Networking

Chapter 4
Network Layer: Data Plane

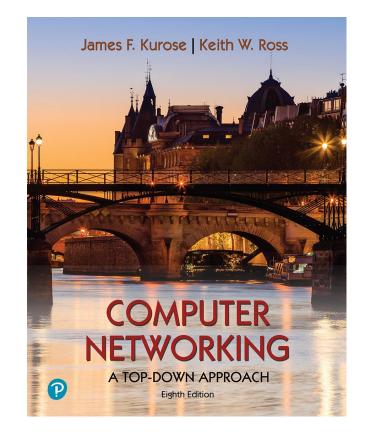








Chapter 4 Network Layer: Data Plane



Computer Networking: A Top-Down Approach

8th edition Jim Kurose, Keith Ross Pearson, 2020





Network layer: our goals

- •understand principles behind network layer services, focusing on data plane:
 - network layer service models
 - forwarding versus routing
 - how a router works
 - addressing
 - generalized forwarding
 - Internet architecture

- instantiation, implementation in the Internet
 - IP protocol
 - NAT, middleboxes





Pearson Network layer: "data plane" roadmap

- Network layer: overview
 - data plane
 - control plane
- What's inside a router
 - input ports, switching, output ports
 - buffer management, scheduling
- IP: the Internet Protocol
 - datagram format
 - addressing
 - network address translation
 - IPv6



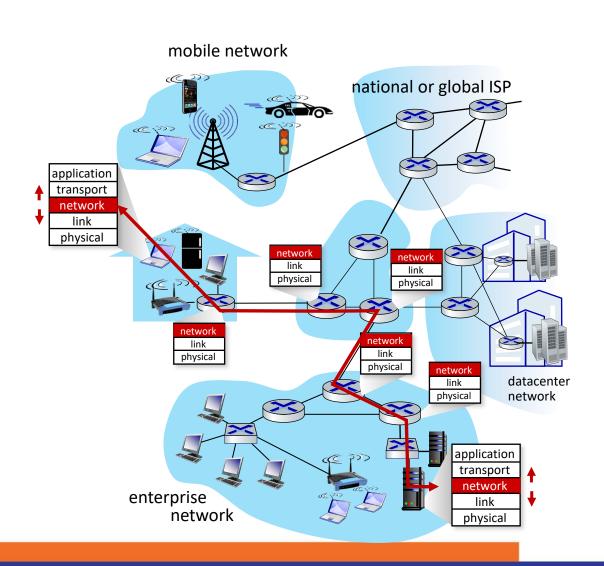
- Generalized Forwarding, SDN
 - Match+action
 - OpenFlow: match+action in action
- Middleboxes





Pearson Network-layer services and protocols

- transport segment from sending to receiving host
 - sender: encapsulates segments into datagrams, passes to link layer
 - receiver: delivers segments to transport layer protocol
- network layer protocols in every Internet device: hosts, routers
- routers:
 - examines header fields in all IP datagrams passing through it
 - moves datagrams from input ports to output ports to transfer datagrams along end-end path







Two key network-layer functions

network-layer functions:

- forwarding: move packets from a router's input link to appropriate router output link
- routing: determine route taken by packets from source to destination
 - routing algorithms

analogy: taking a trip

- *forwarding:* process of getting through single interchange
- routing: process of planning trip from source to destination

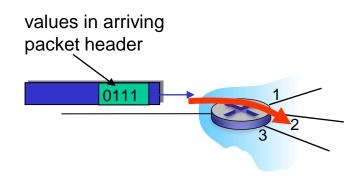






Data plane:

- local, per-router function
- determines how datagram arriving on router input port is forwarded to router output port



Network layer: data plane, control plane

Control plane

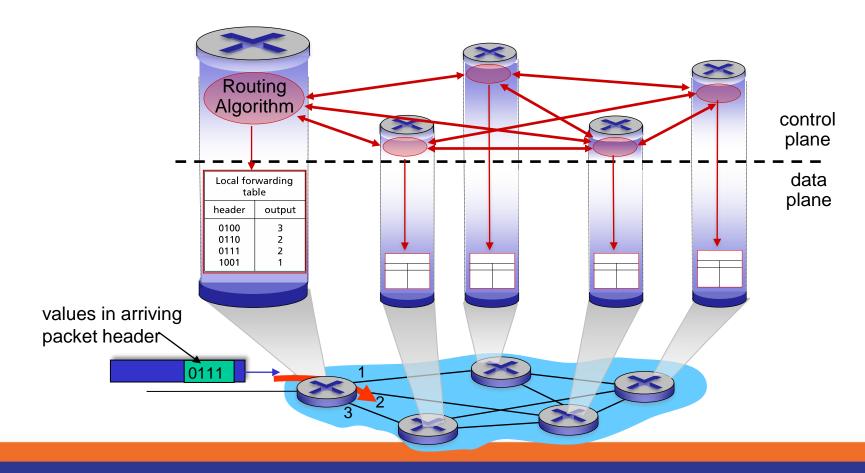
- network-wide logic
- determines how datagram is routed among routers along endend path from source host to destination host
- two control-plane approaches:
 - traditional routing algorithms: implemented in routers
 - software-defined networking (SDN): implemented in (remote) servers





Per-router control plane

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

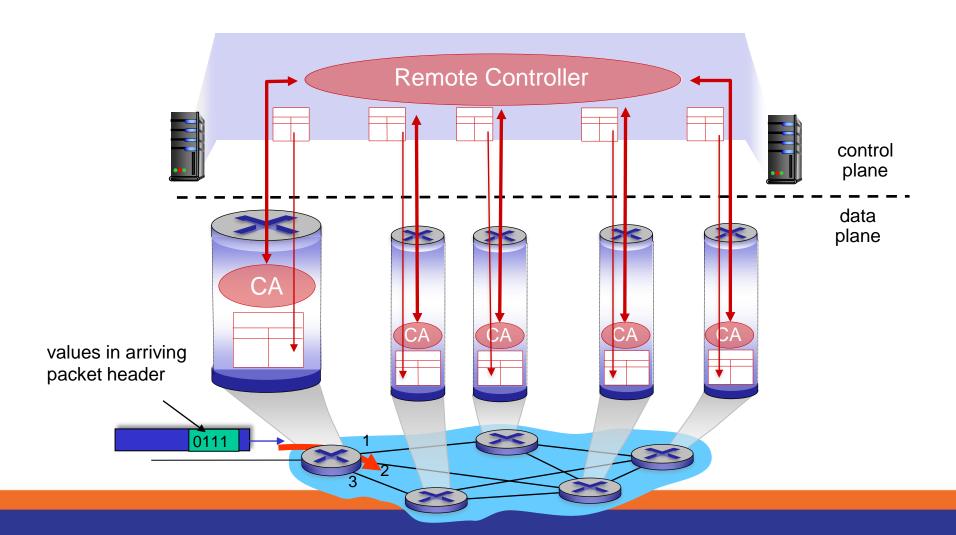






Software-Defined Networking (SDN)

Remote controller computes, installs forwarding tables in routers







Network service model

Q: What service model for "channel" transporting datagrams from sender to receiver?

example services for *individual* datagrams:

- guaranteed delivery
- guaranteed delivery with less than 40 msec delay

example services for a *flow* of datagrams:

- in-order datagram delivery
- guaranteed minimum bandwidth to flow
- restrictions on changes in interpacket spacing





Network-layer service model

 \bigcirc

Network	Service	Quality of Service (QoS) Guarantees ?				
Architecture	Model	Bandwidth	Loss	Order	Timing	
Internet	best effort	none	no	no	no	

Internet "best effort" service model

No guarantees on:

- i. successful datagram delivery to destination
- ii. timing or order of delivery
- iii. bandwidth available to end-end flow





Network-layer service model

Network Service Architecture Model		Service	Quality of Service (QoS) Guarantees ?				
		Bandwidth	Loss	Order	Timing		
	Internet	best effort	none	no	no	no	
	ATM	Constant Bit Rate	Constant rate	yes	yes	yes	
	ATM	Available Bit Rate	Guaranteed min	no	yes	no	
	Internet	Intserv Guaranteed (RFC 1633)	yes	yes	yes	yes	
	Internet	Diffserv (RFC 2475)	possible	possibly	possibly	no	





Pearson Reflections on best-effort service:

- simplicity of mechanism has allowed Internet to be widely deployed adopted
- sufficient provisioning of bandwidth allows performance of real-time applications (e.g., interactive voice, video) to be "good enough" for "most of the time"
- replicated, application-layer distributed services (datacenters, content distribution networks) connecting close to clients' networks, allow services to be provided from multiple locations
- congestion control of "elastic" services helps

