# COMP3310/6331 - #12

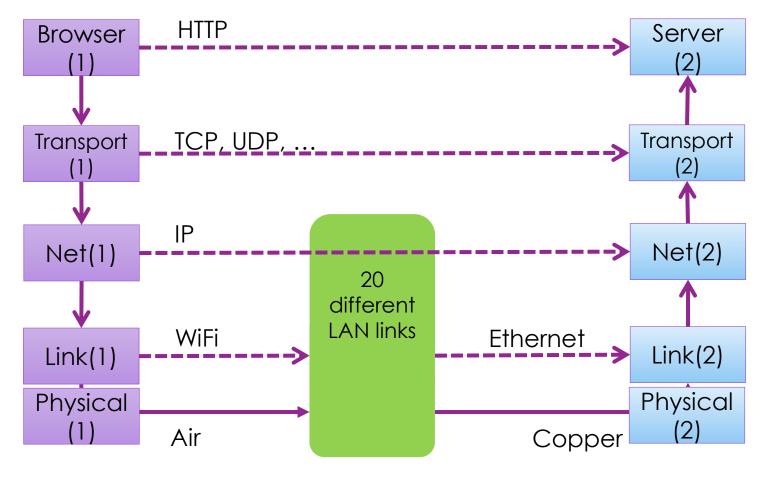
Transport Layer: TCP/UDP intro

<u>Dr Markus Buchhorn:</u> <u>markus.buchhorn@anu.edu.au</u>

#### Where are we?

 Moving further up Application Messages Presentation Session Segments Transport Network (IPv4, v6) **Packets** Link (Ethernet, WiFi, ...) Frames Physical Bits (Cables, Space and Bits)

## Ignore the network, focus on applications



Applications don't know nor care. Unless there is a performance question.

#### Getting into the transport layer

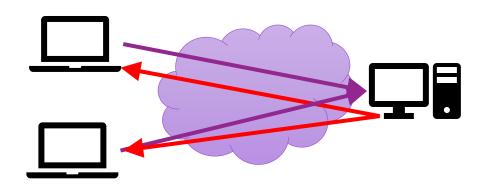
- Leave all the packet to-and-fro to the network layer
- Everything here is a payload for IP packets
  - A Segment



- Offers rich functionality (or not) to Applications
  - Reliability, performance, security, and other quality measures on unreliable IP
- Routers and other network devices do not get in the way
  - They (should) only look at 'the envelope' of a message, not the messages
  - This is pure host-to-host.

#### Simple client/server model

- Servers offer something,
- Clients connect
  - Send a request
  - Server replies



Servers can handle multiple clients

Model breaks in p2p applications – everyone is both.

#### Transport Services

- What <u>common</u> application needs are there?
- Main decision:
  - Reliable everything has to arrive bit-perfect.
    - Transport layer repairs packet loss, mis-ordering (and other damage)
    - I can wait!
  - Unreliable
    - Don't care about eventual perfection,
    - Do care about performance, simplicity, ...
- Two types of communication
  - Messages: self-contained command and response (post office)
  - Byte-stream: generic flow of bytes, chunked into segments (conversation)

#### Which does what?

	Unreliable	Reliable
Messages	UDP (datagrams)	
Byte-stream		TCP (Streams)

- Could have reliable messages but can build that on top of TCP
- Could have unreliable byte-streams but that looks like UDP

Transmission Control Protocol: TCP = IP Protocol 6
User Datagram Protocol: UDP = IP Protocol 17

