.17.48.156, Dst: 172.17.48.22	0010 00 40 00 00 40 00 40 06 81 e3 ac 11 30 9c ac 11
Transmission Control Protocol, Src Port: 49719, Dst Port: 22, Seq: 0, Len: 0	0020 30 16 c2 37 00 16 77 42 38 e5 00 00 00 00 b0 02
Source Port: 49719	
Destination Port: 22	
[Stream index: 8]	
[Conversation completeness: Complete, WITH_DATA (31)]	
[TCP Segment Len: 0]	
Sequence Number: 0 (relative sequence number)	
Sequence Number (raw): 2000828645	
[Next Sequence Number: 1 (relative sequence number)]	
Acknowledgment Number: 0	
Acknowledgment number (raw): 0	
1011 = Header Length: 44 bytes (11)	
> Flags: 0x002 (SYN)	

Observation: new connections use memory!



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What happens if you send a someone lots of SYN packets?

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What happens if you send a someone lots of SYN packets?

SYN flood => type of Denial of Service (DOS) attack



SYN flood => type of Denial of Service (DOS) attack => Especially bad when attack traffic comes from multiple sources (more on this later)

A hacky solution: SYN cookies

- Don't allocate TCB on first SYN
- Encode some state inside the initial sequence number that goes back to the client (in the SYN+ACK)
- What gets encoded?
 - Coarse timestamp
 - Hash of connection IP/port
 - Other stuff (implementation dependent)
- Better ideas?



A hacky solution: SYN cookies

- Don't allocate TCB on first SYN
- Encode some state inside the initial sequence number that goes back to the client (in the SYN+ACK)
- What gets encoded?
 - Coarse timestamp
 - Hash of connection IP/port
 - Other stuff (implementation dependent)
- Nowadays: filtering in kernel (or in network) on number of new connections per time (esp. on servers)
 = > More on this later!



<u>Sending data</u>



Sending data

Flow control: don't send more data than the receiver can handle

• TCP stack divides data into packets called segments

Questions

- When to send data?
- How much data to send?
 - Data is sent in MSS-sized segments
 - MSS = Maximum Segment Size (TCP packet that can fit in an IP packet)
 - Chosen to avoid fragmentation

• Start: app calls Send(), loads send buffer

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Key challenges

- When to send data?
- How much data to send?

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Key challenges

- When to send data?
- How much data to send?

⇒ Flow control (now): don't send more data than the receiver can handle
⇒ Congestion control (much later) don't send more data than the network can handle

Terminology: MSS

MSS: Maximum segment size

- Largest segment a TCP can send
- Can be configurable
- Nowadays: sender and receiver negotiate using TCP options (out of scope for this class)

=> For project: just a fixed value

Simplest TCP sender: stop and wait

Simplest method: Stop and Wait

Consider sending one packet at a time

- S: Send packet, wait
- R: Receive packet, send ACK
- S: Receive ACK, send next packet OR

No ACK within some time (RTO), timeout and retransmit

What can go wrong?

Lost Data









Sequence number example



	A sends	B sends
1	SYN, seq=0	
2		SYN+ACK, seq=0, ack=1 (expecting)
3	ACK, seq=1 , ack=1 (ACK of SYN)	
4	"abc", seq=1 , ack=1	
5		ACK, seq=1, ack=4
6	"defg", seq=4 , ack=1	
7		seq=1, ack=8
8	"foobar", seq=8 , ack=1	
9		seq=1, ack=14 , "hello"
10	seq=14, ack=6, "goodbye"	
11,12	seq=21 , ack=6, FIN	seq=6, ack=21 ;; ACK of "goodbye", crossing packets
13		seq=6, ack=22 ;; ACK of FIN
14		seq=6, ack=22 , FIN
15	seq=22 , ack=7 ;; ACK of FIN	



Problems?

Can we do better?

Better Flow Control: Sliding window

- Part of TCP specification (even before 1988)
- Send multiple packets at once, based on a *window*
- Receiver uses window header field to tell sender how much space it has

TCP and buffering

Recall: TCP stack responsibilities

- Sender: breaking application data into segments
- Receiver: receiving segments, reassembling them in order

• Need to buffer data

Sliding window: in abstract terms

- Window of size w
- Can send at most w packets before waiting for an ACK
- Goal
 - Network "pipe" always filled with data
 - ACKs come back at rate data is delivered => "self-clocking"





Flow Control: Sender



Invariants

- LastByteSent LastByteAcked <= AdvertisedWindow
- EffectiveWindow = AdvertisedWindow (BytesInFlight)
- LastByteWritten LastByteAcked <= MaxSendBuffer

Useful Sliding Window Terminology: RFC 9293, Sec 3.3.1

Flow control: receiver

Useful Sliding Window Terminology: RFC 9293, Sec 3.3.1

- Can accept data if space in window
- Available window = BufferSize- ((NextByteExpected-1) - LastByteRead
- On receiving segment for byte S
 - if s is outside window, ignore packet
 - if s == NextByteExpected:
 - Deliver to application (Update LastByteReceived)
 - If next segment was early arrival, deliver it too
 - If s > NextByteExpected, but within window
 - Queue as early arrival
- Send ACK for highest contiguous byte received, available window



Flow Control

- Advertised window can fall to 0
 - How?
 - Sender eventually stops sending, blocks application
- Resolution: zero window probing: sender sends 1-byte segments until window comes back > 0



Some Visualizations

- Normal conditions: <u>https://www.youtube.com/watch?v=zY3Sxvj8kZA</u>
- With packet loss: <u>https://www.youtube.com/watch?v=lk27yiITOvU</u>

How do ACKs work?

- ACK contains next expected sequence number
- If one segment is missed but new ones received, send duplicate ACK
- Retransmit when:
 - Receive timeout (RTO) expires
 - Possibly other conditions, for certain TCP variants (eg. 3 dup ACKs)
- How to set RTO?

When to time out?

Should expect an ACK within one Round Trip Time (RTT)

- Problem: RTT can be highly variable
- Strategy: expected RTT based on ACKs received
 - Use exponentially weighted moving average (EWMA)
 - RFC793 version ("smoothed RTT"):

SRTT = $(\alpha * SRTT) + (1 - \alpha) * RTT_{Measured}$ RTO = max(RTO_{Min}, min($\beta * SRTT$, RTO_{Max}))

$$\alpha$$
 = "Smoothing factor": .8-.9
 β = "Delay variance factor": 1.3-2.0

RFC793, Sec 3.7

This is only the beginning...

- Problem 1: what if segment is a retransmission?
 Solution: don't update RTT if segment was retransmitted
- Problem 2: RTT can have high variance

 Initial implementation doesn't account for this
 - Congestion control: modeling network load