It's hard to argue with success of best-effort service model





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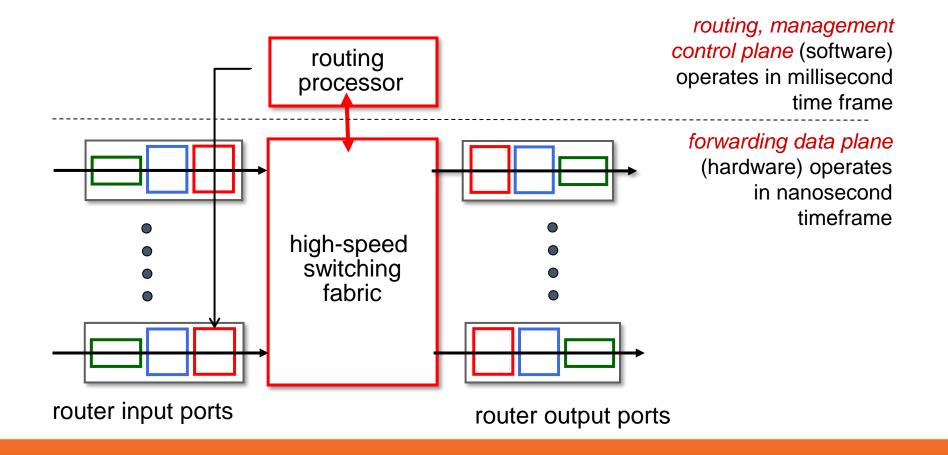
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Router architecture overview

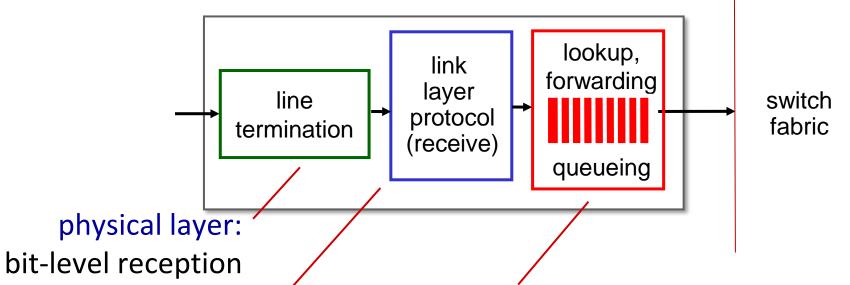
high-level view of generic router architecture:







Input port functions



link layer:

e.g., Ethernet (chapter 6)

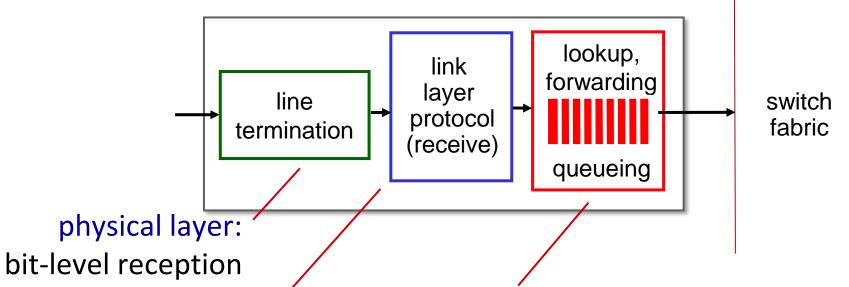
decentralized switching:

- using header field values, lookup output port using forwarding table in input port memory ("match plus action")
- goal: complete input port processing at 'line speed'
- input port queuing: if datagrams arrive faster than forwarding rate into switch fabric





Input port functions



link layer:

e.g., Ethernet (chapter 6)

decentralized switching:

- using header field values, lookup output port using forwarding table in input port memory ("match plus action")
- destination-based forwarding: forward based only on destination IP address (traditional)
- generalized forwarding: forward based on any set of header

field values





Destination-based forwarding

forwarding table					
Destination Address Range				Link Interface	
11001000 0	00010111	000 <mark>10000</mark>	0000000	n	
11001000 0 through	00010111	000 <mark>10000</mark>	00000100	3	
11001000 0	00010111	000 <mark>10000</mark>	00000111	•	
11001000 0	00010111	000 <mark>11000</mark>	11111111		
11001000 0 through	00010111	000 <mark>11001</mark>	0000000	2	
11001000 0	00010111	000 <mark>11111</mark>	11111111		
otherwise				3	

former and in our for bull-

Q: but what happens if ranges don't divide up so nicely?





longest prefix match

when looking for forwarding table entry for given destination address, use *longest* address prefix that matches destination address.

Destination A	Link interface			
11001000	00010111	00010***	*****	0
11001000	00010111	00011000	*****	1
11001000	00010111	00011***	*****	2
otherwise	3			

examples:

11001000 00010111 00010110 10100001 which interface?

11001000 00010111 00011000 10101010 which interface?





longest prefix match

when looking for forwarding table entry for given destination address, use *longest* address prefix that matches destination address.

Destination A	Link interface			
11001000	00010111	00010***	*****	0
11001000	000.0111	00011000	*****	1
11001000	match! 1	00011***	*****	2
otherwise				3
11001000	00010111	00010110	1010001	which interface?

examples





longest prefix match

when looking for forwarding table entry for given destination address, use *longest* address prefix that matches destination address.

Destination .	Link interface			
11001000	00010111	00010***	*****	0
11001000	00010111	00011000	*****	1
11001000	00010111	00011***	*****	2
otherwise	1			3
11001000	match!	00010110	10100001	which interface?

examples:

1001000 00010111 00011 000 10101010 which interface





longest prefix match

when looking for forwarding table entry for given destination address, use *longest* address prefix that matches destination address.

Destination A	Link interface			
11001000	00010111	00010***	*****	0
11001000	00010111	00011000	*****	1
11001000	000 0111	00011***	******	2
otherwise	match! —			3

examples:

10101010 which interface

which interface?





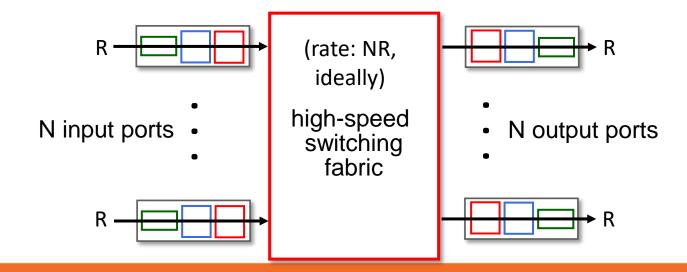
- we'll see why longest prefix matching is used shortly, when we study addressing
- longest prefix matching: often performed using ternary content addressable memories (TCAMs)
 - content addressable: present address to TCAM: retrieve address in one clock cycle, regardless of table size
 - Cisco Catalyst: ~1M routing table entries in TCAM





Switching fabrics

- transfer packet from input link to appropriate output link
- switching rate: rate at which packets can be transfer from inputs to outputs
 - often measured as multiple of input/output line rate
 - N inputs: switching rate N times line rate desirable

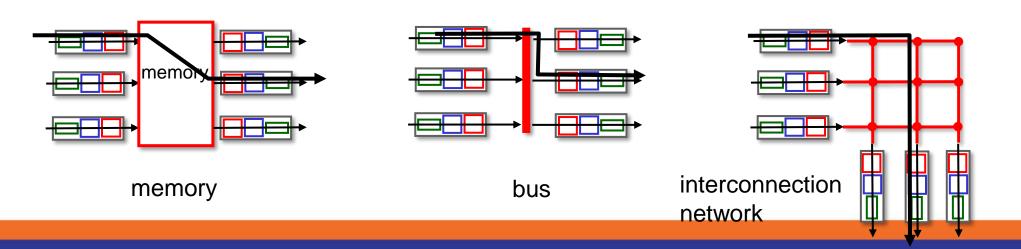






Switching fabrics

- transfer packet from input link to appropriate output link
- switching rate: rate at which packets can be transfer from inputs to outputs
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 - N inputs: switching rate N times line rate desirable
- three major types of switching fabrics:



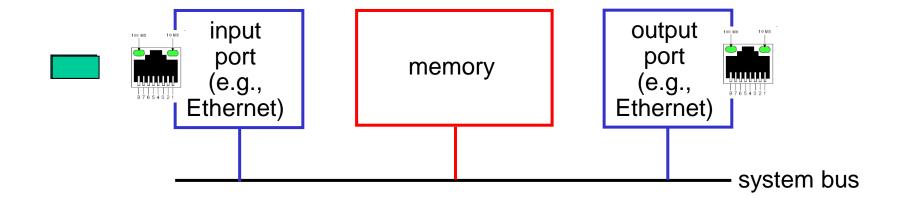




Switching via memory

first generation routers:

- traditional computers with switching under direct control of CPU
- packet copied to system's memory
- speed limited by memory bandwidth (2 bus crossings per datagram)

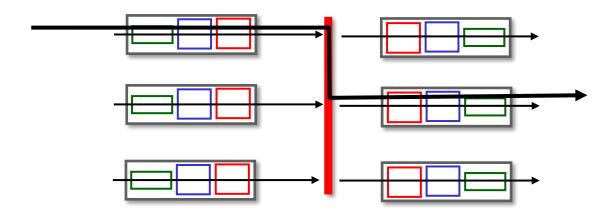






Switching via a bus

- datagram from input port memory to output port memory via a shared bus
- bus contention: switching speed limited by bus bandwidth
- 32 Gbps bus, Cisco 5600: sufficient speed for access routers

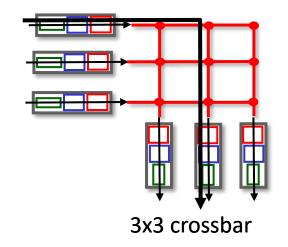


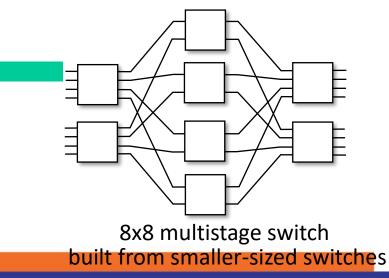




Switching via interconnection network

- Crossbar, Clos networks, other interconnection nets initially developed to connect processors in multiprocessor
- multistage switch: nxn switch from multiple stages of smaller switches
- exploiting parallelism:
 - fragment datagram into fixed length cells on entry
 - switch cells through the fabric, reassemble datagram at exit



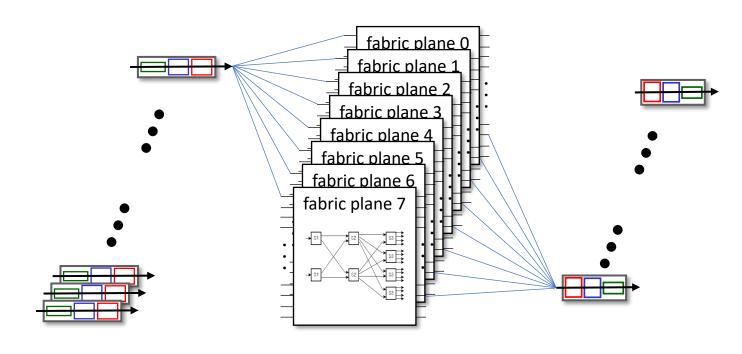






Switching via interconnection network

- scaling, using multiple switching "planes" in parallel:
 - speedup, scaleup via parallelism
- Cisco CRS router:
 - basic unit: 8 switching planes
 - each plane: 3-stage interconnection network
 - up to 100's Tbps switching capacity

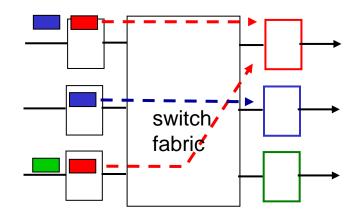




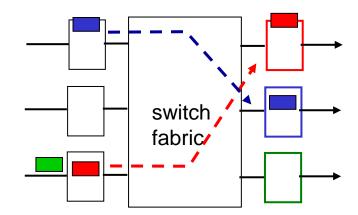


Input port queuing

- If switch 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 datagram at front of queue prevents others in queue from moving forward



output port contention: only one red datagram can be transferred. lower red

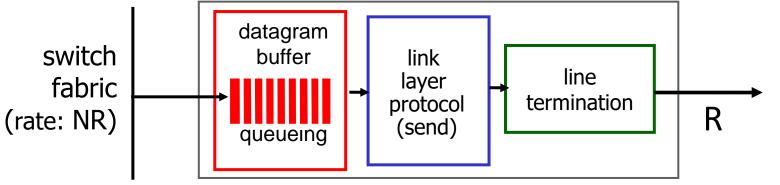


one packet time later: green packet experiences HOL blocking



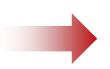






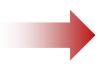
This is a really important slide

Buffering required when datagrams arrive from fabric faster than link transmission rate. Drop policy: which datagrams to drop if no free buffers?



Datagrams can be lost due to congestion, lack of buffers

 Scheduling discipline chooses among queued datagrams for transmission

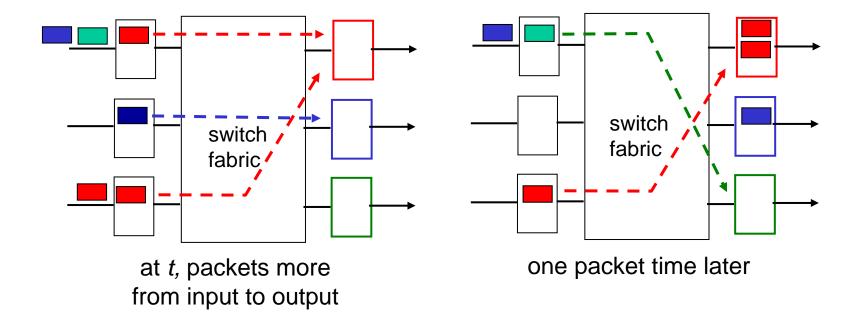


Priority scheduling – who gets best performance, network neutrality





Output port queuing



- buffering when arrival rate via switch exceeds output line speed
- queueing (delay) and loss due to output port buffer overflow!





How much buffering?

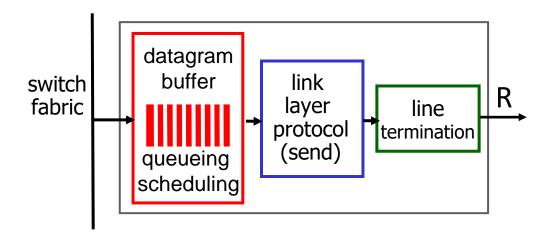
- RFC 3439 rule of thumb: average buffering equal to "typical" RTT (say 250 msec) times link capacity C
 - e.g., C = 10 Gbps link: 2.5 Gbit buffer
- more recent recommendation: with N flows, buffering equal to

$$\frac{\mathsf{RTT} \cdot \mathsf{C}}{\sqrt{\mathsf{N}}}$$

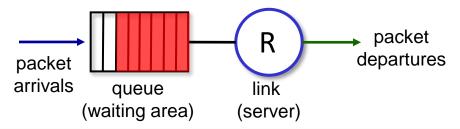
- but too much buffering can increase delays (particularly in home routers)
 - long RTTs: poor performance for realtime apps, sluggish TCP response
 - recall delay-based congestion control: "keep bottleneck link just full enough (busy) but no fuller"







Abstraction: queue



Buffer Management

buffer management:

- drop: which packet to add, drop when buffers are full
 - tail drop: drop arriving packet
 - priority: drop/remove on priority basis
- marking: which packets to mark to signal congestion (ECN, RED)



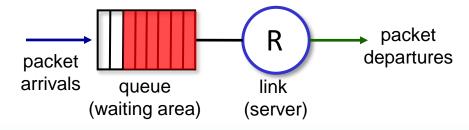


Packet Scheduling: FCFS

packet scheduling: deciding which packet to send next on link

- first come, first served
- priority
- round robin
- weighted fair queueing

Abstraction: queue



FCFS: packets transmitted in order of arrival to output port

- also known as: First-in-firstout (FIFO)
- real world examples?

