Don't panic: TCP gearup III



Overview

- Final TCP stuff
- Any questions you have

Roadmap

Milestone I

- Start of your API and TCP stack
- Listen and establish connections => create sockets/TCB
- TCP handshake
- accept, connect, and start of Is REPL commands

Roadmap

Milestone II

- Basic sending and receiving using your sliding window/send receive buffers
- Plan for the remaining features

Roadmap

Final deadline

- Retransmissions (+ computing RTO from RTT)
- Out-of-order packets
- Sending and receiving files (sf, rf)
- Zero-window probing
- Connection teardown ((LL))

Sendfile/Recvfile

Using your socket API, send/recv a file

Sendfile

Open a file, VConnect, call VWrite in a loop

-UP to IMB

Recvfile

Listen on a port, Open a file, call VRead in a loop

=> This is the ultimate test: your implementation should be similar to how you'd use a real socket API!

Demo!

So how do we get there?

Relevant materials

- Lecture 15 (10/26): Sliding window, retransmissions, zero window probing
- Lecture 16 (10/31): connection teardown

Testing and tools stuff: "TCP getting started" in docs
 New Reference for testing with packet loss => announcement soon

<u>Retransmissions</u>

Usually, make a "retransmission queue"

When segment sent, add segment to queue with some metadata

=> What to store? You decide!

LO WHEN YOU SENT IT.

<u>Retransmissions</u>

Usually, make a "retransmission queue"

- When segment sent, add segment to queue with some metadata
 - => What to store? You decide!
- Start RTO timer > ONE TIMEN PEN SOCKET.
- When you get an ACK, reset

<u>Retransmissions</u>

Usually, make a "retransmission queue"

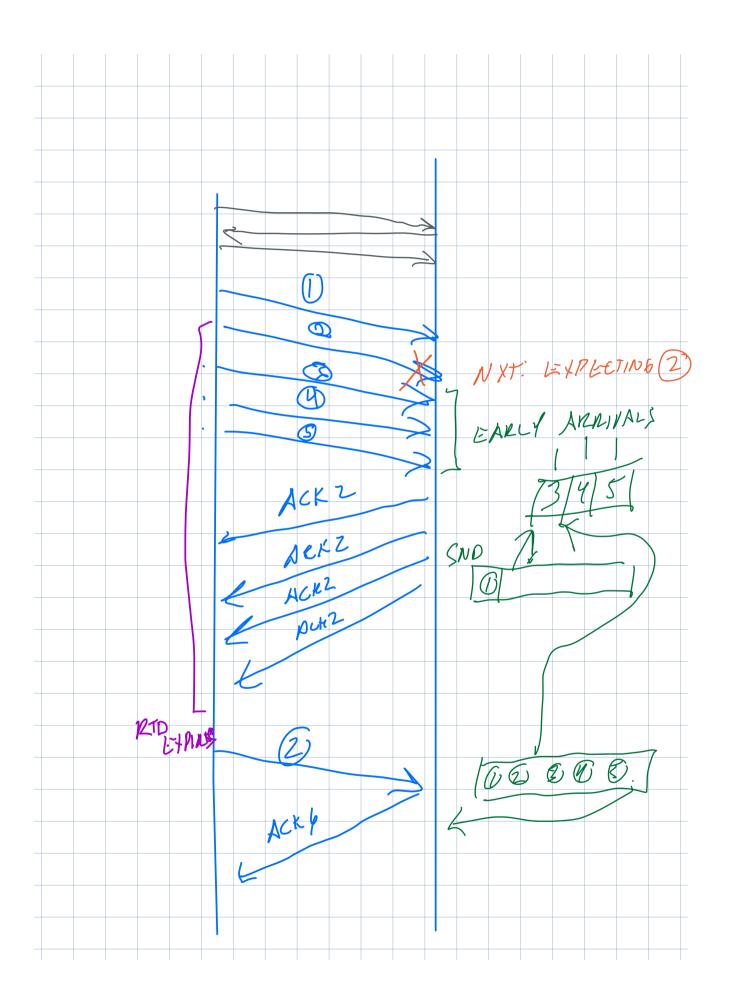
- When segment sent, add segment to queue with some metadata
 - => What to store? You decide!
- Start RTO timer, reset on ACK

When RTO timer expires

- Retransmit earliest unACK'd segment
- RTO = 2 * RTO (up to max)
- If no data after N retransmits => give up, terminate connection

11/2/3/4/

 \Rightarrow RFC6298 is your friend! Use it! (edge cases, etc.)



