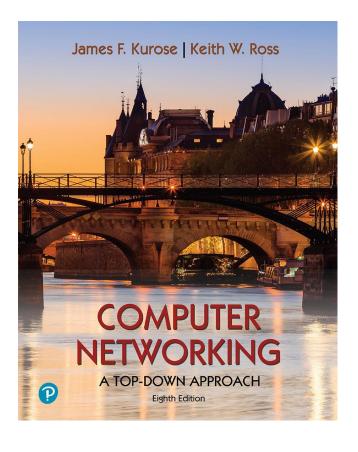
Chapter 4 Network Layer: Data Plane



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

Network layer: "data plane" roadmap

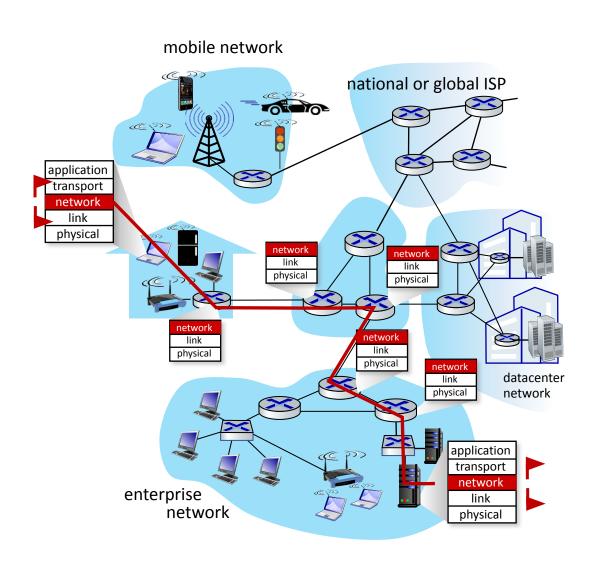
- Network layer: overview
 - data plane
 - control plane
- What's inside a router
 - input ports, switching, output ports
 - buffer management, scheduling
- IP: the Internet Protocol
 - datagram format
 - addressing
 - network address translation
 - IPv6



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

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



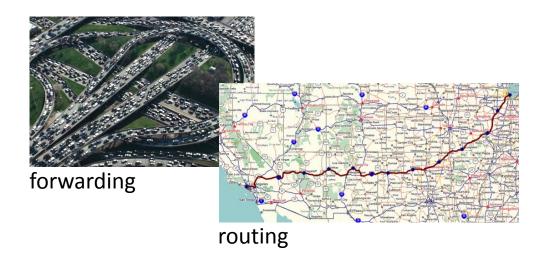
Two key network-layer functions

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

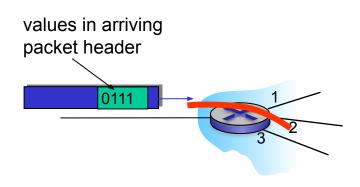
- forwarding: process of getting through single interchange
- routing: process of planning trip from source to destination



Network layer: data plane, control plane

Data plane:

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

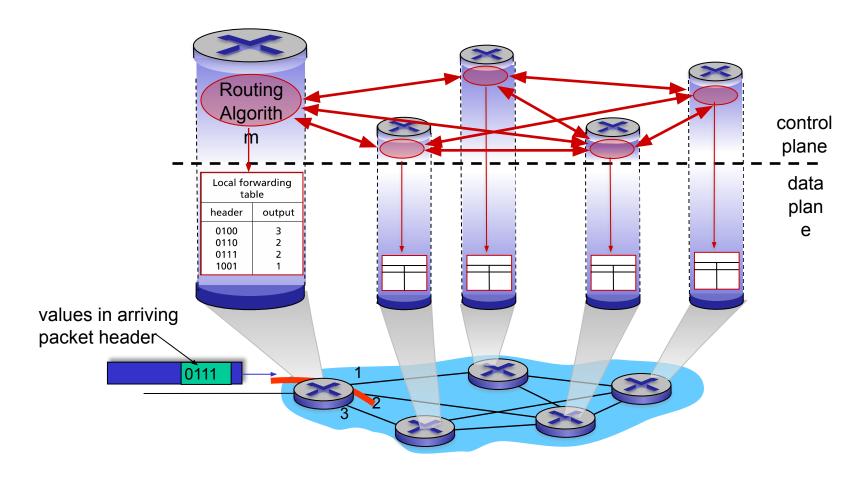


Control plane

- network-wide logic
- determines how datagram is routed among routers along end-end 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