ICMP = 1, IGMP = 2, IPv6 encapsulation = 41, 130+ more

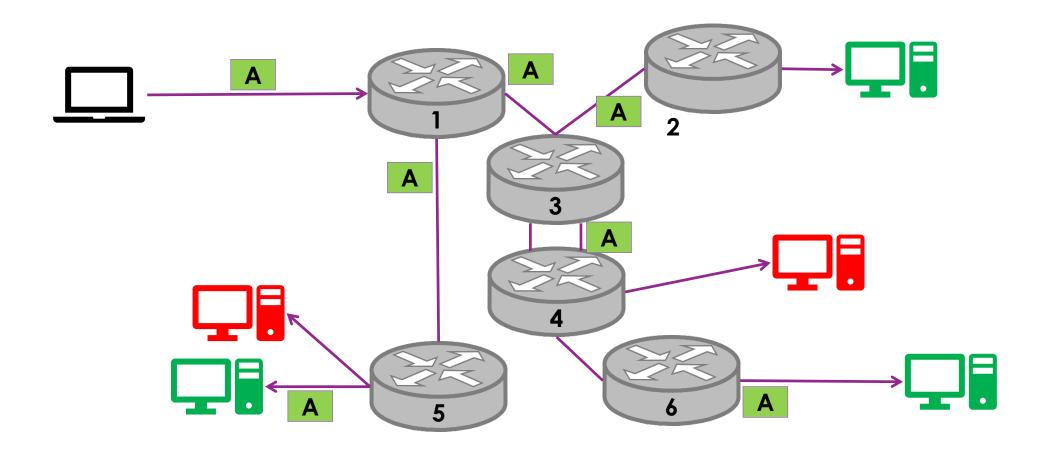
# Compare them

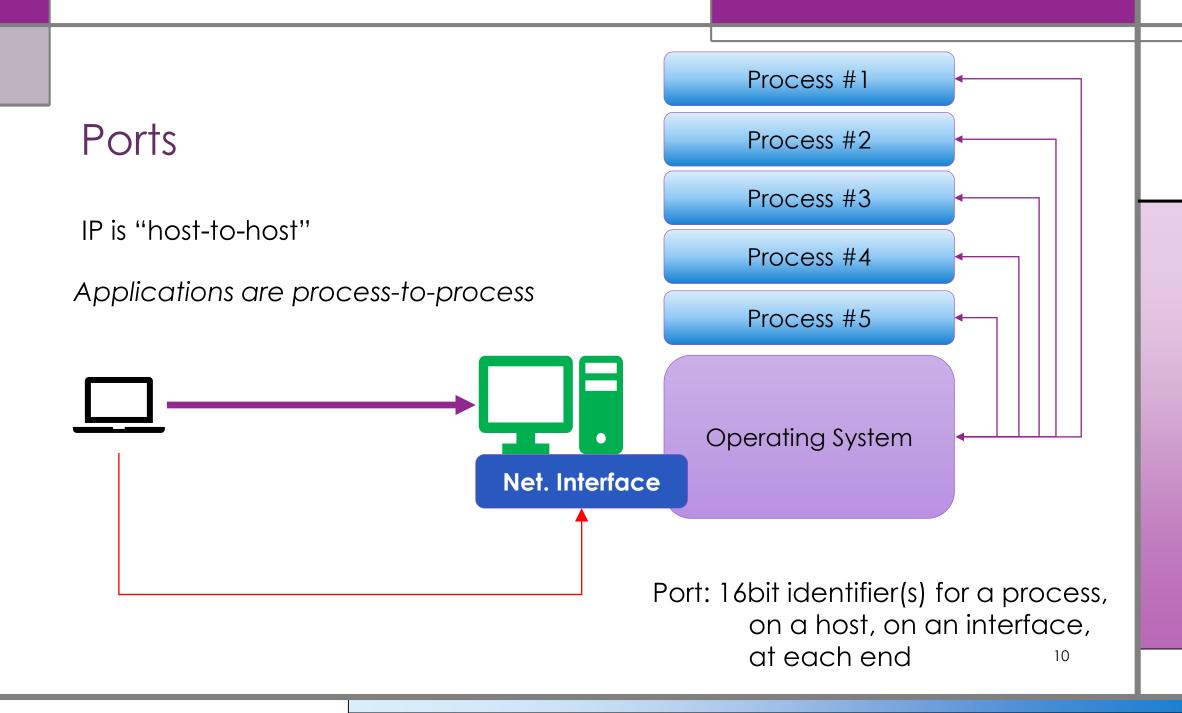
TCP	UDP
Connection-oriented (significant state in transport layer @host)	Connectionless (minimal state in transport layer)
Delivers BYTES: once, reliably, in order (to your process)	Delivers MESSAGES: 0-n times, any order
Any number of bytes (in a stream)	Fixed message size
Flow control (sender/receiver negotiate)	Don't care
Congestion control (sender/network negotiate)	Don't care

- UDP is an enhanced IP packet
- TCP is a lifestyle choice many features

#### IP Multicast: UDP

Connectionless, maybe time-sensitive Replica packets are fine!