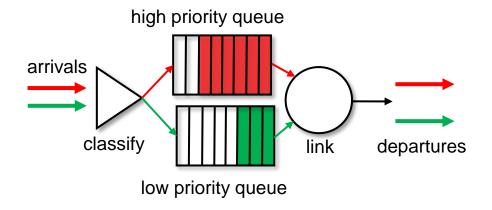


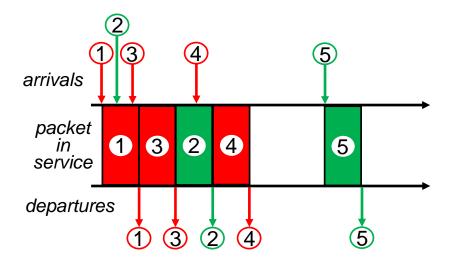


Scheduling policies: priority

Priority scheduling:

- arriving traffic classified, queued by class
 - any header fields can be used for classification
- send packet from highest priority queue that has buffered packets
 - FCFS within priority class





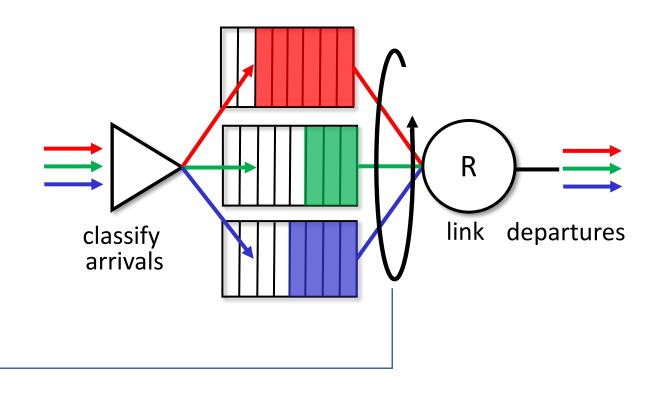




Scheduling policies: round robin

Round Robin (RR) scheduling:

- arriving traffic classified, queued by class
 - any header fields can be used for classification
- server cyclically, repeatedly scans class queues, sending one complete packet from each class (if available) in turn







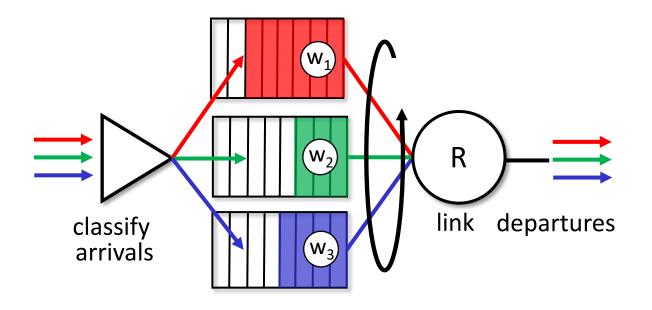
Scheduling policies: weighted fair queueing

Weighted Fair Queuing (WFQ):

- generalized Round Robin
- each class, i, has weight, w_i, and gets weighted amount of service in each cycle:

$$\frac{w_i}{\sum_j w_j}$$

 minimum bandwidth guarantee (per-traffic-class)







Sidebar: Network Neutrality

What is network neutrality?

- technical: how an ISP should share/allocation its resources
 - packet scheduling, buffer management are the mechanisms
- social, economic principles
 - protecting free speech
 - encouraging innovation, competition
- enforced *legal* rules and policies

Different countries have different "takes" on network neutrality





Sidebar: Network Neutrality

2015 US FCC Order on Protecting and Promoting an Open Internet: three "clear, bright line" rules:

- no blocking ... "shall not block lawful content, applications, services, or non-harmful devices, subject to reasonable network management."
- no throttling ... "shall not impair or degrade lawful Internet traffic on the basis of Internet content, application, or service, or use of a non-harmful device, subject to reasonable network management."
- no paid prioritization. ... "shall not engage in paid prioritization"





ISP: telecommunications or information service?

Is an ISP a "telecommunications service" or an "information service" provider?

the answer really matters from a regulatory standpoint!

US Telecommunication Act of 1934 and 1996:

- Title II: imposes "common carrier duties" on telecommunications services: reasonable rates, non-discrimination and requires regulation
- Title I: applies to information services:
 - no common carrier duties (not regulated)
 - but grants FCC authority "... as may be necessary in the execution of its functions".





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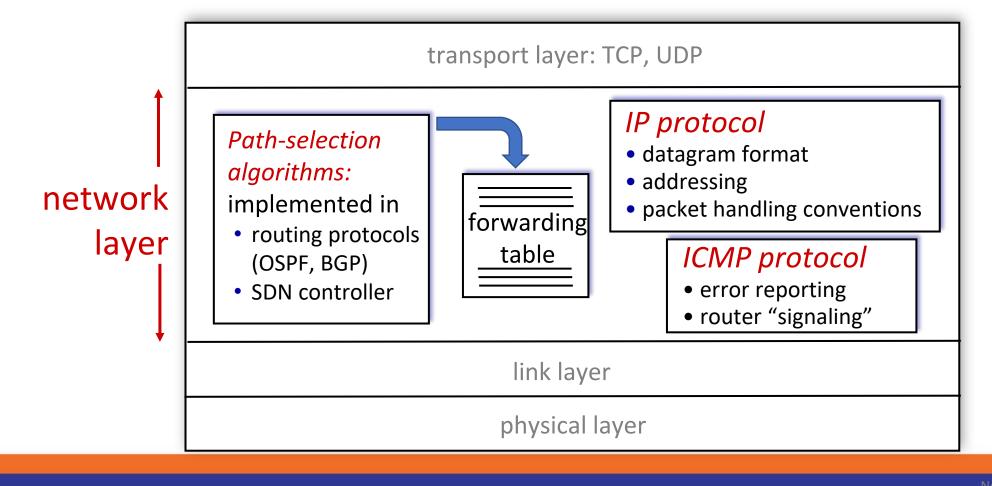
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Network Layer: Internet

host, router network layer functions:







IP Datagram format

IP protocol version number

header length(bytes)

"type" of service:

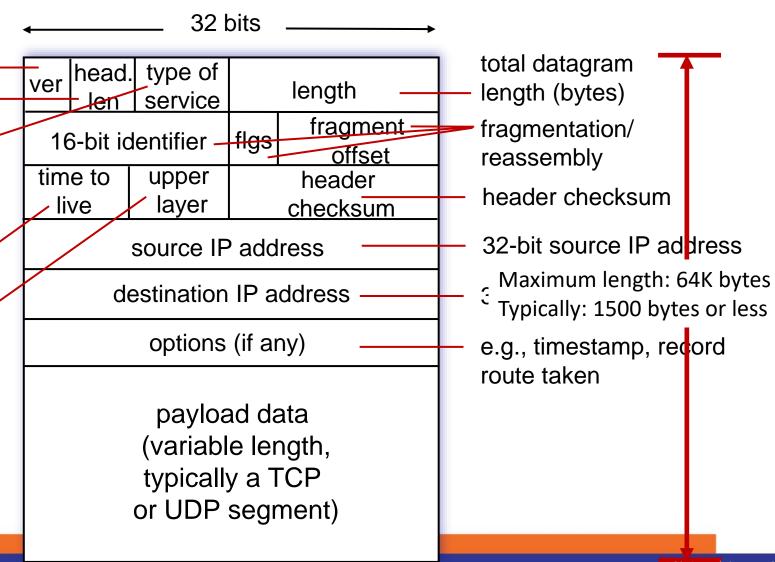
- diffserv (0:5)
- ECN (6:7)

TTL: remaining max hops (decremented at each router)

upper layer protocol (e.g., TCP or UDP)

overhead

- 20 bytes of TCP
- 20 bytes of IP
- = 40 bytes + app layer overhead for TCP+IP

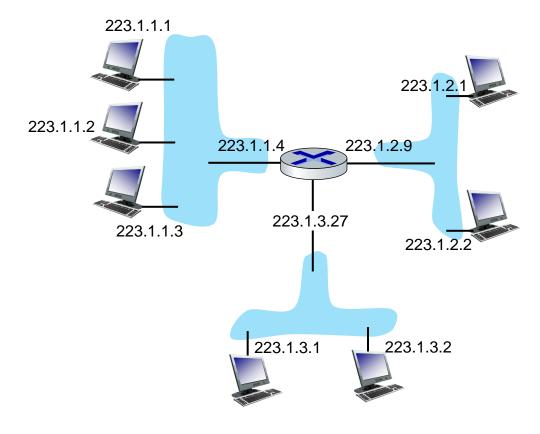






IP addressing: introduction

- IP address: 32-bit identifier associated with each host or router interface
- interface: connection between host/router and physical link
 - router's typically have multiple interfaces
 - host typically has one or two interfaces (e.g., wired Ethernet, wireless 802.11)



dotted-decimal IP address notation:

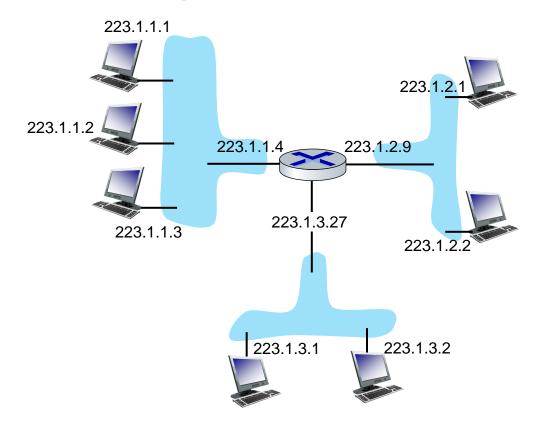
223.1.1.1 = 110111111 00000001 00000001 00000001





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dotted-decimal IP address notation:

223.1.1.1 = 110111111 00000001 00000001 00000001





IP addressing: introduction

223.1.1.1

Q: how are interfaces actually connected?

