Computer Organization and Networks

Chapter 9: Networking II

Winter 2022/2023



IPv6

- Internet Protocol, version 6
- Successor to IPv4

- Not natively interoperable with IPv4
 - IPv4-only devices cannot communicate with IPv6-only devices
 - Most modern devices implement both IPv4 and IPv6
 - Eventually, IPv4 will be phased out...

IPv6 addressing

```
Ethernet adapter Ethernet:
Link-local IPv6 Address . . . . : fe80::10e5:f700:f6ab:7afc
```

- 128-bit address
 - Notation: 16-bit hexadecimal blocks separated by colons (:)
 - Zero blocks can be omitted using double colon (::)
 - fe80::10e5:f700:f6ab:7afc is the same as
 fe 80 00 00 00 00 00 10 e5 f7 00 f6 ab 7a fc

IPv6 addressing

```
Ethernet adapter Ethernet:

Link-local IPv6 Address . . . . : fe80::10e5:f700:f6ab:7afc
```

- 64-bit network prefix, 64-bit interface identifier
- A single interface (e.g.: a network card) may have multiple addresses
 - Addresses share the *interface identifier*
- Addresses have a scope in which they are valid

IPv6 scoping

- Global addresses
 - Valid in any network connected to the internet
 - May be routed on the public internet
- Unique-local addresses (in **fc00::/7**)
 - Same idea as IPv4 private networks
 - No assignment/registration needed
 - Routed only in local networks, but not on the public internet
- Link-local addresses (in **fe80**::/64)
 - Only valid within the Link Layer network

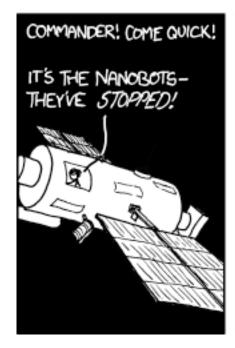
IPv6 packet overview

- Similar fields to IPv4 packets
 - Version is always 0110 (version 6)
 - Length, Source and Destination fields
 - Optional extension header blocks
- Header checksum removed
 - Relies on Link Layer to provide error detection
- Fragmentation (mostly) removed
 - No fragmentation by routers
 - Fragmentation by hosts only as an extension
 - Transport Layer is expected to perform fragmentation

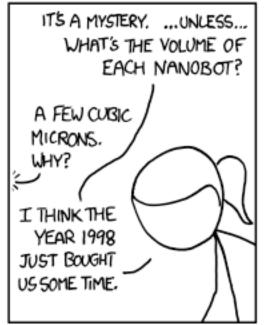
Offsets	Octet	0							1									2							3								
Octet	Bit	0	1	2	3	4	5	6	7	8	9	10	11	12	13	14	1	5 16	17	18	19	20	21	22	23	24	25	26	27	28	29	30	31
0	0	Version Traffic class Flow label																															
4	32	Payload length Next header Hop lim														t																	
8	64																																
12	96		Source address																														
16	128																																
20	160																																
24	192																																
28	224														r)oeti	inat	ion ac	Idros														
32	256														L	Jesu	IIai	ion ac	lui es	00													
36	288																																

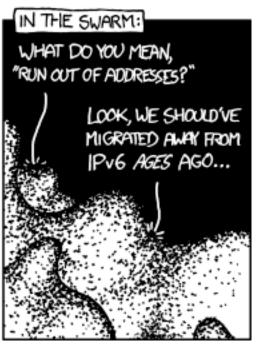
IPv6 recap

- Successor to IPv4
 - "Permanent" solution to IP address exhaustion
 - We'll talk about IPv4 workarounds in a bit!
 - Some protocol-level improvements
 - Not interoperable with IPv4
- Supported by most modern end-user devices
 - Server-side support is... still lacking [https://ipv6.watch]
- 128-bit addresses (64-bit network part, 64-bit interface identifier)
 - 2⁶⁴ networks, each consisting of 2⁶⁴ hosts











The Transport Layer

The Transport Layer

- Computers A and B are capable of sending data to each other
- Goal: Allow multiple applications to communicate reliably

Concerns:

- How to distinguish which application data is meant for? (multiplexing)
- What if data is lost on the lower layers? (reliability)
- How much data can the network handle? (congestion control)
- How much data can the receiver handle? (flow control)

The Transport Layer

- The internet has two widely-used protocols at the Transport Layer:
 - <u>Transmission</u> <u>Control</u> <u>Protocol</u>
 - Focused on reliable delivery
 - Connection-based
 - <u>U</u>ser <u>D</u>atagram <u>P</u>rotocol
 - Focused on speed
 - Connectionless

The Transport Layer: Ports

- Concept used for both TCP and UDP
- Source and destination identified by port number
 - 16 bits (65536 available ports)
 - TCP and UDP ports are separate
 - The protocols implement the same idea, but each only cares about its own ports...
- Common notation: Port number after IP address
 - 127.0.0.1:8000 is port 8000 at host 127.0.0.1
 - [::1]:8000 is port 8000 at host ::1

UDP

- Fire-and-forget transmission of single datagrams
 - Useful for real-time applications
- Data may never arrive, may arrive out of order, ...
 - Data loss must be tolerable for the upper-layer application
- Extremely simple and straightforward

UDP datagram header Offsets Octet 0 1 2 3 Octet Bit 0 1 2 3 0 0 Source port Destination port Length Checksum

TCP

- Highly reliable transmission of a byte stream
 - Acknowledgments and re-transmission
 - Guaranteed to maintain data ordering
- Non-trivial protocol overhead
 - Still better than re-inventing the wheel if you need it!