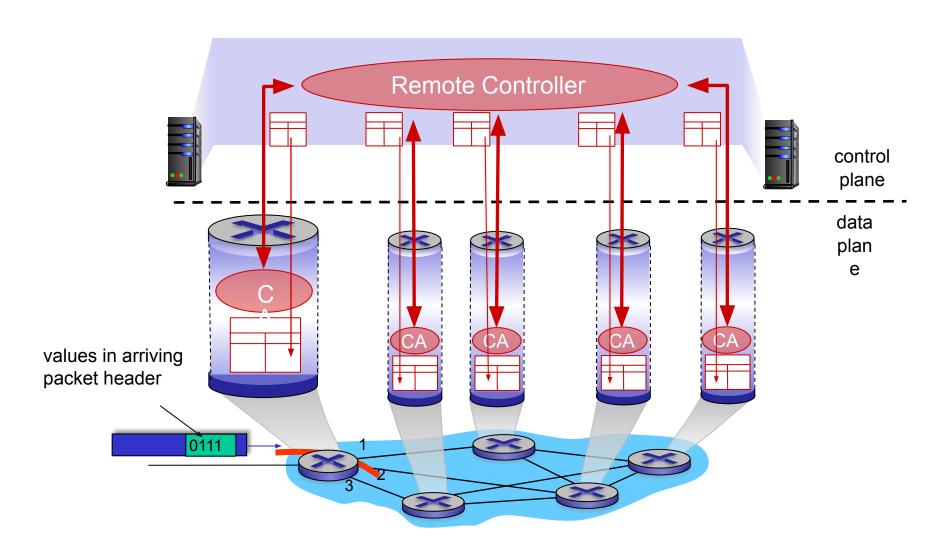
Per-router control plane

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



Software-Defined Networking (SDN) control plane

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 inter-packet spacing

Network-layer service model

Network Architecture		Service	Quality of Service (QoS) Guarantees?				
		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 Architecture		Service	Quality of Service (QoS) Guarantees ?				
		Model	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	

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

Network layer: "data plane" roadmap

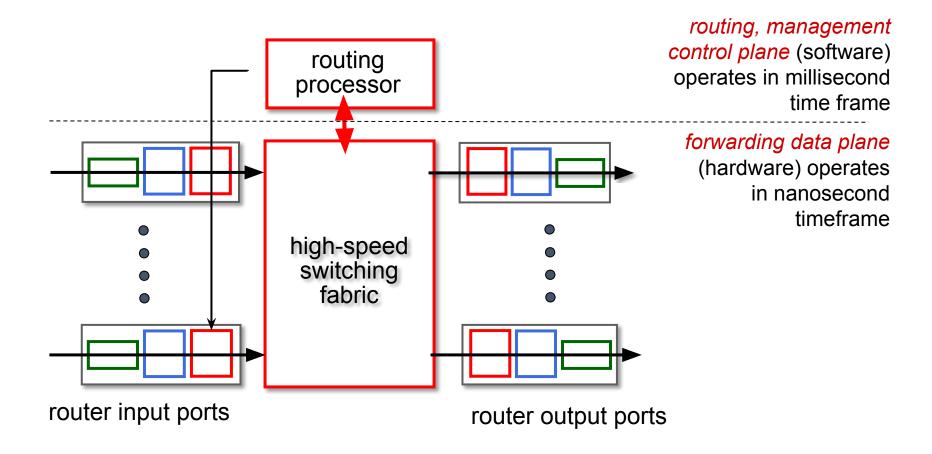
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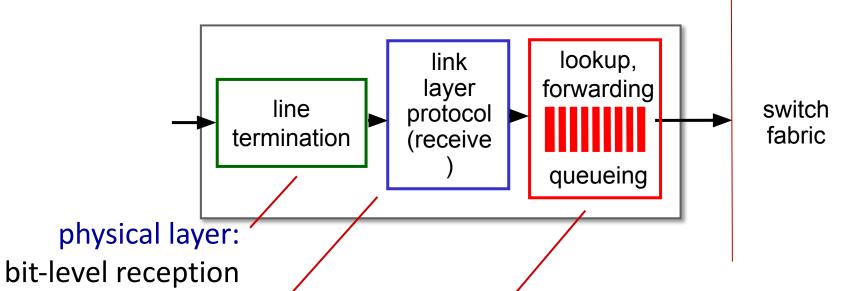
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Router architecture overview