#### Well-known (and other) ports

- https://www.iana.org/assignments/service-names-port-numbers/service-names-port-numbers.xhtml
- Opening ports below 1024 requires extra privileges

20,21	ftp	File transfer	
22	ssh	Secure shell	
25	smtp	Email – outbound	
80	http	Web	
110	pop3	Email – inbound	
143	imap	Email – inbound	
443	https	Secure-Web	

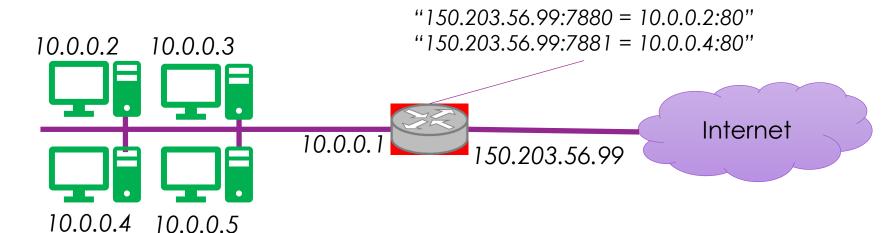
#### A Port is just a start

- Inetd/xinetd
  - Don't continually run every server-service somebody may eventually talk to
  - Single service, launch appropriate service on demand
  - Listens to all (registered) ports and protocols (tcp, udp)
  - Spawns the service to have the conversation

- Port mapping
  - (e.g. remote procedure calls, bittorrent, ...)
  - Listen on a well-known port
  - Accept connections
  - Redirect them to a spawned service on another port
    - Services can register with the portmapper

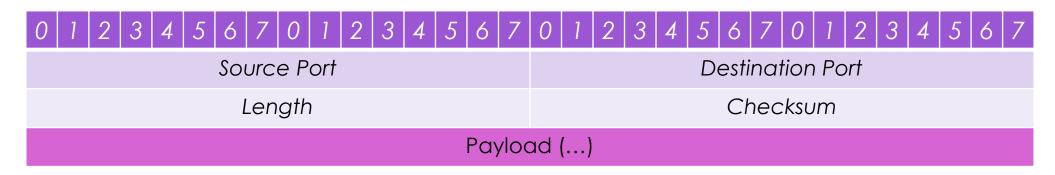
#### NAT is actually NAPT

- NAT has everyone 'hiding' behind a single public IP address
- But everyone wants access to/from the Internet at the same time
- So translate addresses and ports
- Router maintains a table
  - Dynamically for outbound. Can be static for inbound.