TCP

• TCP connections have two sides: server and client

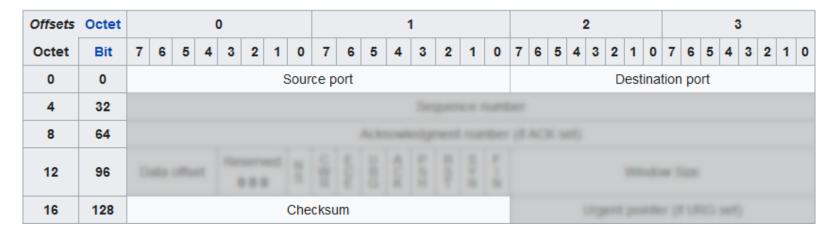
- Server listens on a specific port
 - Server port is fixed for all connections
- Client connects to that port on the server
 - Client uses a "random" ephemeral port, different for each connection
 - See for yourself: netstat -onb (Win) or netstat -tnap (Linux, Mac)
- Connections are uniquely identified by client IP + client port

The Transport Layer: Ports 2

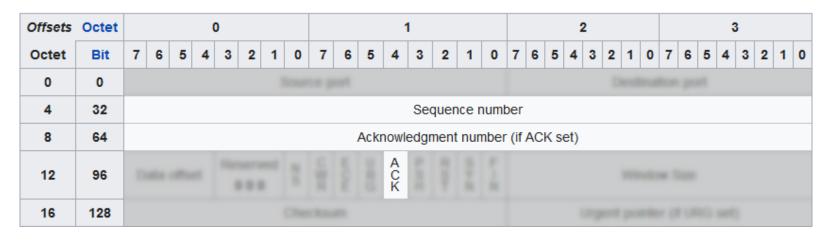
- Two applications can't use the same port number
- Client needs to know which port number to connect to

- Port numbers are standardized by IANA
 - 0–1023: well-known ports
 - Examples: 22 (SSH), 80 (HTTP), 123 (NTP), 194 (IRC), 443 (HTTPS), ...
 - 1024–49151: *registered* ports
 - Most server applications will use this range (even unregistered ones...)
 - 49152–65535: *dynamic* ports
 - Most OS will use this range for ephemeral (client) ports

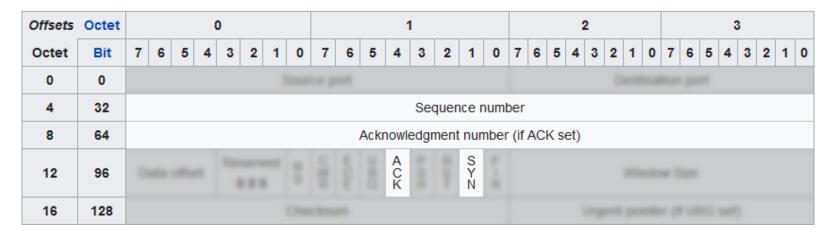
TCP packet overview



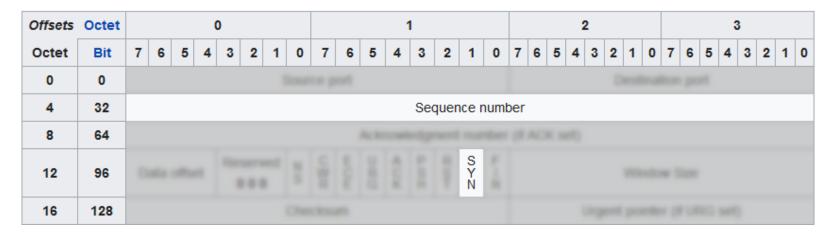
- Source + destination ports allow identification of connection
- Checksum over entire header + data



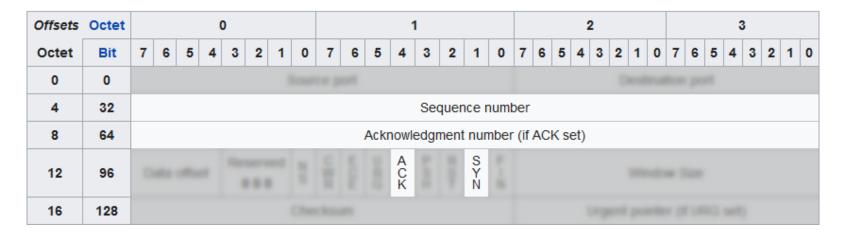
- TCP maintains a sequence number across the entire connection
 - Separate number for each end's packets
- Receipt of contiguous data confirmed via acknowledgment number
 - Acknowledgement number := next expected sequence number
- This allows ordering of data and re-sending of lost packets!