high-level view of generic router architecture:



Input port functions



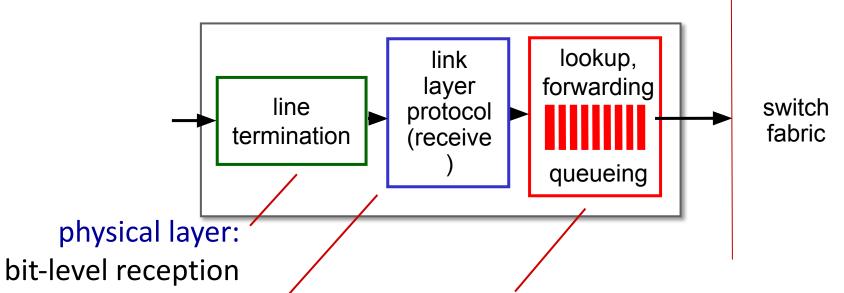
link layer:

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

Destinatio	n Address R	arding table ——	Link Interface	
11001000 through	00010111	00010000	0000000	0
11001000 through	00010111	000 <mark>10000</mark>	00000100	3
11001000	00010111	00010000	00000111	
11001000	00010111	00011000	11111111	
11001000 through	00010111	00011001	00000000	2
	00010111	00011111	11111111	
otherwise				3

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:

which interface?	10100001	00010110	00010111	11001000
which interface?	10101010	00011000	00010111	11001000

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	10100001	which interface?

examples:

11001000 00010111 00010 110 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.

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

examples:

longest prefix match

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

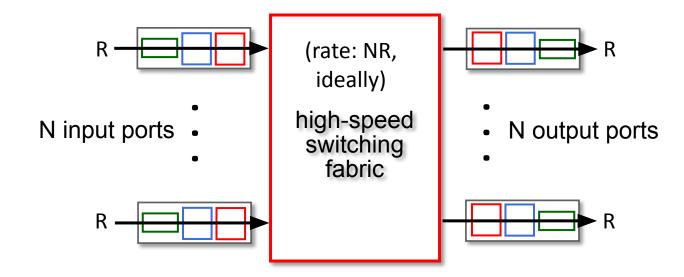
Destination A	Link interface			
11001000	00010111	00010***	*****	0
11001000	00010111	00011000	*****	1
11001000	0000111	00011***	*****	2
otherwise	match!			3

examples:

- 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

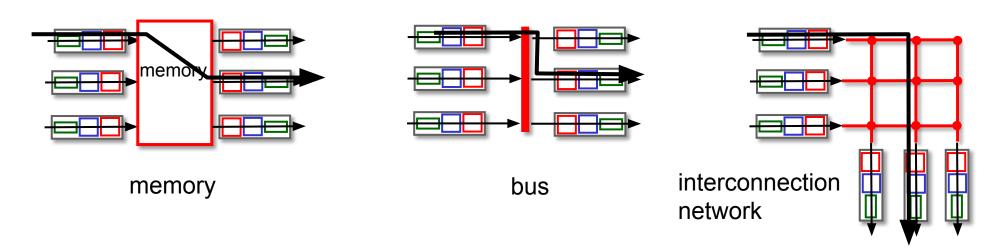
Switching fabrics

- 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

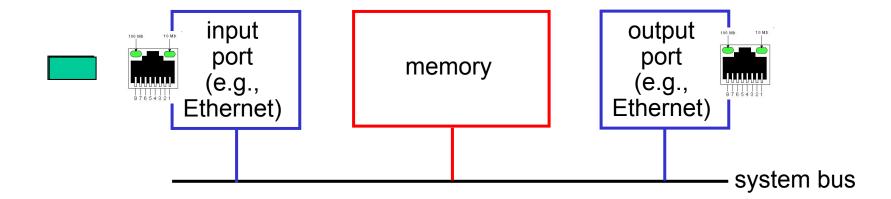
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 - N inputs: switching rate N times line rate desirable
- three major types of switching fabrics:



Switching via memory

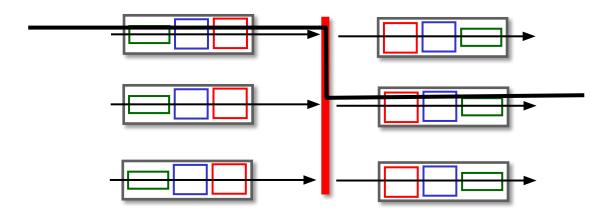
first generation routers:

- 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)



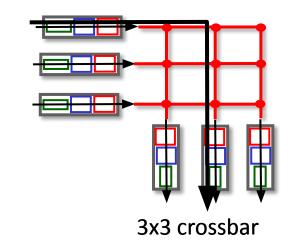
Switching via a bus

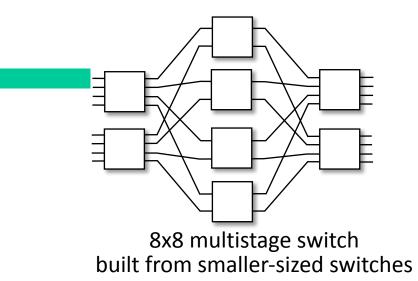
- datagram from input port memory to output port memory via a shared bus
- *bus contention: switching speed limited by bus bandwidth
- •32 Gbps bus, Cisco 5600: sufficient speed for access routers



Switching via interconnection network

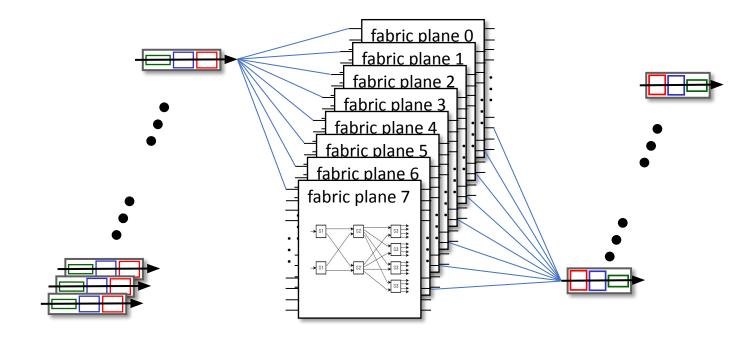
- Crossbar, Clos networks, other interconnection nets initially developed to connect processors in multiprocessor
- multistage switch: nxn switch from multiple stages of smaller switches
- exploiting parallelism:
 - fragment datagram into fixed length cells on entry
 - switch cells through the fabric,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