UDP





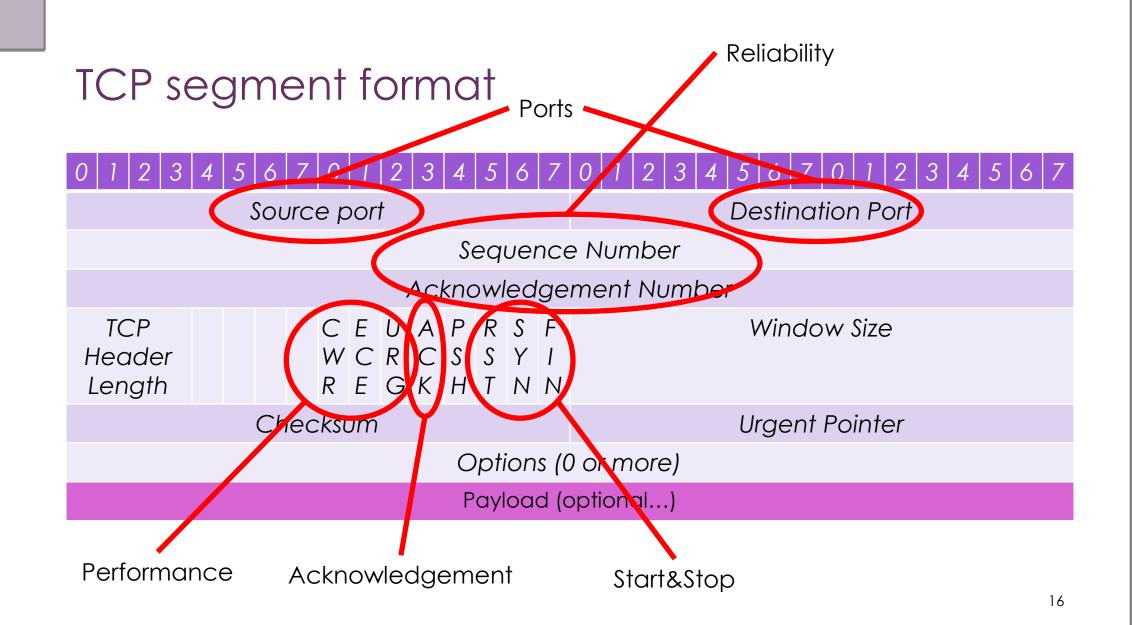
- UDP adds to IP: Ports, payload length and a Checksum
- And nothing else...

#### Byte-streams

- TCP segments carry chunks of a byte-stream
  - "Message" boundaries are not preserved
- Sender packetises (eventually) on write()
  - Multiple writes can be one packet and vice-versa buffer dependent



- Receiver unpacks
  - Applications read() a stream of bytes
- Hence: <u>Segments</u>



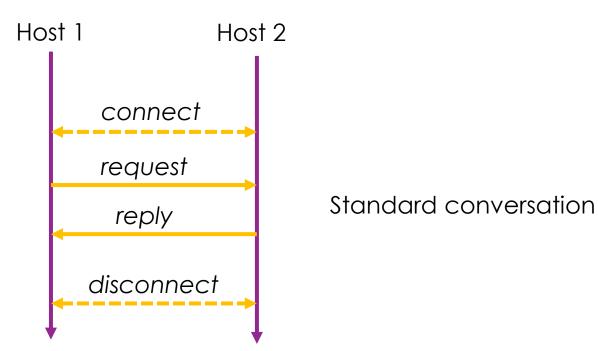
#### TCP Options

These actually get used...

- <u>Maximum Segment Size</u>: how much each end is willing to take
- Window Scale: When 64kB is not enough multiply
- <u>Timestamp</u>: For computing rtt and expanding sequence number space
- <u>Selective Acknowledgement</u>: Like ACK, but better.

#### Programming connections

- "Socket" programming an address, a port, and a need to communicate
- Connections are identified in the Operating System by a '5-tuple'
  - source/destination ip, source/destination port, protocol

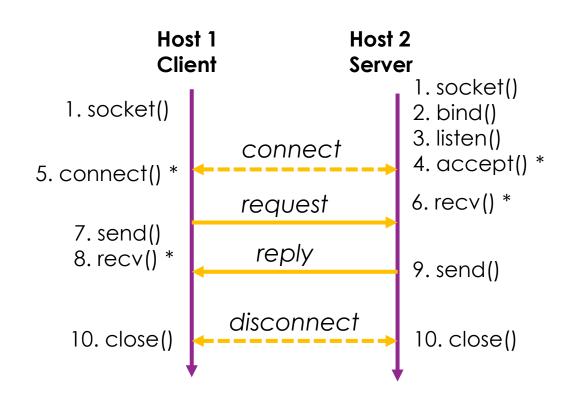


#### Socket API

Primitive (function)	What it does	
SOCKET	Create an object/descriptor	
BIND	Attach a local address and port	
LISTEN (tcp)	Tell network layer to get ready	
ACCEPT (tcp)	Be ready!	
CONNECT (tcp)	Connect	
SEND(tcp) or SENDTO(udp)	Send	
RECEIVE(tcp) or RECEIVEFROM(udp)	Receive	
CLOSE	Release the connection/socket	