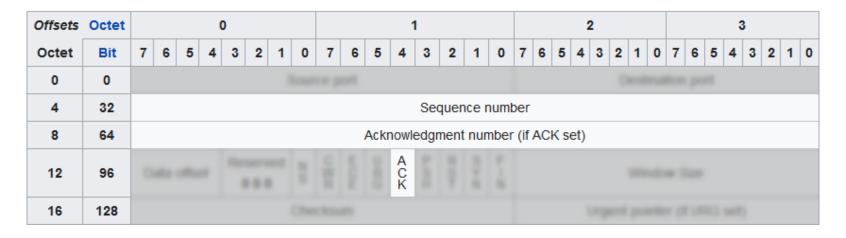
- Connection establishment: Three-way handshake
 - Client -> Server: SYN
 - Server -> Client: SYN + ACK
 - Client -> Server: ACK



- Client -> Server: SYN
 - Sequence number: **seq_c**, chosen randomly



- Server -> Client: SYN + ACK
 - Sequence number: **seq_s**, chosen randomly
 - Acknowledgement: **seq_c+1**

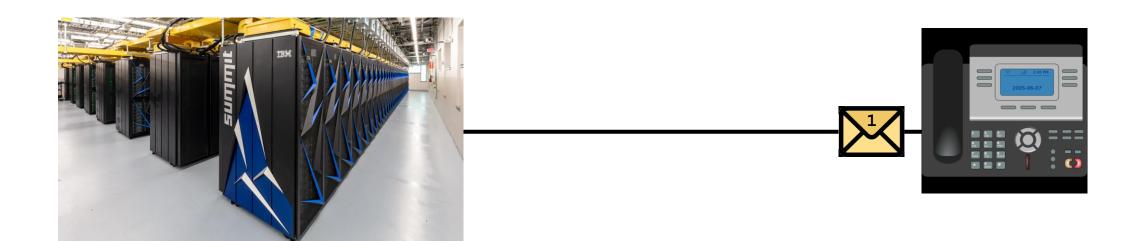


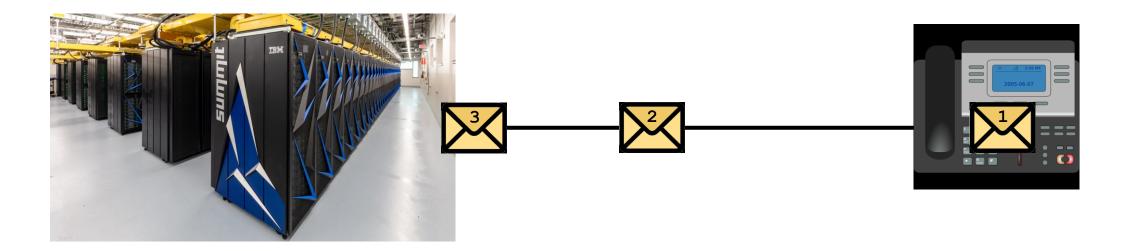
Client -> Server: ACK

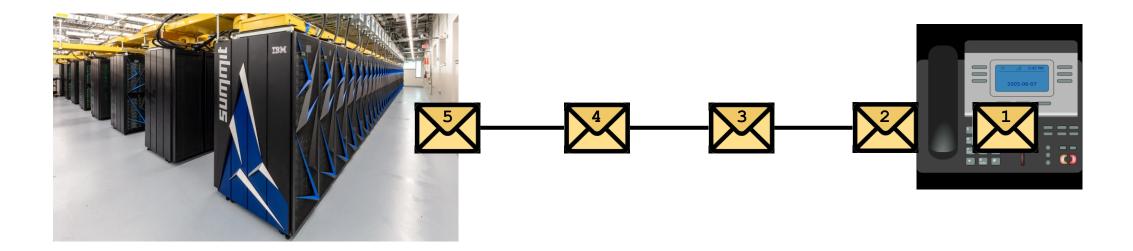
• Sequence number: seq c+1

Acknowledgement: seq_s+1

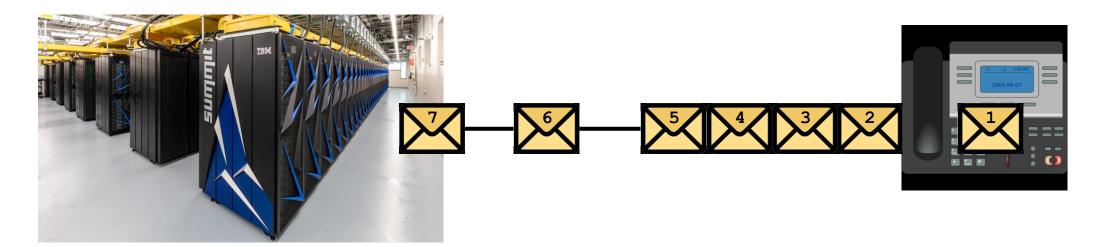
- Now both sides know that the other side has their sequence number
 - Ready to communicate!