A: we'll learn about that in chapters 6, 7

A: wired

Ethernet interfaces
connected by
Ethernet switches

223.1.2.1 223.1.1.4 223.1.2.9 223.1.3.27 223.1.1.3 223.1.3.1 223.1.3.2

A: wireless WiFi interfaces connected by WiFi base station

For now: don't need to worry about how one interface is connected to another (with no intervening router)

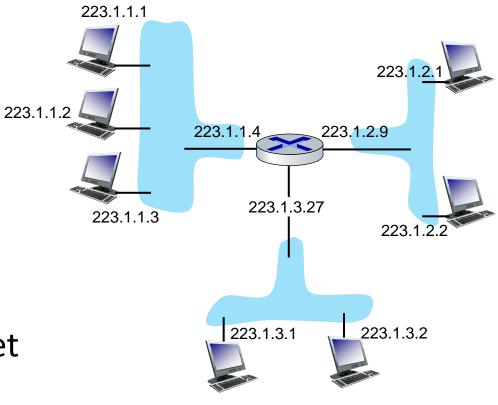




Subnets

■ What's a subnet?

- device interfaces that can physically reach each other without passing through an intervening router
- IP addresses have structure:
 - subnet part: devices in same subnet have common high order bits
 - host part: remaining low order bits



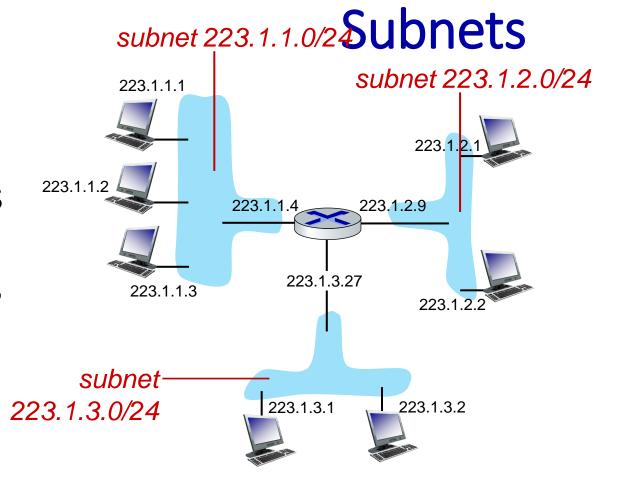
network consisting of 3 subnets





Recipe for defining subnets:

- detach each interface from its host or router, creating "islands" of isolated networks
- each isolated network is called a *subnet*



subnet mask: /24

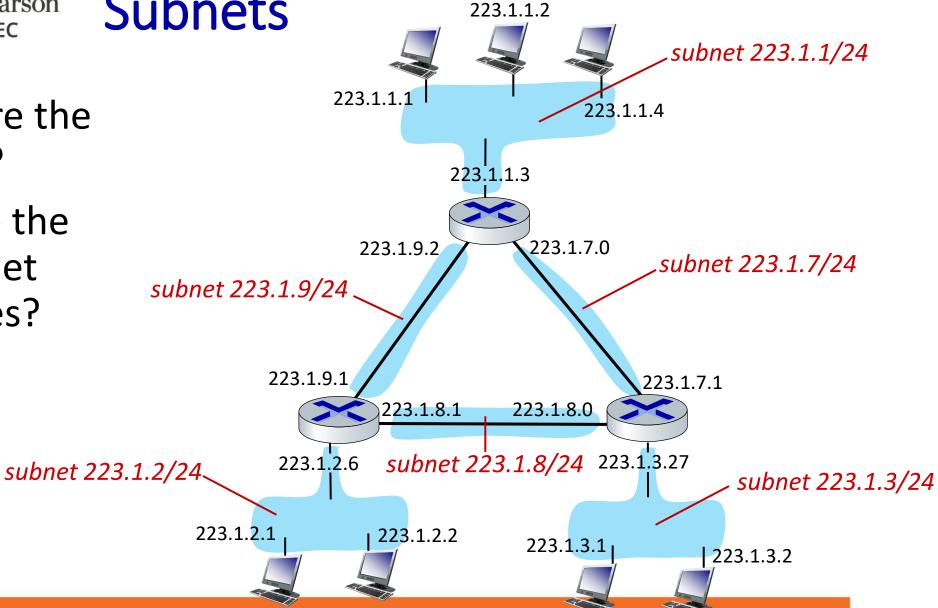
(high-order 24 bits: subnet part of IP address)





Subnets

- where are the subnets?
- what are the /24 subnet addresses?







IP addressing: CIDR

CIDR: Classless InterDomain Routing (pronounced "cider")

- subnet portion of address of arbitrary length
- address format: a.b.c.d/x, where x is # bits in subnet portion of address



11001000 00010111 00010000 00000000

200.23.16.0/23





IP addresses: how to get one?

That's actually two questions:

- 1. Q: How does a *host* get IP address within its network (host part of address)?
- 2. Q: How does a *network* get IP address for itself (network part of address)

How does *host* get IP address?

- hard-coded by sysadmin in config file (e.g., /etc/rc.config in UNIX)
- DHCP: Dynamic Host Configuration Protocol: dynamically get address from as server
 - "plug-and-play"





DHCP: Dynamic Host Configuration Protocol

goal: host dynamically obtains IP address from network server when it "joins" network

- can renew its lease on address in use
- allows reuse of addresses (only hold address while connected/on)
- support for mobile users who join/leave network

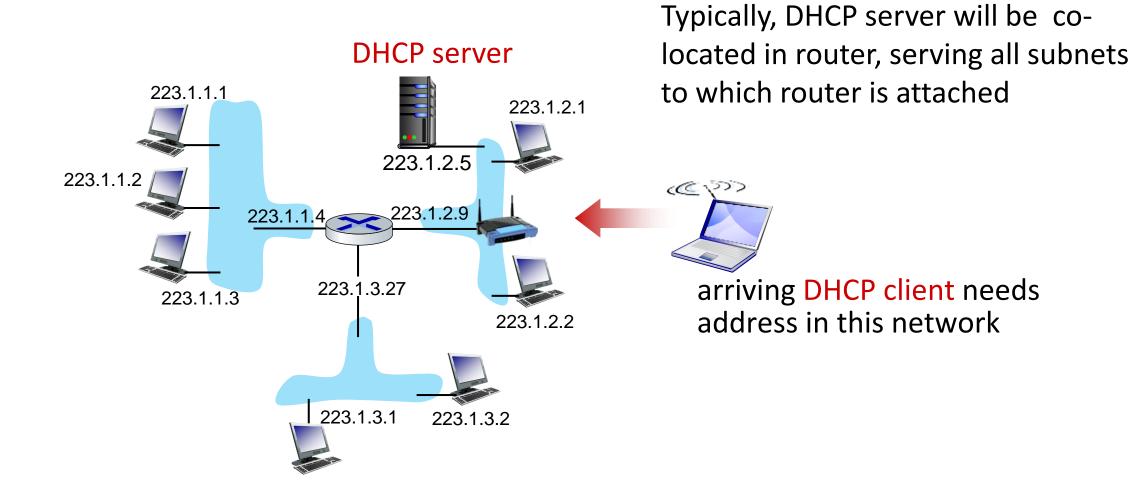
DHCP overview:

- host broadcasts DHCP discover msg [optional]
- DHCP server responds with DHCP offer msg [optional]
- host requests IP address: DHCP request msg
- DHCP server sends address: DHCP ack msg





