Lecture 16: Network Layer Overview, Internet Protocol COMP 411, Fall 2022 Victoria Manfredi





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Today

1. TCP congestion control

2. Network layer

- overview
- what's inside a router
- Internet protocol (IP)

TCP CONGESTION CONTROL

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Slow start: when to stop exponential increase?



Slow start

- initially cwnd = 1 MSS
- every time ACK received, double cwnd

Congestion avoidance



When cwnd = ssthresh

- go to congestion avoidance
- use AIMD

Timeout



Restart slow start when timeout

- ssthresh = cwnd/2
- cwnd = 1 MSS

3 duplicate ACKs



If 3 duplicate ACKs go to fast recovery

- ssthresh = cwnd/2
- cwnd = ssthresh + 3 MSS

Average TCP throughput

Focus on AIMD

- ignore slow start, assume always data to send





Setting window size

Window is min (rwnd, cwnd)

•	Transmission Control Protocol, Src Port: 443 (443), Dst Port: 52232 (52232), Seq: 0, Ack: 1,								
	Source Port: 443								
	Destination Port: 52232								
	[Stream index: 0]								
	[TCP Segment Len: 0]								
	Sequence number: 0 (relative sequence number)								
	Acknowledgment number: 1 (relative ack number)								
	Header Length: 32 bytes								
	▼ Flags: 0x012 (SYN, ACK)								
	000 = Reserved: Not set								
	0 = Nonce: Not set								
	<pre> 0 = Congestion Window Reduced (CWR): Not set</pre>								
	0 = Urgent: Not set								
	1 = Acknowledgment: Set								
	0 = Push: Not set								
	0 = Reset: Not set								
	<pre>1. = Syn: Set</pre>								
	0 = Fin: Not set								
	[TCP Flags: *****A**S*]								
	Window size value: 8190 rwnd								
	[Calculated window size: 8190]								
	Charleson Aveloa [velidation dischlad]								

Network Layer OVERVIEW

5-layer Internet protocol stack

	Layer	Service provided to upper layer	Protocols	Unit of information
5	Application	 Support network applications 	FTP, DNS, SMTP, HTTP	Message 1 message may be split into multiple segments
4	Transport	 Deliver messages to app endpoints Flow control Reliability 	TCP (reliable) UDP (best-effort)	Segment (TCP) Datagram (UDP) 1 segment may be split into multiple packets
3	Network	 Route segments from source to destination host 	IP (best-effort) Routing protocols	Packet (TCP) Datagram (UDP)
2	Link	 Move packet over link from one host to next host 	Ethernet, 802.11	Frame MTU is 1500 bytes
1	Physical	 Move individual bits in frame from one host to next "bits on wire" 	Ethernet phy 802.11 phy Bluetooth phy DSL	Bit

Internet's network layer

Network layer functions on hosts and routers



Network layer

Goal

- move pkt from one host to another

How done on Internet?

- routers
 - examine header fields in every IP pkt
 - determines outgoing link

Internet e2e argument

- some functionality only properly implemented in end systems
- smart hosts vs. dumb routers



Network layer is in every host and router on Internet

Encapsulation and decapsulation

Sender

- encapsulates segments into packets, puts src, dest IP in IP pkt hdr

Receiver

- decapsulates packets into segments, delivers to transport layer

Max length of IP packet in bytes

- MTU: Maximum Transmission Unit
- 1500 bytes if Ethernet used as link layer protocol

Max length of TCP data in bytes

- MSS: Maximum Segment Size
- MSS = MTU IP hdr TCP hdr
 - TCP header >= 20bytes



Division of network layer functionality

1. Control plane

- comprises traffic only between routers, to compute routes between src and dst
- network-wide: routers run routing algorithms

2. Data plane

- comprises traffic between end hosts, forwarded by routers
- forwarding table set based on routes computed in control plane
- local: each router stores, forwards packets

Interplay between routing and forwarding

Routing (slower time scale)

- routers view Internet as graph
- run shortest path algorithms

Forwarding (faster time scale)

- routers use paths to choose best output link for packet destination IP address
- if one link fails, chooses another



- 1. traditional routing algorithms implemented in routers
- 2. software-defined networking (SDN) implemented in (remote) servers

Approach 1: per-router control plane

Individual routing algorithm components in each and every router interact in the control plane



Approach 2: logically centralized control plane

A distinct (typically remote) controller interacts with local control agents (CAs)



Network layer service model

Q: What service model does network layer provide to transport layer for moving packets from sender to receiver?

Example services

- individual packets
 - guaranteed delivery
 - guaranteed delivery with less than 40 ms delay

- flow of packets

- in-order packet delivery
- guaranteed minimum bandwidth to flow
- restrictions on changes in inter-packet spacing

Network layer service models

Network		Service	Guarantees ?				Congestion
Arch	hitecture	Model	Bandwidth	Loss	Order	Timing	feedback
	Internet	best effort	none	no	no	no	no (inferred via loss)
	ATM	CBR	constant	yes	yes	yes	no
			rate				congestion
	ATM	VBR	guaranteed	yes	yes	yes	no
			rate				congestion
-	ATM	ABR	guaranteed	no	yes	no	yes
			minimum				
	ATM	UBR	none	no	yes	no	no

ATM: Asynchronous Transfer Mode e.g., used in public switched telephone network

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Network Layer WHAT'S INSIDE A ROUTER?

What does a router need to do?

Run routing protocols (control) and store and forward pkts (data)



Input port functions



Switching fabrics

Transfer packet

- from input buffer to appropriate output buffer

Switching rate

- rate at which packets can be transferred from inputs to outputs
- N inputs: switching rate = N x line rate desirable

3 types of switching fabrics



pkts in parallel

Contention at input ports

If switching fabric slower than input ports combined

- queueing may occur at input queues
- queueing delay and loss due to input buffer overflow!

Head-of-the-Line (HOL) blocking

- queued pkt at front of queue prevents others from moving forward





Output port contention: only one red packet can be transferred. Lower red packet is blocked

One packet time later: green packet experiences HOL blocking

Contention at output ports



Buffering

- when packets arrive from fabric faster than transmission rate
- packet loss: due to congestion, lack of buffers

Scheduling

- chooses next among queued packets to transmit on link
- net neutrality: who gets best performance

Scheduling mechanisms

FIFO (first in first out)

send in order of arrival to queue

Priority

- packet arrivals queue link packet (waiting area) (server)
- multiple classes, with different priorities (e.g., based on hdr info)
 - send highest priority queued packet

Round robin scheduling

- multiple classes, cyclically scan class queues
 - send one packet from each class (if available)

Weighted fair queueing

- generalized round robin
 - · each class gets weighted amount of service in each cycle

In practice: hardware queues use FIFO, need software to do priority

Network Layer INTERNET PROTOCOL

Internet Protocol (IP)

THE network layer protocol of the Internet

- protocol your device <u>must</u> implement to run on Internet
- RFC published ~1980
- Provides
 - best effort service
 - to get pkts from one end host to another across many interconnected networks using dst IP address in IP hdr
 - addressing
 - format and usage of addresses
 - fragmentation
 - e.g., if pkt size exceeds Ethernet MTU of 1500 bytes
 - some error detection

Q: what does IP not provide?

QoS, reliability, ordering, persistent state for e2e flows, connections