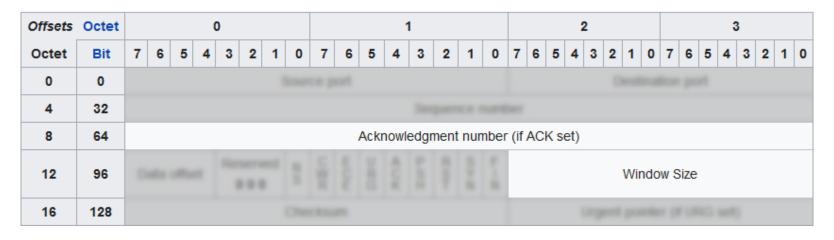




• Imagine: a supercomputer talking to a desk phone via a 100Gbps link

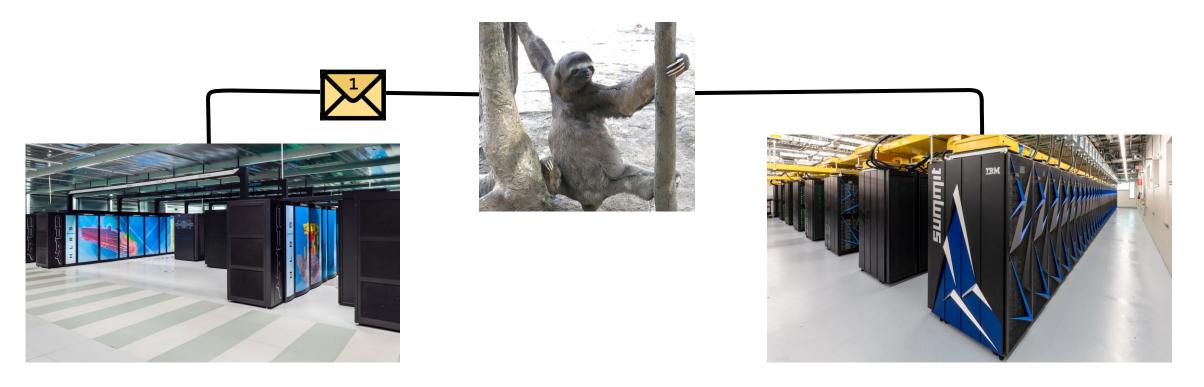


The desk phone doesn't stand a chance to keep up!

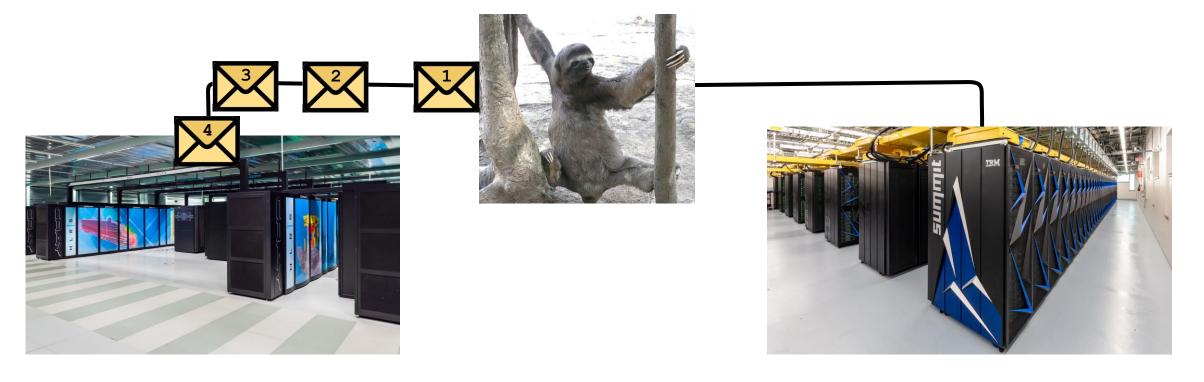


- Window size indicates how much more data the host can handle
- The other end must throttle its transmission rate to accommodate
 - Window size is relative to the last ACK'd packet

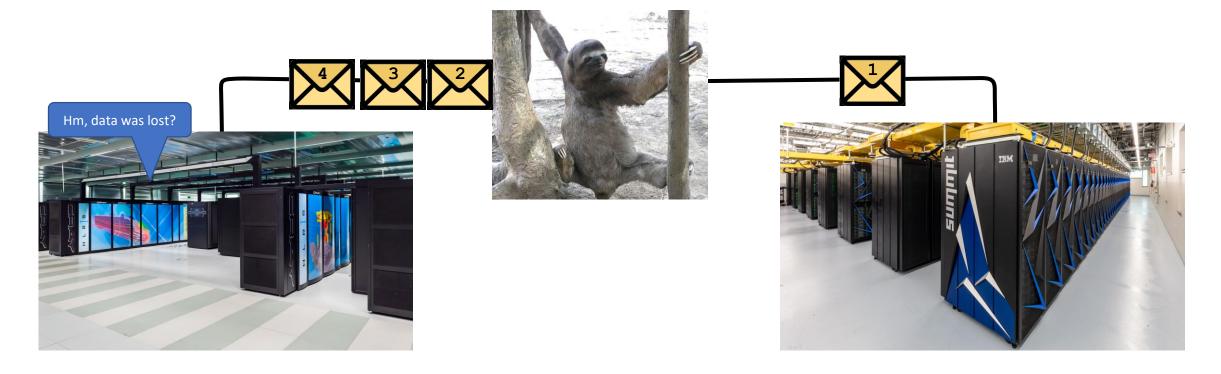
- Imagine: two supercomputers talking via a dial-up connection
 - Keep in mind: the "dial-up connection" could be some intermediate network



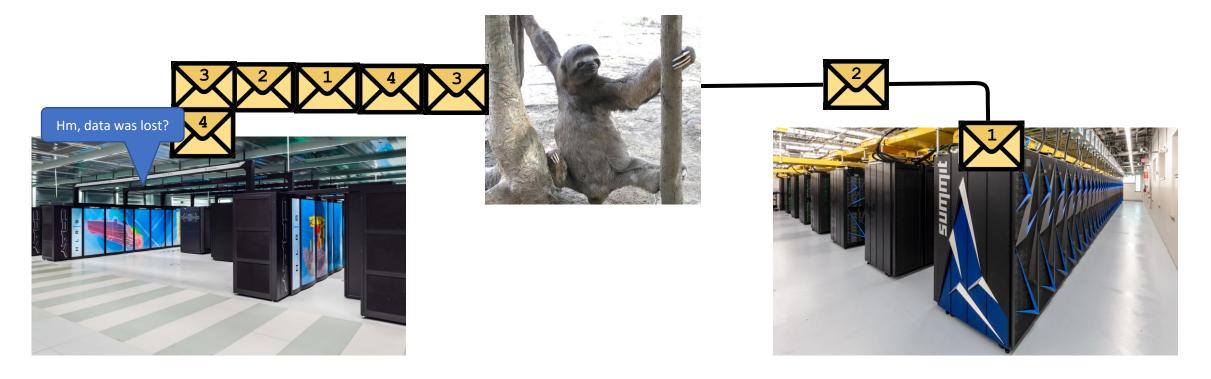
- Imagine: two supercomputers talking via a dial-up connection
 - Keep in mind: the "dial-up connection" could be some intermediate network