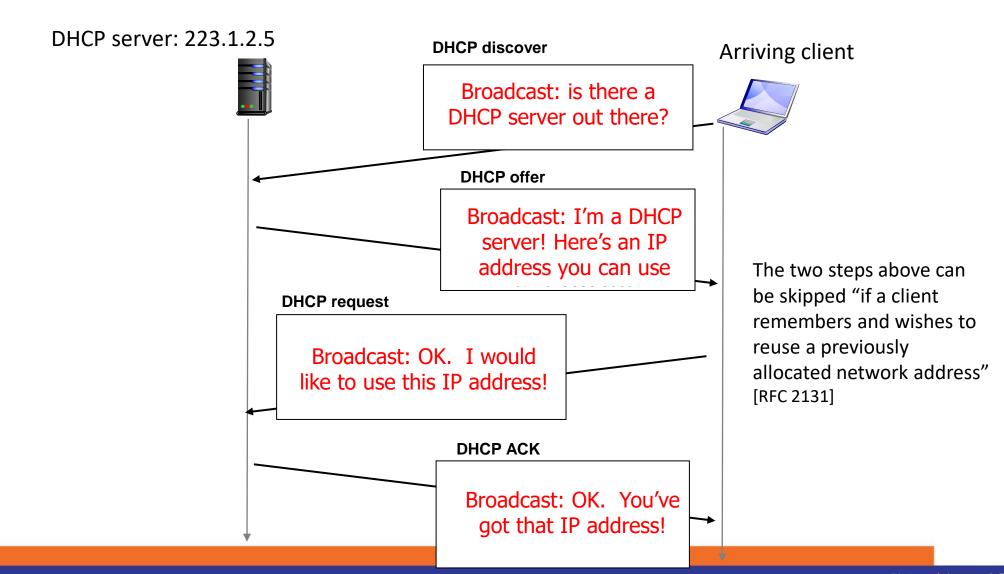
DHCP client-server scenario







DHCP client-server scenario







DHCP: more than IP addresses

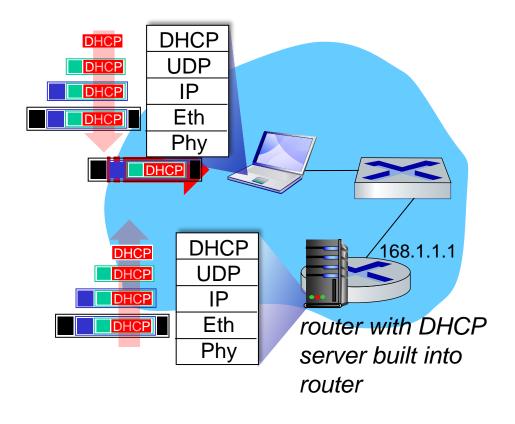
DHCP can return more than just allocated IP address on subnet:

- address of first-hop router for client
- name and IP address of DNS sever
- network mask (indicating network versus host portion of address)





DHCP: example



- Connecting laptop will use DHCP to get IP address, address of firsthop router, address of DNS server.
- DHCP REQUEST message encapsulated in UDP, encapsulated in IP, encapsulated in Ethernet
- Ethernet demux'ed to IP demux'ed,
 UDP demux'ed to DHCP





DHCP UDP IΡ Eth Phy **DHCP UDP** IΡ Eth router with DHCP Phy server built into router

DHCP: example

- DCP server formulates DHCP ACK containing client's IP address, IP address of first-hop router for client, name & IP address of DNS server
- encapsulated DHCP server reply forwarded to client, demuxing up to DHCP at client
- client now knows its IP address, name and IP address of DNS server, IP address of its first-hop router





IP addresses: how to get one?

Q: how does network get subnet part of IP address?

A: gets allocated portion of its provider ISP's address space

ISP's block 11001000 00010111 00010000 00000000 200.23.16.0/20

ISP can then allocate out its address space in 8 blocks:

 Organization 0
 11001000 00010111 0001000
 00000000
 200.23.16.0/23

 Organization 1
 11001000 00010111 0001001
 00000000
 200.23.18.0/23

 Organization 2
 11001000 00010111 0001010
 00000000
 200.23.20.0/23

...

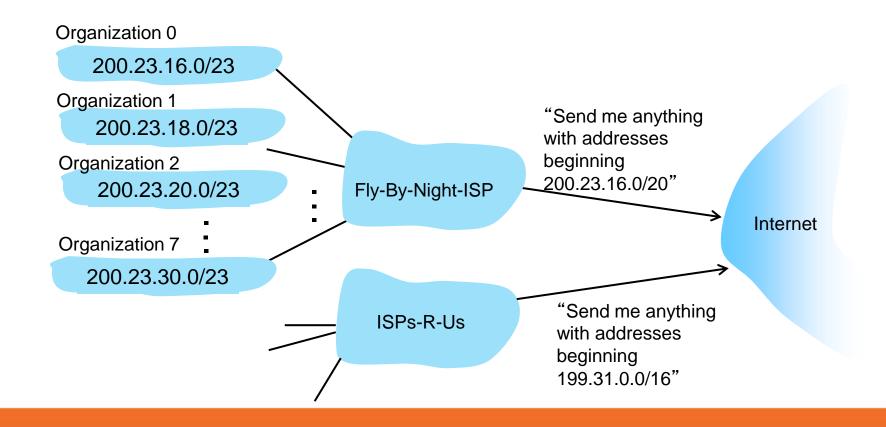
Organization 7 11001000 00010111 00011110 00000000 200.23.30.0/23





Hierarchical addressing: route aggregation

hierarchical addressing allows efficient advertisement of routing information:

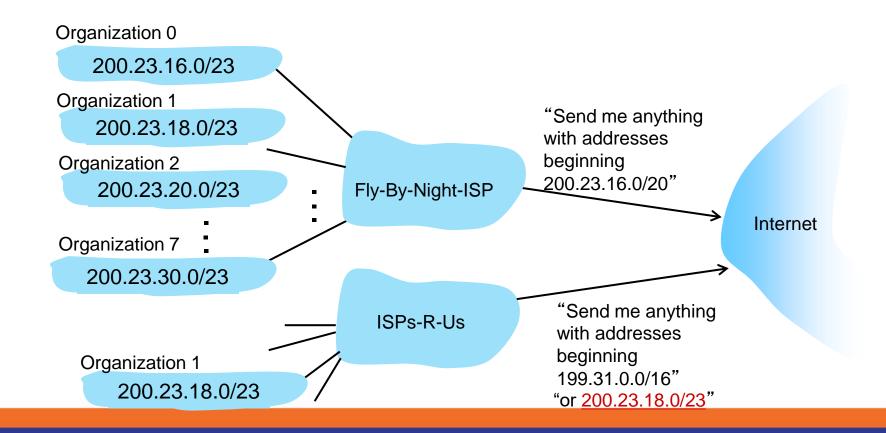






Hierarchical addressing: more specific routes

- Organization 1 moves from Fly-By-Night-ISP to ISPs-R-Us
- ISPs-R-Us now advertises a more specific route to Organization 1

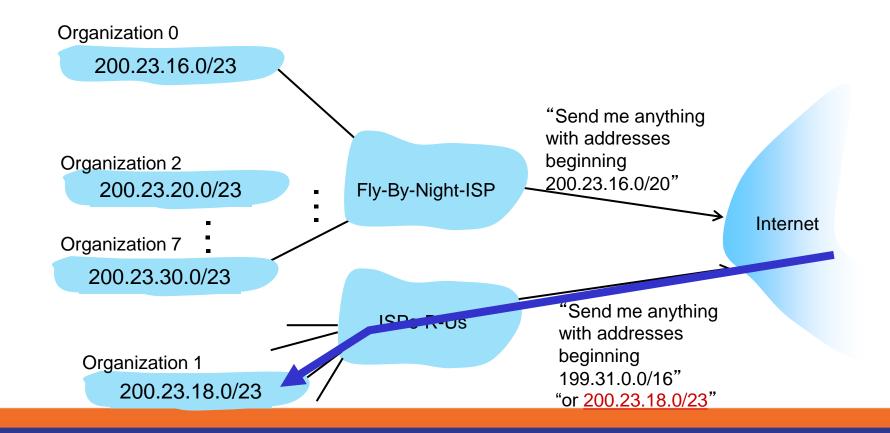






Hierarchical addressing: more specific routes

- Organization 1 moves from Fly-By-Night-ISP to ISPs-R-Us
- ISPs-R-Us now advertises a more specific route to Organization 1







IP addressing: last words ...

Q: how does an ISP get block of addresses?

A: ICANN: Internet Corporation for Assigned Names and Numbers http://www.icann.org/

- allocates IP addresses, through 5
 regional registries (RRs) (who may
 then allocate to local registries)
- manages DNS root zone, including delegation of individual TLD (.com, .edu, ...) management

Q: are there enough 32-bit IP addresses?