RTO = Retransmission Timeout (RTO)

WHEN TO

=> Based on expected RTT: "how long until you SHOULD get an ACK?"

Sr

When you get an ACK, update RTO

12TT = ONE REASUMENT

SRTT = SMOOPHED ETT

=> WEIGHTED AVG

Example upper/lower bounds

RTOmin ~= 100ms

RTOmax ~= 5sec

RTO?

RTO = Retransmission Timeout (RTO)

=> Based on expected RTT: "how long until you SHOULD get an ACK?"

When you get an ACK, update RTO

=> Smoothed weighted moving average of recent RTTs

X p

Example upper/lower bounds
RTOmin ~= 100ms
RTOmax ~= 5sec

Computing RTO

Strategy: measure expected RTT based on ACKs received

Use exponentially weighted moving average (EWMA)

• RFC793 version ("smoothed RTT"):

```
SRTT = (\alpha * SRTT_{Last}) + (1 - \alpha) * RTT_{Measured}
RTO = max(RTO<sub>Min</sub>, min(\beta * SRTT, RTO<sub>Max</sub>))
```

 α = "Smoothing factor": .8-.9

 β = "Delay variance factor": 1.3—2.0

 $RTO_{Min} = 1$ second

RFC793, Sec 3.7 RFC6298 (slightly more complicated, also measures variance)

UPDATE on perf requirement

Performance requirement: send/recv process MUST be event driven

- No busy-waiting
- time.Sleep MUST NOT BLOCK SEND/RECV process

*Okay to use sleep, time. Ticker to have separate thread trigger an event, like retransmissions

Where does this apply?

- REPL: s, r, sf, rf
- VRead/VWrite
- Deciding when to send, or check for new data
- Retransmissions

=> Channels, condition variables, etc. are your friends

Out of order segments

Usually, make a "early arrival queue"

- When segment arrives, add to queue if it's not the next segment
 => What to store? You decide!
- As more segments arrive, check the top of the queue to see if it fills in any gaps

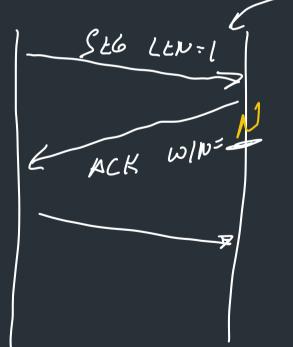
Zero window probing (ZwP)