- Imagine: two supercomputers talking via a dial-up connection
 - Keep in mind: the "dial-up connection" could be some intermediate network



- Imagine: two supercomputers talking via a dial-up connection
 - Keep in mind: the "dial-up connection" could be some intermediate network



- Imagine: two supercomputers talking via a dial-up connection
 - Keep in mind: the "dial-up connection" could be some intermediate network

- If you just keep shoving data...
 - ... it will get stuck in a queue somewhere ...
 - ... so you think it was lost and send it again ...
 - ... now your queue is twice the size ...
 - ... and nothing useful gets done.

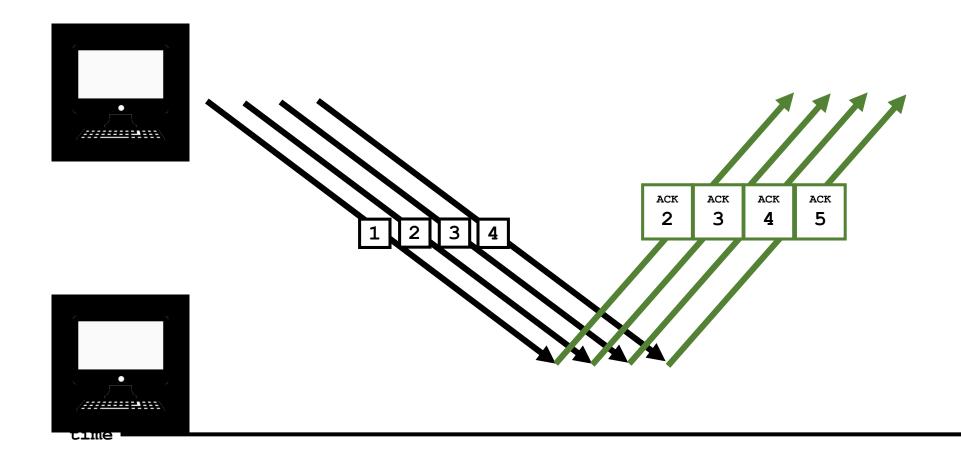
How do we avoid that?

- Each side throttles its data transmission rate independently
 - No cooperation required
 - Different OS have different algorithms

- Basic concept:
 - Start at a relatively slow rate, then increase speed until data gets lost
 - Once data is lost, assume we overloaded the connection and slow down again