#### So...

- Server needs to be prepared for connections
- Client initiates the connection



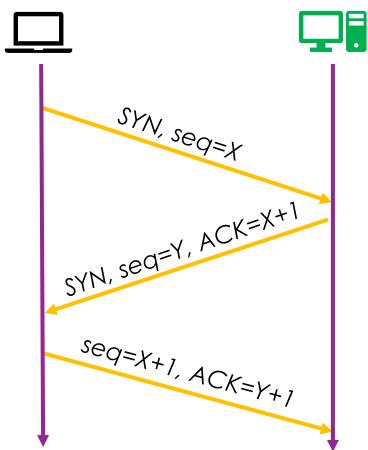
\* = call blocks

#### TCP and reliability

- TCP is a reliable, bidirectional byte-stream
  - Uses <u>Sequence Numbers</u> and <u>Acknowledgements</u> to provide reliability
  - Piggybacks control information on data segments in reverse direction
    - If there's no data, just sends feedback
- <u>Sequence numbers:</u> N-bit counter that wraps (e.g. ..., 253, 254, 255, 0, 1, 2...)
  - Byte count (pointer) in a stream a cumulative ACK
  - Can wrap quickly on high-speed links ( $2^{32} = 4GB$ ) can use timestamps too
  - Does not start from zero (for security)
- Acknowledgements: Which bytes have been received/is expected

# Getting connected – 3 way handshake

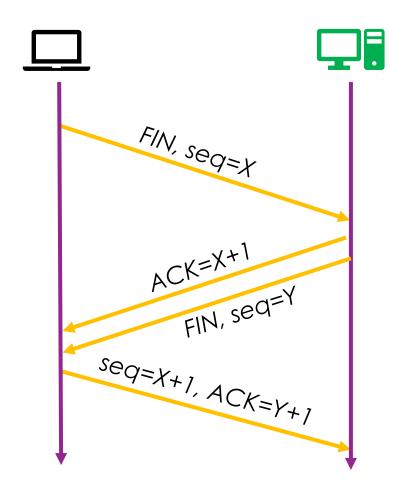
- TCP is full-duplex = two simplex paths
  - Both need to start together(\*)
    - Synchronise Sequence numbers in both directions
- Connecting
  - Receiving transport stack decides:
    - anybody listen()ing on that port?
      - If not, ReSeT
      - If yes, passed to receiving process listen()ing,
  - Transport stack ACKnowledges
  - Originator ACKs that SYN/ACK
    - and off they go



## Hanging up

- Both need to end together
  - Ideally...
  - Time to flush buffers

- Disconnecting
  - One side initiates close()
  - Triggers a FIN(alise)
  - Other side ACKs and FINs too
- And if FIN is lost? Resend…



#### Socket states:

State	Description		
LISTEN	Accepting connections		
ESTABLISHED	Connection up and passing data		
SYN_SENT	Waiting for reply from remote endpoint		
SYN_RECV	Session requested by remote, for a listen()ing socket		
LAST_ACK	Closed; remote shut down; waiting for a final ACK		
CLOSE_WAIT	Remote shut down; kernel waiting for application to close() socket		
TIME_WAIT	Socket is waiting after close() for any packets left on the network		
CLOSED	Socket is being cleared		
CLOSING	Our socket shut; remote shut; not all data has been ACK'ed		
FIN_WAIT1	We sent FIN, waiting on ACK		
FIN_WAIT2	We sent FIN, got ACK, waiting on their FIN		