When receiver's window is full, sender enters zero window probing mode

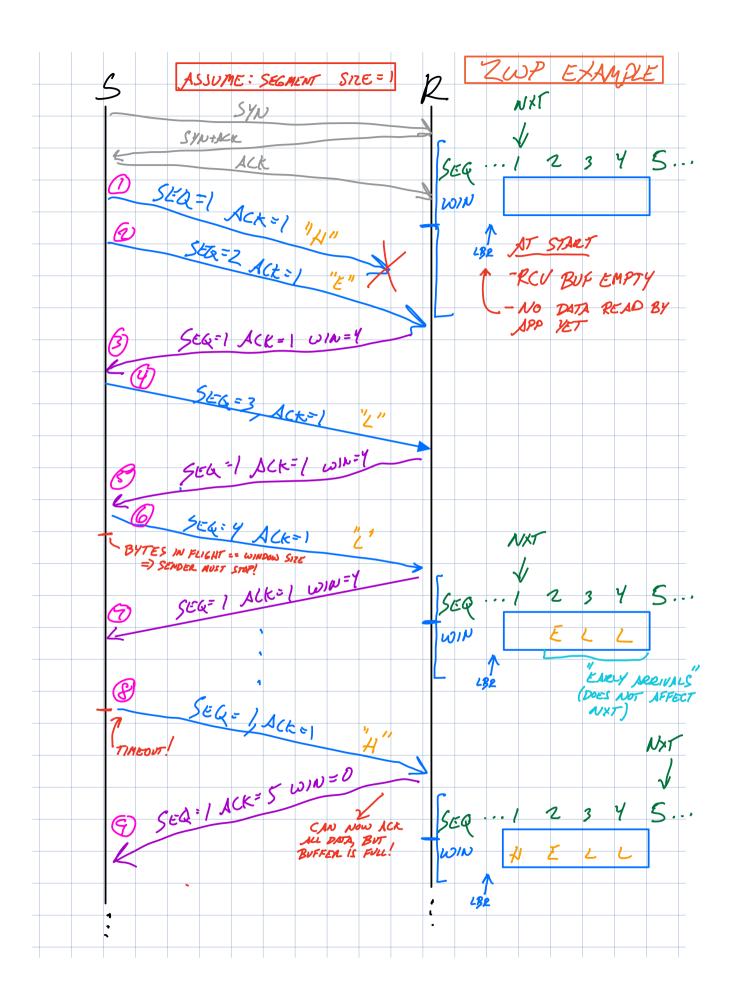
Stop sending segments

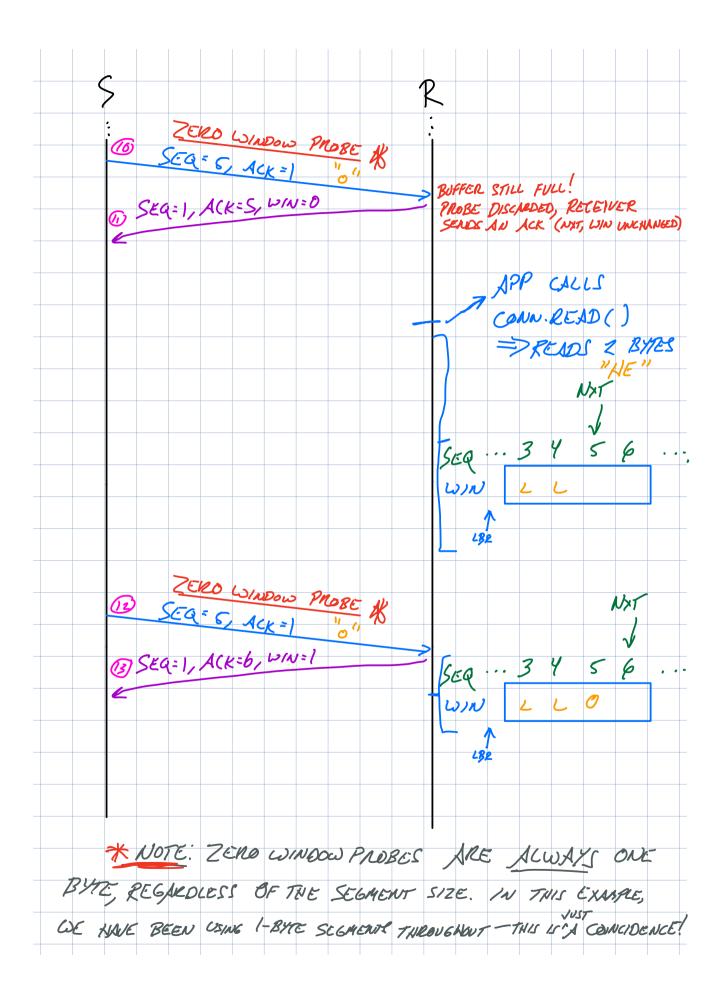
 At a periodic intervals, send 1 byte segments until receiver sends back window > 0 bytes

/BYTE OF REAL DATA, WHATEVER IS NEXT IN SOUP BUF.



								bin		
								liscu vork		
			ır bı							





Zero window probing

When receiver's window is full, sender enters zero window probing mode

- Stop sending segments
- At a periodic intervals, send 1 byte segments until receiver sends back window > 0 bytes

How to test?

- On one side, listen on a port: a 9999
- On other side, send a file

Connection teardown

4-way connection close process => see the lecture for details

- VClose just starts the connection close process
 - => TCB not deleted until connection goes to CLOSED state

Testing with packet loss

New REPL command in vrouter reference (out soon):

```
> drop 0.01  // Drop 1% of packets
> drop 0.5  // Drop 50% of packets (way too aggressive)

> drop 1  // Drop ALL packets (equivalent to "down")
> drop 0  // Drop no packets
```

Also: can set by running vrouter with --drop

Custom vnet_run configurations



<u>Useful wireshark mechanics</u>

- SEQ/ACK analysis
- Follow TCP stream
- Validating the checksum

Note: watching traffic in wireshark works differently in this project!

=> See Gearup II, "TCP getting started" guide for details

Reference implementation

- Our implementation of TCP
- Try it and compare with your version!

Note: we're using a new reference this year (after 8+ years!)