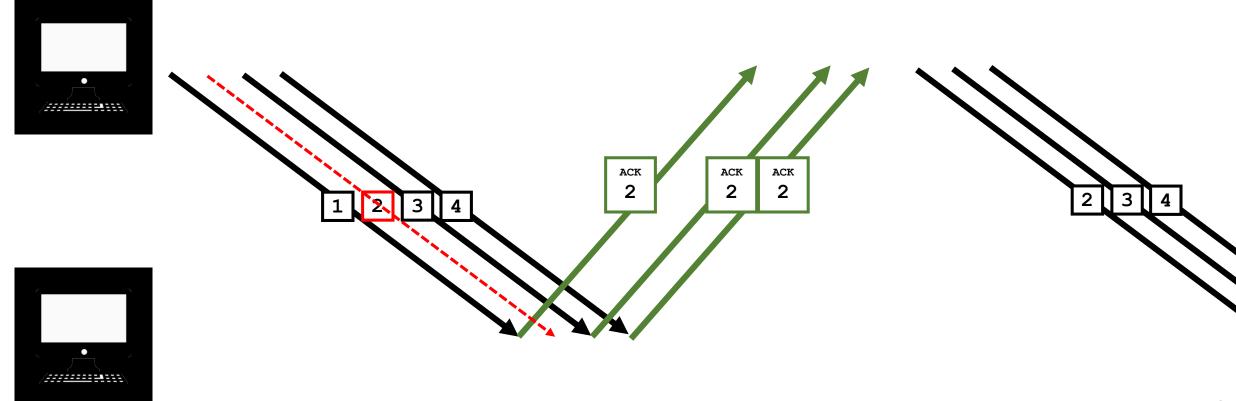
Details differ from OS to OS

TCP: <u>Selective Ack</u>nowledgment



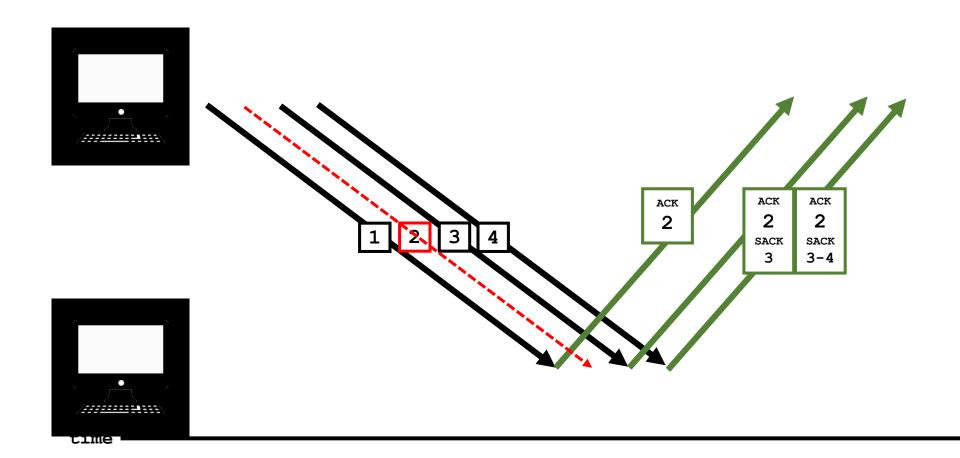
TCP: <u>Selective Ack</u>nowledgment

- "Standard" TCP does not deal with packet loss efficiently
 - Superfluous data is re-sent, wasting time and bandwidth!



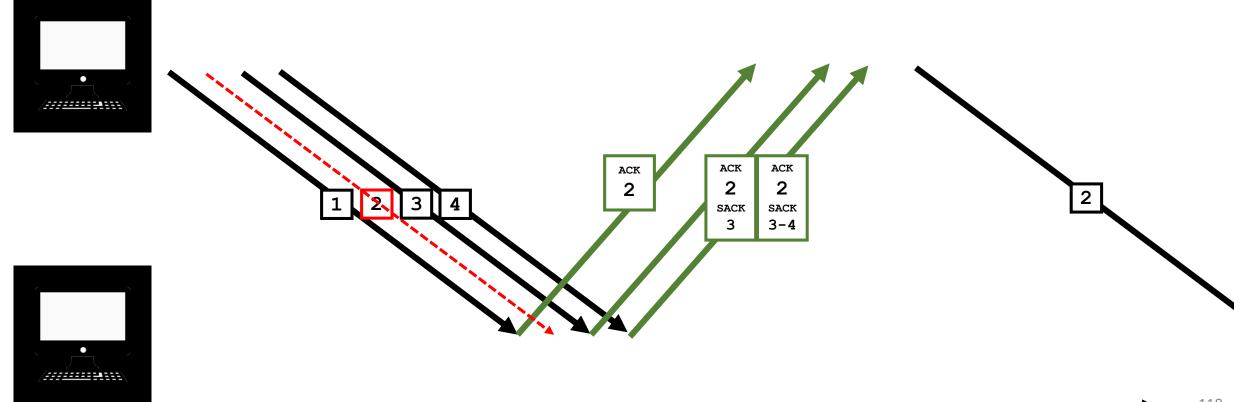
TCP: <u>Selective Ack</u>nowledgment

The SACK extension lets the recipient acknowledge further ranges



TCP: <u>Selective Ack</u>nowledgment

- The SACK extension lets the recipient acknowledge further ranges
 - These ranges do not need to be re-sent!



Transport Layer recap

- Two main protocols: TCP and UDP
 - TCP: highly reliable, but comes with overhead
 - UDP: low overhead, but no reliability guarantees
- Port numbers identify target application
 - By convention, low port numbers (0–1023) are reserved for specific services
 - 1024–49151 are used by other servers
 - 49152–65535 are used for ephemeral ports

TCP recap

- Client establishes connection to Server
 - Server listens on a pre-agreed port
 - Client uses a "random" port (49152–65535)
- Sequence numbers and acknowledgement numbers
 - Client and server have separate counters
 - Acknowledgement of received data using the other side's counter
 - Re-ordering and re-sending if necessary

TCP recap

- Flow Control protects the recipient
 - Recipient advertises its capacity
 - Sender has to abide by it
- Congestion Control protects the network
 - Transmission rate is gradually increased
 - Throttled back if packet loss is detected
 - Each side handles this independently
 - Details differ from OS to OS



The Network Layer

(again?)

Recap: IPv4 address exhaustion

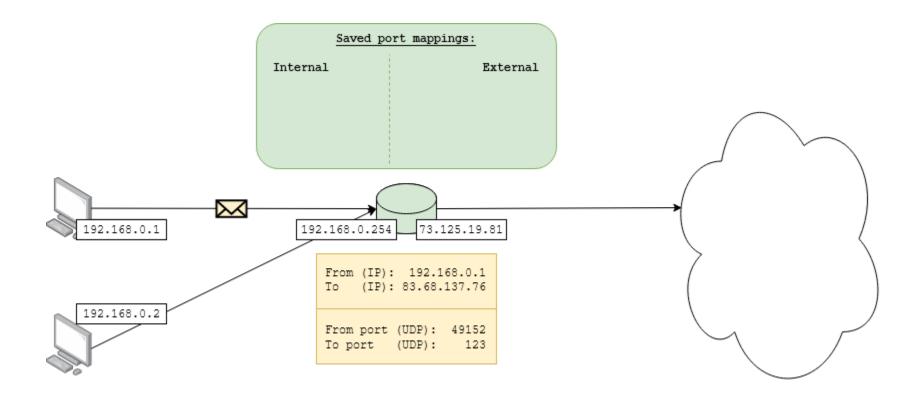
- IPv4 addresses are 32 bits long
 - 2³² is about 4 billion
- Every Internet-enabled device needs an address to communicate
 - There are a lot of devices

The internet is (mostly) out of IPv4 addresses!