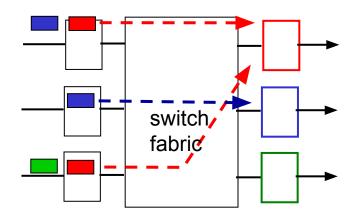
Switching via interconnection network

- 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

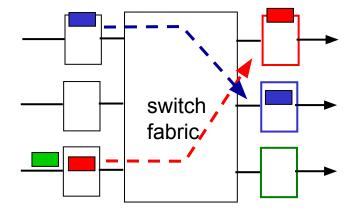


Input port queuing

- 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

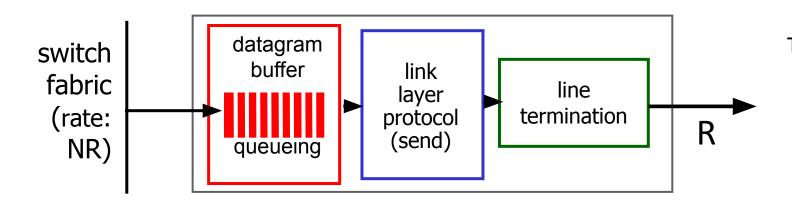


output port contention: only one red datagram can be transferred. lower red packet is *blocked*



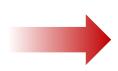
one packet time later: green packet experiences HOL blocking

Output port queuing





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



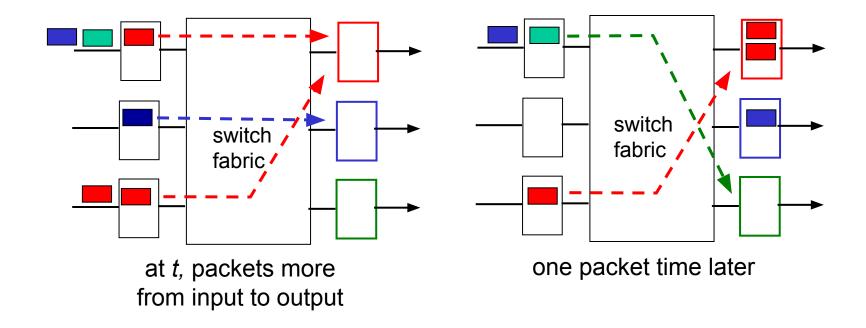
Datagrams can be lost due to congestion, lack of buffers

 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!

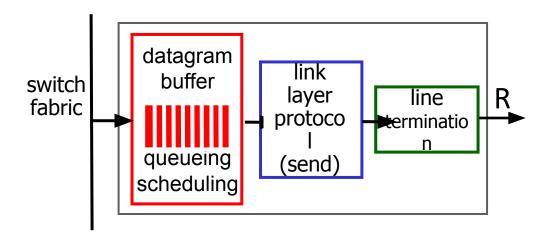
How much buffering?

- 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

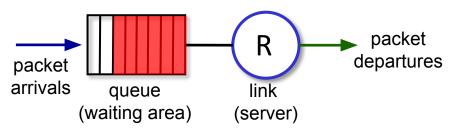
$$\frac{\mathsf{RTT} \cdot \mathsf{C}}{\sqrt{\mathsf{N}}}$$

- 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"

Buffer Management



Abstraction: queue



buffer management:

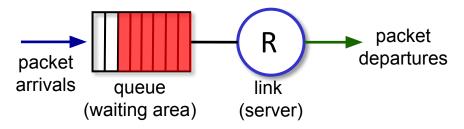
- drop: which packet to add, drop when buffers are full
 - tail drop: drop arriving packet
 - priority: drop/remove on priority basis
- marking: which packets to mark to signal congestion (ECN, RED)

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



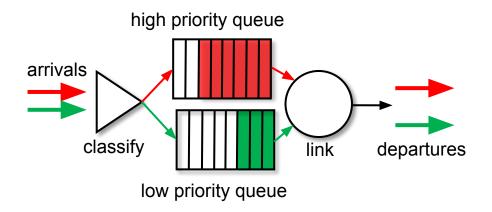
FCFS: packets transmitted in order of arrival to output port

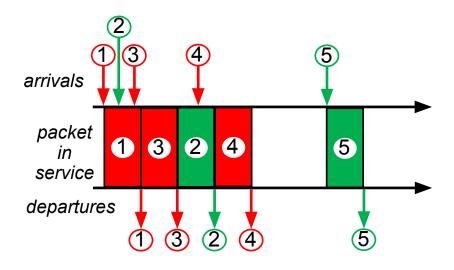
- also known as:First-in-first-out (FIFO)
- real world examples?