- We've tested as best we can, but there may be bugs
- See Ed FAQ, docs FAQ for list of known bugs
- Let us know if you have issues!

```
⇒ If the spec disagrees with the reference implementation,
the spec wins--don't propagate buggy behavior
(please help us find any discrepancies!)
```

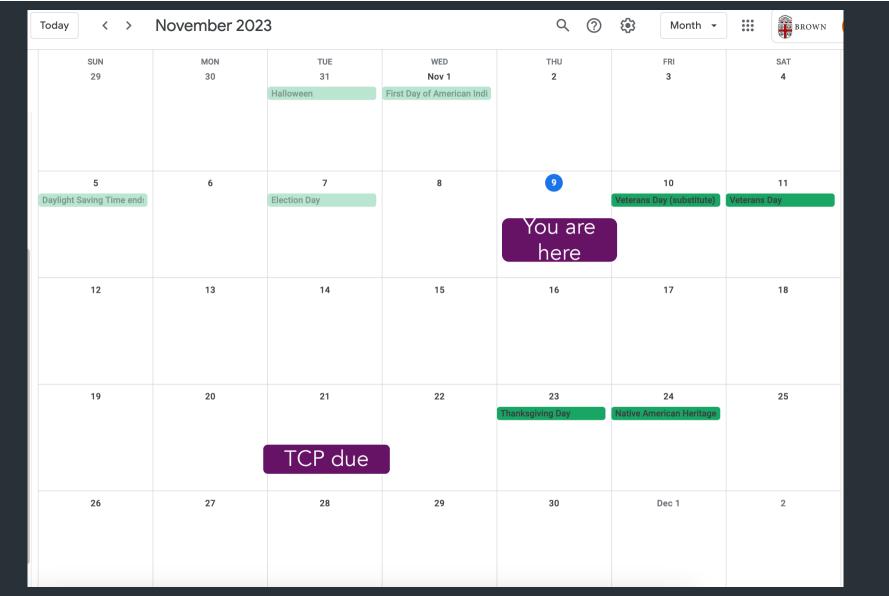
Closing thoughts

Do not underestimate these last parts--it will take time to debug and test them.

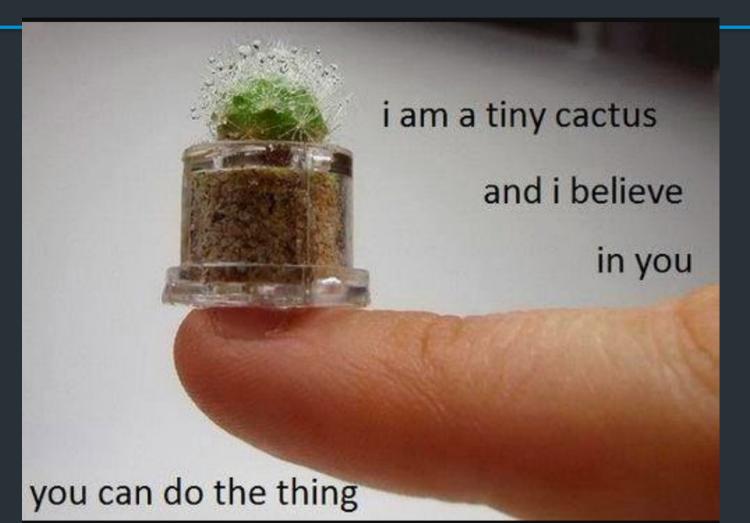
When stuck, take a break and come back to it. It will help.

=> <u>Do NOT wait until the last minute.</u>

Don't panic.



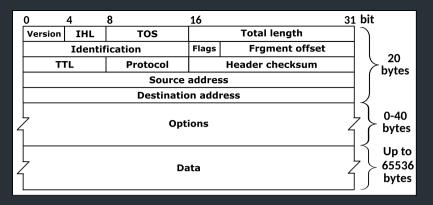
Breathe

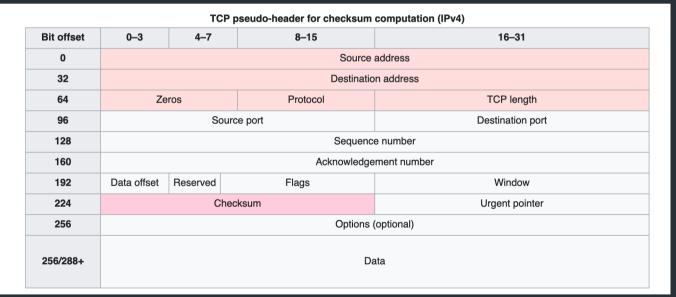


The TCP checksum

... is pretty weird

Computing the TCP checksum involves making a "pesudo-header" from TCP header + IP header fields:





 \Rightarrow See the TCP-in-IP example for a demo of how to compute/verify it

Where to get more info