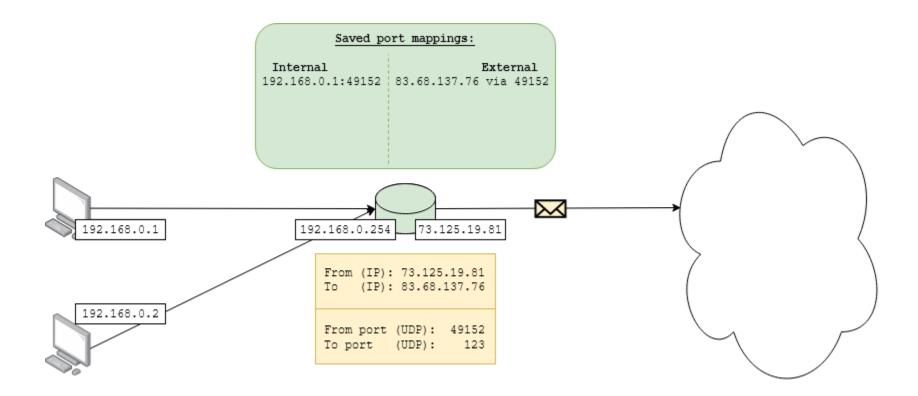
- "Hide" an entire private network behind a single public IP
 - Rewrite IP packets at the boundary

- Also known as:
 - "PAT"
 - Network Address Translation ("NAT")
 - Network Address and Port Translation ("NAPT")
 - NAT overloading
 - IP masquerading

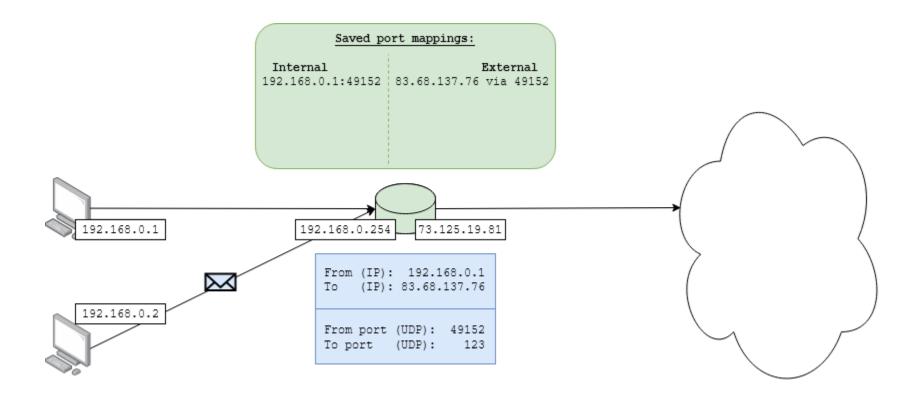
- "Hide" an entire private network behind a single public IP
 - Rewrite IP packets at the boundary



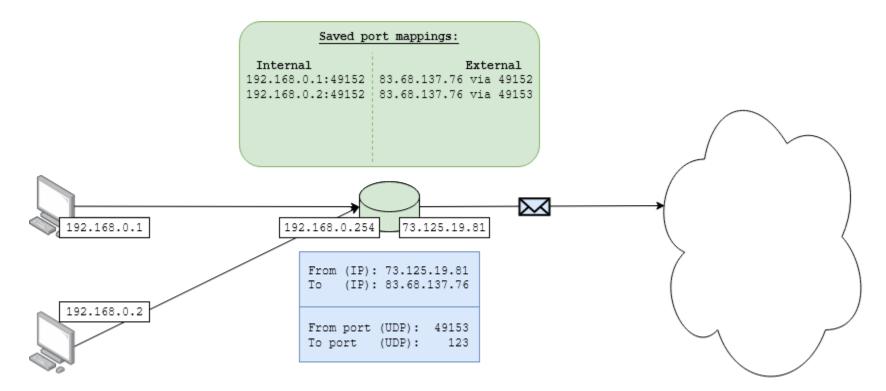
- "Hide" an entire private network behind a single public IP
 - Rewrite IP packets at the boundary



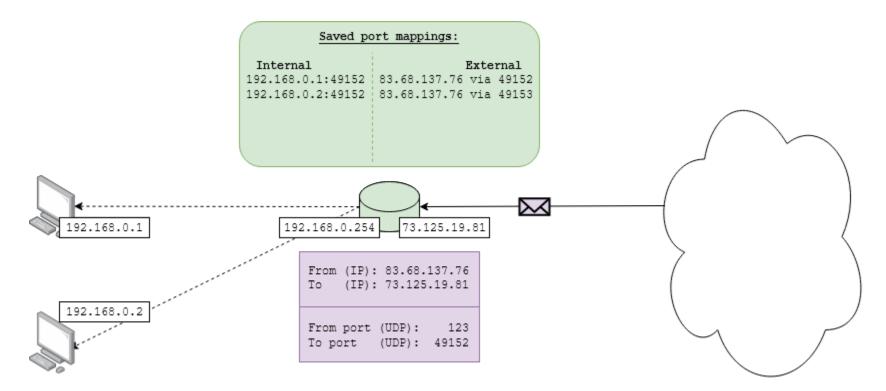
- "Hide" an entire private network behind a single public IP
 - Rewrite IP packets at the boundary