- ICANN allocated last chunk of IPv4 addresses to RRs in 2011
- NAT (next) helps IPv4 address space exhaustion
- IPv6 has 128-bit address space

"Who the hell knew how much address space we needed?" Vint Cerf (reflecting on decision to make IPv4 address 32 bits long)





Pearson Network layer: "data plane" roadmap

- Network layer: overview
 - data plane
 - control plane
- What's inside a router
 - input ports, switching, output ports
 - buffer management, scheduling
- IP: the Internet Protocol
 - datagram format
 - addressing

IPv6

- network address translation

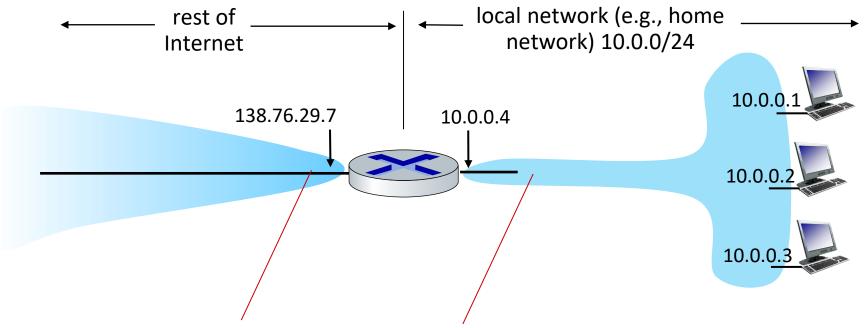


- Generalized Forwarding, SDN
 - match+action
 - OpenFlow: match+action in action
- Middleboxes





NAT: all devices in local network share just one IPv4 address as far as outside world is concerned



all datagrams *leaving* local network have *same* source NAT IP address: 138.76.29.7, but *different* source port numbers

datagrams with source or destination in this network have 10.0.0/24 address for source, destination (as usual)





- all devices in local network have 32-bit addresses in a "private" IP address space (10/8, 172.16/12, 192.168/16 prefixes) that can only be used in local network
- advantages:
 - just one IP address needed from provider ISP for all devices
 - can change addresses of host in local network without notifying outside world
 - can change ISP without changing addresses of devices in local network
 - security: devices inside local net not directly addressable, visible by outside world



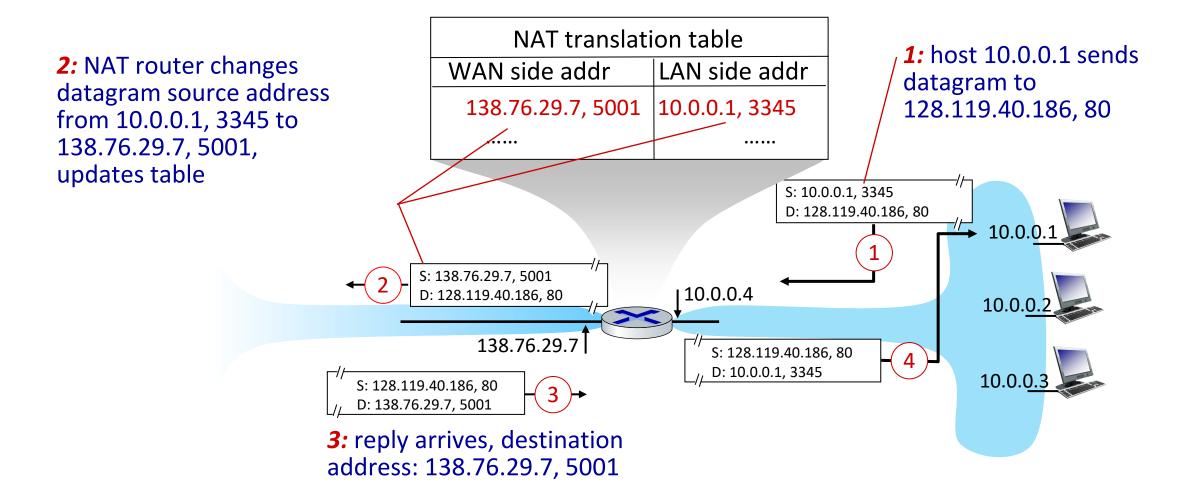


implementation: NAT router must (transparently):

- outgoing datagrams: replace (source IP address, port #) of every outgoing datagram to (NAT IP address, new port #)
 - remote clients/servers will respond using (NAT IP address, new port
 #) as destination address
- remember (in NAT translation table) every (source IP address, port #) to (NAT IP address, new port #) translation pair
- incoming datagrams: replace (NAT IP address, new port #) in destination fields of every incoming datagram with corresponding (source IP address, port #) stored in NAT table











- NAT has been controversial:
 - routers "should" only process up to layer 3
 - address "shortage" should be solved by IPv6
 - violates end-to-end argument (port # manipulation by network-layer device)
 - NAT traversal: what if client wants to connect to server behind NAT?
- but NAT is here to stay:
 - extensively used in home and institutional nets, 4G/5G cellular nets





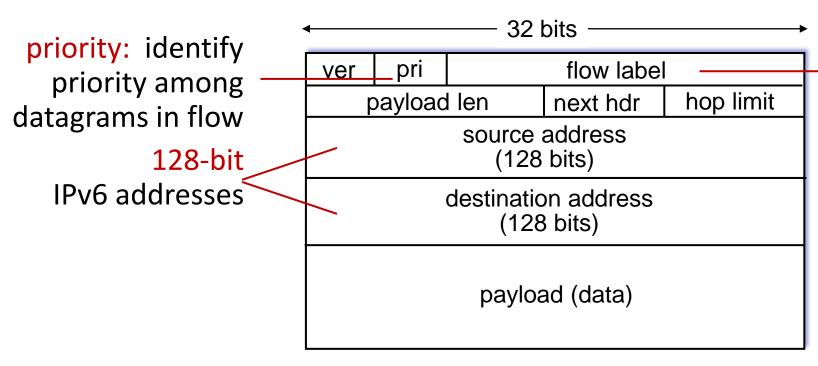
IPv6: motivation

- initial motivation: 32-bit IPv4 address space would be completely allocated
- additional motivation:
 - speed processing/forwarding: 40-byte fixed length header
 - enable different network-layer treatment of "flows"





IPv6 datagram format



flow label: identify datagrams in same "flow." (concept of "flow" not well defined).

What's missing (compared with IPv4):

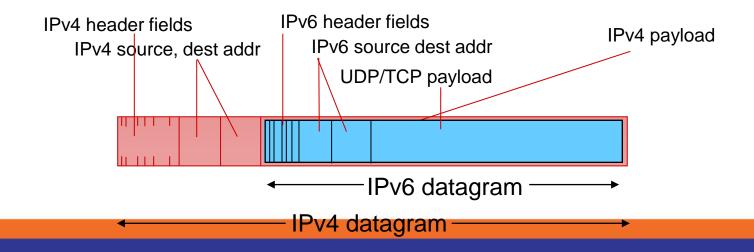
- no checksum (to speed processing at routers)
- no fragmentation/reassembly
- no options (available as upper-layer, next-header protocol at router)





Transition from IPv4 to IPv6

- not all routers can be upgraded simultaneously
 - no "flag days"
 - how will network operate with mixed IPv4 and IPv6 routers?
- tunneling: IPv6 datagram carried as payload in IPv4 datagram among IPv4 routers ("packet within a packet")
 - tunneling used extensively in other contexts (4G/5G)

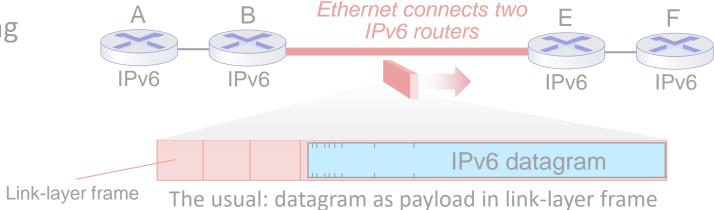




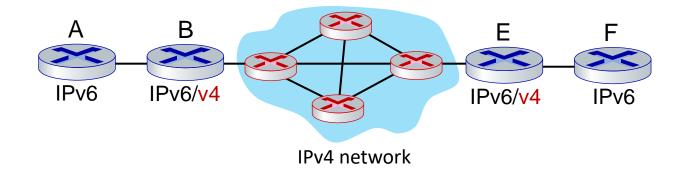


Tunneling and encapsulation

Ethernet connecting two IPv6 routers:



IPv4 network connecting two IPv6 routers

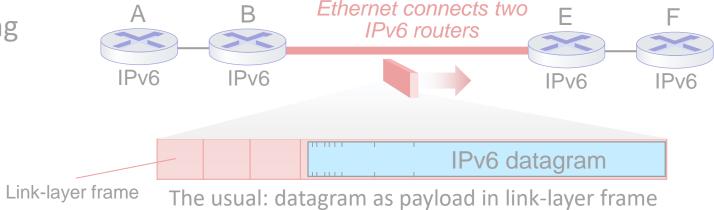




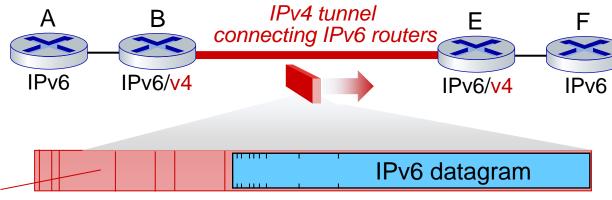


Tunneling and encapsulation

Ethernet connecting two IPv6 routers:



IPv4 tunnel connecting two IPv6 routers



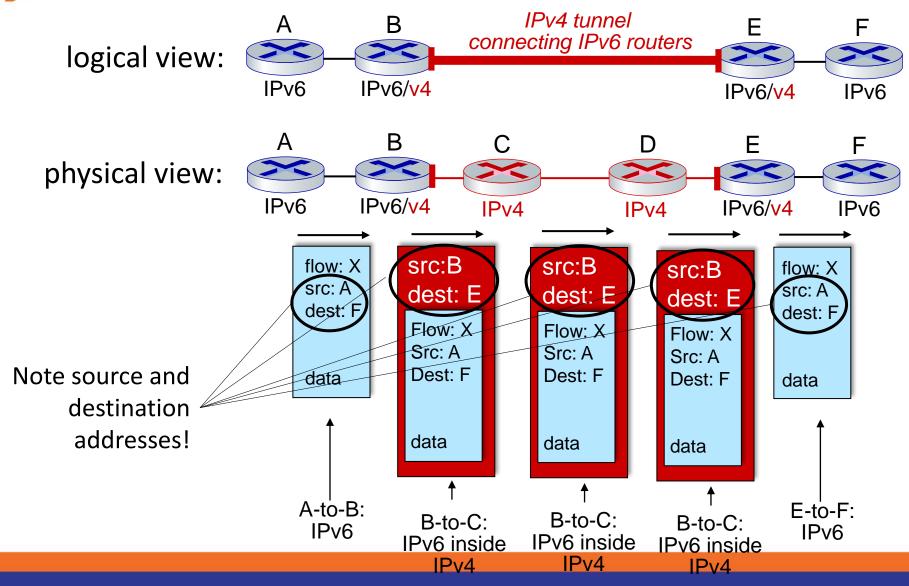
IPv4 datagram

tunneling: IPv6 datagram as payload in a IPv4 datagram





Tunneling





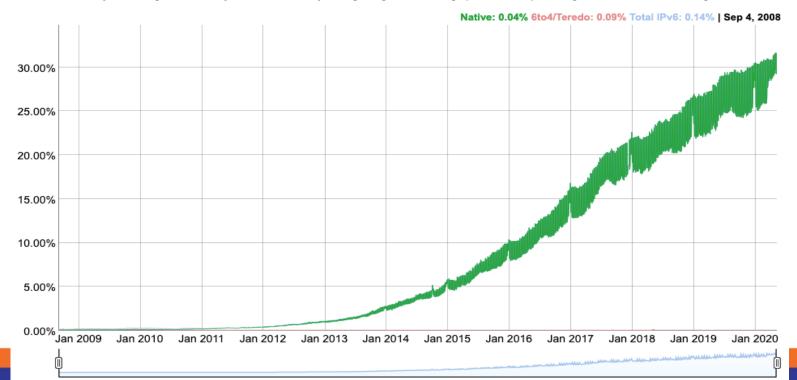


IPv6: adoption

- Google¹: ~ 30% of clients access services via IPv6
- NIST: 1/3 of all US government domains are IPv6 capable

IPv6 Adoption

We are continuously measuring the availability of IPv6 connectivity among Google users. The graph shows the percentage of users that access Google over IPv6.



https://www.google.com/intl/en/ipv6/statistics.html





IPv6: adoption

- Google¹: ~ 30% of clients access services via IPv6
- NIST: 1/3 of all US government domains are IPv6 capable
- Long (long!) time for deployment, use
 - 25 years and counting!
 - think of application-level changes in last 25 years: WWW, social media, streaming media, gaming, telepresence, ...
 - Why?





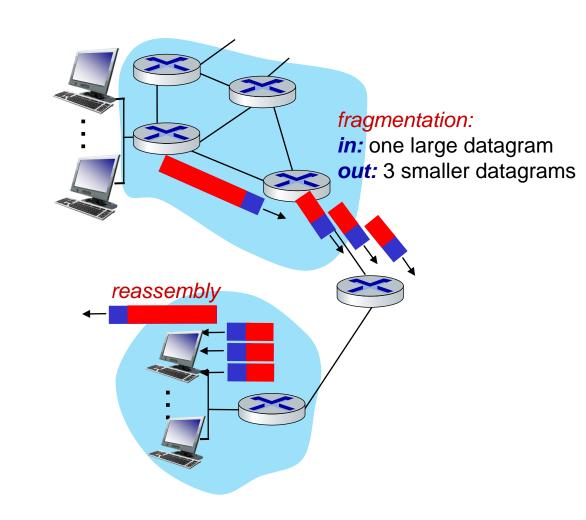
Additional Chapter 4 slides





IP fragmentation/reassembly

- network links have MTU (max. transfer size) - largest possible link-level frame
 - different link types, different MTUs
- large IP datagram divided ("fragmented") within net
 - one datagram becomes several datagrams
 - "reassembled" only at destination
 - IP header bits used to identify, order related fragments







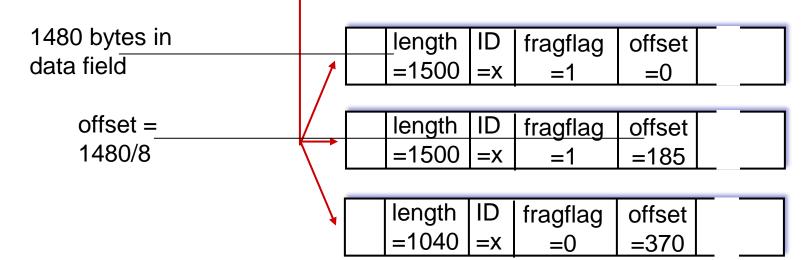
IP fragmentation/reassembly

example:

- 4000 byte datagram
- MTU = 1500 bytes



one large datagram becomes several smaller datagrams







DHCP: Wireshark output (home LAN)

Message type: **Boot Request (1)**

Hardware type: Ethernet Hardware address length: 6

Hops: 0

request

Transaction ID: 0x6b3a11b7

Seconds elapsed: 0

Bootp flags: 0x0000 (Unicast) Client IP address: 0.0.0.0 (0.0.0.0)

Your (client) IP address: 0.0.0.0 (0.0.0.0) Next server IP address: 0.0.0.0 (0.0.0.0) Relay agent IP address: 0.0.0.0 (0.0.0.0)

Client MAC address: Wistron 23:68:8a (00:16:d3:23:68:8a)

Server host name not given Boot file name not given

Magic cookie: (OK)

Option: (t=53,l=1) **DHCP Message Type = DHCP Request**

Option: (61) Client identifier

Length: 7; Value: 010016D323688A;

Hardware type: Ethernet

Client MAC address: Wistron_23:68:8a (00:16:d3:23:68:8a)

Option: (t=50,l=4) Requested IP Address = 192.168.1.101

Option: (t=12,l=5) Host Name = "nomad" **Option: (55) Parameter Request List**

Length: 11; Value: 010F03062C2E2F1F21F92B

1 = Subnet Mask; 15 = Domain Name 3 = Router: 6 = Domain Name Server 44 = NetBIOS over TCP/IP Name Server Message type: Boot Reply (2)

Hardware type: Ethernet Hardware address length: 6

reply Hops: 0

Transaction ID: 0x6b3a11b7

Seconds elapsed: 0

Bootp flags: 0x0000 (Unicast)

Client IP address: 192.168.1.101 (192.168.1.101)

Your (client) IP address: 0.0.0.0 (0.0.0.0)

Next server IP address: 192.168.1.1 (192.168.1.1)

Relay agent IP address: 0.0.0.0 (0.0.0.0)

Client MAC address: Wistron 23:68:8a (00:16:d3:23:68:8a)

Server host name not given Boot file name not given Magic cookie: (OK)

Option: (t=53,l=1) DHCP Message Type = DHCP ACK

Option: (t=54,l=4) **Server Identifier** = 192.168.1.1 Option: (t=1,l=4) Subnet Mask = 255.255.255.0

Option: (t=3,l=4) Router = 192.168.1.1

Option: (6) Domain Name Server

Length: 12; Value: 445747E2445749F244574092:

IP Address: 68.87.71.226; IP Address: 68.87.73.242; IP Address: 68.87.64.146

Option: (t=15,l=20) Domain Name = "hsd1.ma.comcast.net."