#### % netstat -n

What's happening on my machine?

```
192.168.178.34:5353
       192.168.178.34:52848
       192.168.178.34:61842
UDP
       [::]:123
       [::]:3702
UDP
UDP
       [::]:3702
       [::]:3702
UDP
UDP
       [::]:3702
UDP
       [::]:3702
UDP
       [::]:3702
       [::]:5353
UDP
UDP
       [::]:5353
UDP
       [::]:5355
UDP
       [::]:49533
UDP
       [::]:52366
UDP
       [::]:54902
       [::]:61002
UDP
       [::]:65404
UDP
       [::1]:1900
UDP
       [::1]:61841
UDP
       [fe80::81c:86ce:5a66:a430%15]:546 *:*
UDP
UDP
       [fe80::81c:86ce:5a66:a430%15]:1900 *:*
```

Proto	Local Address	Foreign Address	State
TCP	0.0.0.0:135	0.0.0.0:0	LISTENING
TCP	0.0.0.0:445	0.0.0.0:0	LISTENING
TCP	0.0.0.0:5040	0.0.0.0:0	LISTENING
TCP	0.0.0.0:5357	0.0.0.0:0	LISTENING
TCP	0.0.0.0:7969	0.0.0.0:0	LISTENING
TCP	0.0.0.0:8501	0.0.0.0:0	LISTENING
TCP	0.0.0.0:8502	0.0.0.0:0	LISTENING
TCP	0.0.0.0:8866	0.0.0.0:0	LISTENING
TCP	0.0.0.0:8968	0.0.0.0:0	LISTENING
TCP	0.0.0.0:9330	0.0.0.0:0	LISTENING
TCP	192.168.178.34:49369	104.24.126.250:443	ESTABLISHED
TCP	192.168.178.34:49371	162.125.83.7:443	CLOSE WAIT
TCP	192.168.178.34:49379	52.98.0.194:443	TIME WAIT
TCP	192.168.178.34:49386	52.114.76.34:443	TIME_WAIT
TCP	192.168.178.34:49388	52.98.0.194:443	TIME_WAIT
TCP	192.168.178.34:49390	13.107.3.128:443	TIME_WAIT
TCP	192.168.178.34:49393	162.125.83.3:443	CLOSE_WAIT
TCP	192.168.178.34:49396	52.5.51.106:443	ESTABLISHED
TCP	192.168.178.34:49398	192.168.178.1:49000	TIME_WAIT
TCP	192.168.178.34:49404	40.100.146.18:443	ESTABLISHED
TCP	192.168.178.34:49405	52.98.5.226:443	ESTABLISHED
TCP	192.168.178.34:49409	104.98.4.162:80	ESTABLISHED
TCP	192.168.178.34:49411	54.154.198.3:443	ESTABLISHED
TCP	192.168.178.34:49412	162.125.34.137:443	ESTABLISHED
TCP	192.168.178.34:50471	52.230.7.59:443	ESTABLISHED
TCP	192.168.178.34:53421	104.154.164.197:443	ESTABLISHED
TCP	192.168.178.34:53451	74.125.68.125:5222	ESTABLISHED
TCP	192.168.178.34:53568	52.98.0.34:443	ESTABLISHED
TCP	192.168.178.34:53591	34.226.253.48:443	ESTABLISHED
TCP	192.168.178.34:54258	192.168.178.44:445	ESTABLISHED
TCP	192.168.178.34:54451	54.84.185.96:443	ESTABLISHED
TCP	192.168.178.34:56177	35.229.34.229:443	ESTABLISHED
TCP	192.168.178.34:61528	40.100.146.18:443	ESTABLISHED
TCP	192.168.178.34:62647	13.112.202.196:443	ESTABLISHED
TCP	192.168.178.34:64245	216.58.200.106:443	CLOSE_WAIT
TCP	192.168.178.34:64328	172.217.167.74:443	CLOSE_WAIT
TCP	192.168.178.34:65180	162.125.34.129:443	ESTABLISHED
TCP	192.168.178.34:65439	52.98.0.194:443	TIME_WAIT
TCP	192.168.178.34:65496	162.125.34.129:443	ESTABLISHED
TCP	192.168.178.34:65503	52.98.0.146:443	ESTABLISHED
TCP	192.168.178.34:65511	52.98.0.178:443	ESTABLISHED