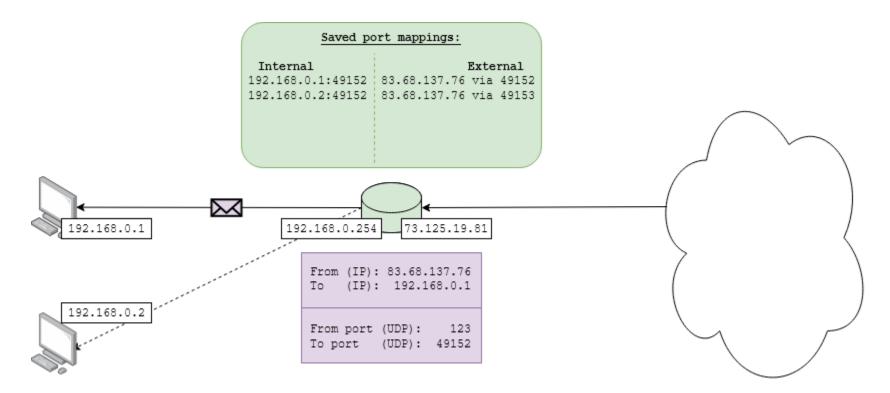
- "Hide" an entire private network behind a single public IP
 - Rewrite IP packets at the boundary
 - Rewrite TCP/UDP ports to disambiguate



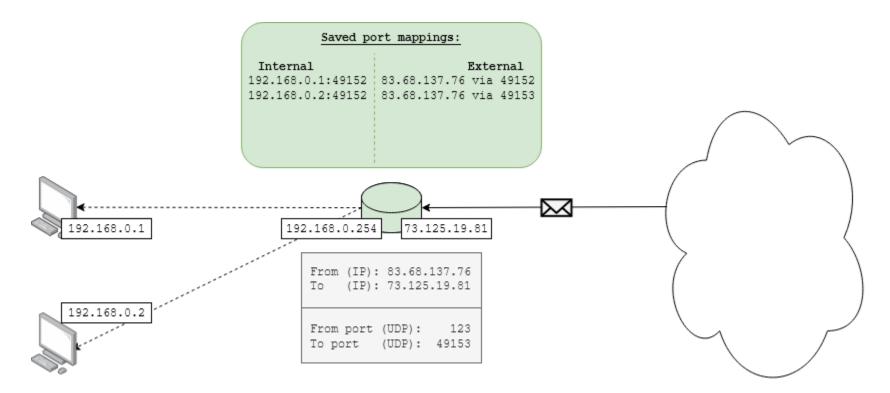
- "Hide" an entire private network behind a single public IP
 - Rewrite IP packets at the boundary
 - Rewrite TCP/UDP ports to disambiguate



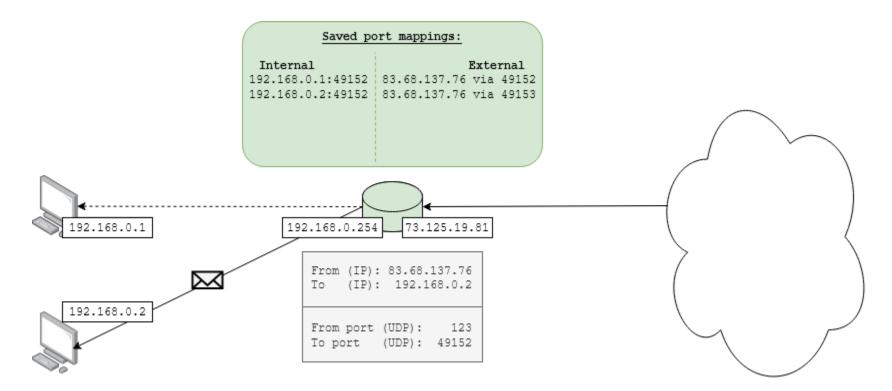
- "Hide" an entire private network behind a single public IP
 - Rewrite IP packets at the boundary
 - Rewrite TCP/UDP ports to disambiguate



- "Hide" an entire private network behind a single public IP
 - Rewrite IP packets at the boundary
 - Rewrite TCP/UDP ports to disambiguate



- "Hide" an entire private network behind a single public IP
 - Rewrite IP packets at the boundary
 - Rewrite TCP/UDP ports to disambiguate



- "Hide" an entire private network behind a single public IP
 - Rewrite IP packets at the boundary
 - Rewrite TCP/UDP ports to disambiguate
- Transparent if a client "inside" connects to a server "outside"
 - The reverse will not work (by default)
- You can have PAT networks nested within PAT networks
 - Entire ISPs can connect all their clients using one publicly-routable IP address!
- Your home ISP router almost definitely does this!
 - Compare your **ipconfig/ifconfig** address with "what's my ip" (google)

IPv4 fragmentation – Issues

- 16-bit packet ID is insufficient for high transmission rates

 - No acknowledgments ⇒ ID can't be reused until TTL expires
 - 65536 packets ÷ 128 seconds = 512 packets per second



IPv4 fragmentation – Issues

- Fragmentation splits TCP/UDP transmission units across IP packets
 - Only the first fragment has the transport layer header!
 - If it even does...? There's no minimum fragment size...
 - Which host should a PAT firewall forward the data to?
 - Firewalls can't effectively filter these packets either...
- Reassembly is very fragile
 - How long should fragments be kept around for?
 - Denial-of-service attacks!
 - How do you handle overlapping fragments?
 - They are valid as per the protocol spec...