#### TCP Sliding Windows

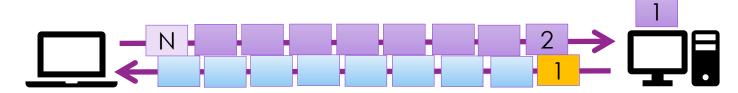
- Want reliability and throughput (of course!)
- Start with ARQ stop-and-wait
  - Single segment outstanding = problem on high bandwidth\*delay networks



- Say one-way-delay=50ms so round-trip-time (RTT)=2d=100ms
- Single segment per RTT = 10 packets/s
  - Typical packet ? Say 1000 bytes =  $\sim$ 10,000 bits -> 100kb/s
- Even if bandwidth goes up, throughput doesn't!

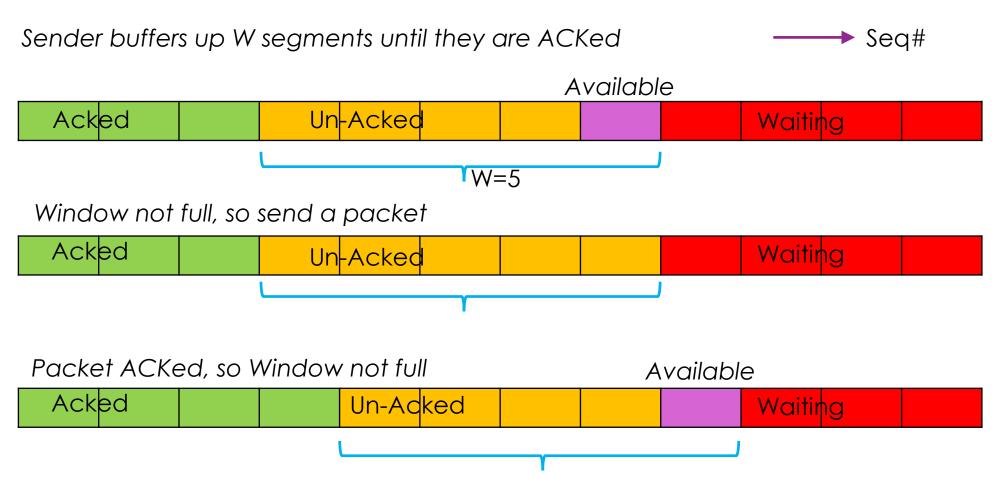
#### TCP Sliding Windows

- Allow W segments to be 'outstanding' (unACKed) per RTT
  - Fill a pipeline/conveyor-belt with segments



- Set up a 'window' of W segments
- W=2\*Bandwidth\*delay
- At 100Mb/s, delay=50ms means <u>W=10Mb</u>
  - Assuming same 10kb segments, W=1000 segments
  - 500 are out there somewhere!

## Sliding Window approach



#### If(lost) then: ARQ – "Go Back N"

- Receiver buffers just a single segment
- If it's the next one in sequence, ACK it, everyone happy
- If it's not, drop it,

- 1 2 3 4 5
- Let sender retransmit what I'm actually waiting for

- Sender has a single timer. After timeout, resend (all) from (first) ACK-less.
- Really simple, but somewhat inefficient

#### ARQ – "Selective Repeat"

- Receiver buffers many segments
  - Reduce retransmissions
- ACK what has been received in order
- And also ACK received segments that aren't
  - Any gaps indicates missing segment!
  - Selective ACK (SACK)
  - TCP header has an ACK flag (1bit), and a SACK Option (32bits...)
  - 3 duplicate ACKs (plus SACKs) trigger resend
- Sender has a timer per unACKed-segment
  - As each timer expires, resend that segment
- Cope with (some) misordering. Way more efficient, now widespread

#### Everybody runs the same TCP...?

No. There is no single TCP stack

- Many years of various optimisations, experiments, algorithms, ...
  - Suited to various circumstances
  - And as vulnerabilities have been found and mitigated (and found and ...)

Doesn't impact the network, only hosts, so you can do what you want...