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

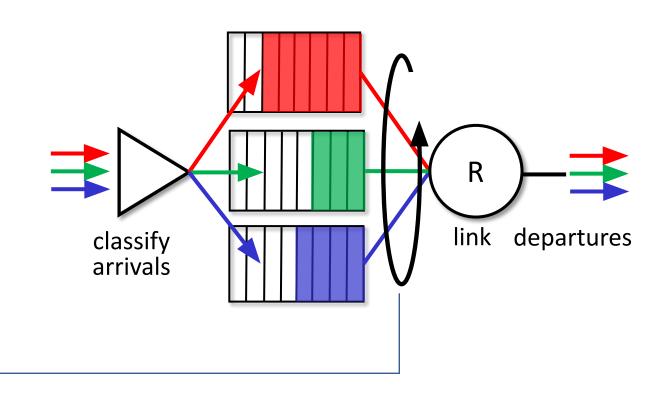




Scheduling policies: round robin

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



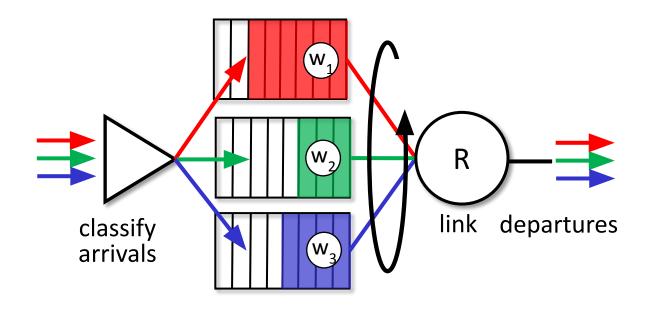
Scheduling policies: weighted fair queueing

Weighted Fair Queuing (WFQ):

- generalized Round Robin
- each class, i, has weight, w, 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".

Network layer: "data plane" roadmap

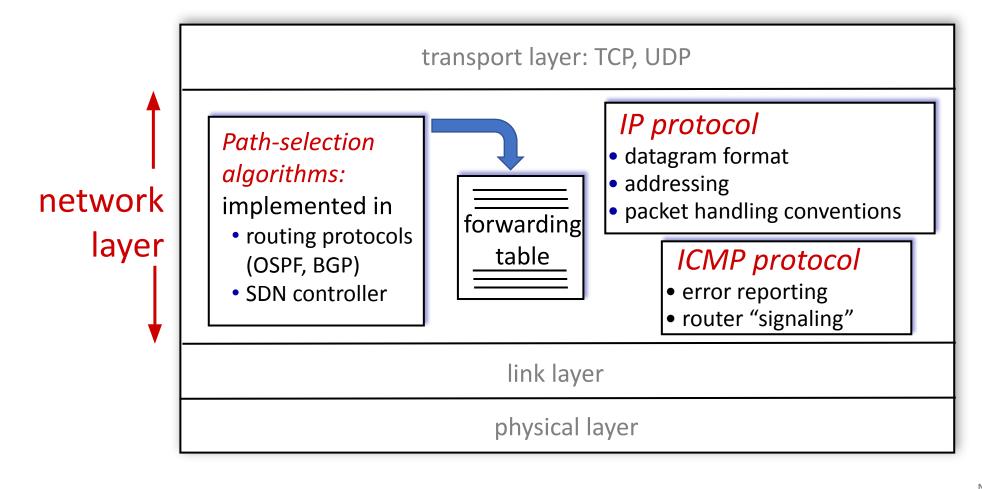
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Network Layer: Internet

host, router network layer functions:



IP Datagram format

IP protocol version number head. header length(bytes) Jen "type" of service: diffserv (0:5) time to • ECN (6:7) live 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

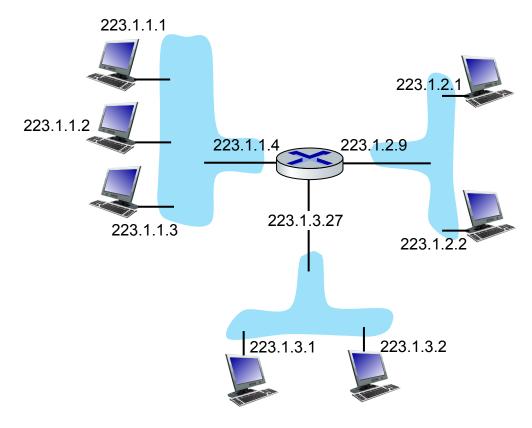
TCP+IP

layer overhead for

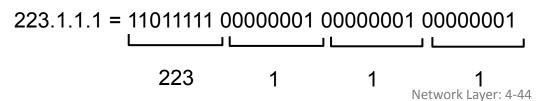
32 bits total datagram type of length (bytes) length service fragment. fragmentation/ 16-bit identifier | flgs offset reassembly upper header header checksum layer checksum 32-bit source IP address source IP address Maximum length: 64K bytes destination IP address Typically: 1500 bytes or less options (if any) e.g., timestamp, record route taken payload data (variable length, typically a TCP or UDP segment) Network Layer: 4-43

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)

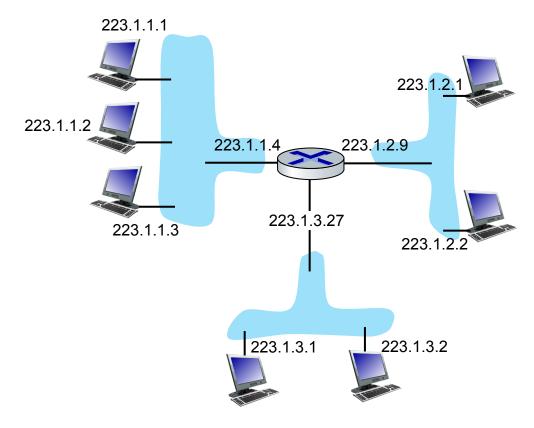


dotted-decimal IP address notation:



IP addressing: introduction

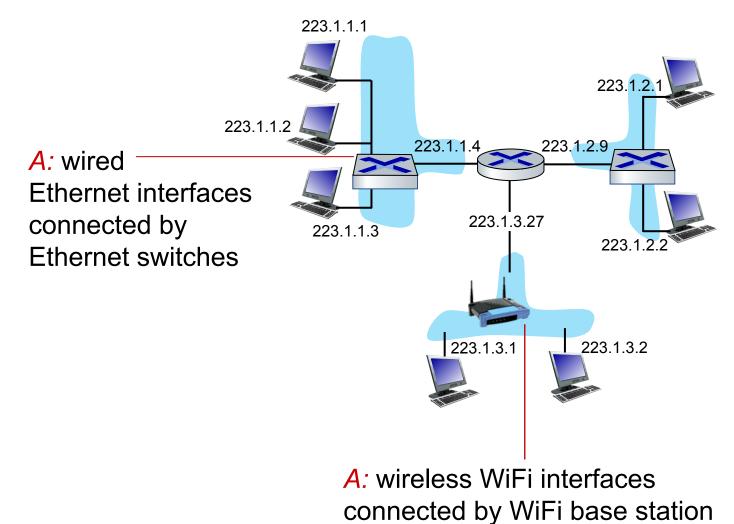
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IP addressing: introduction



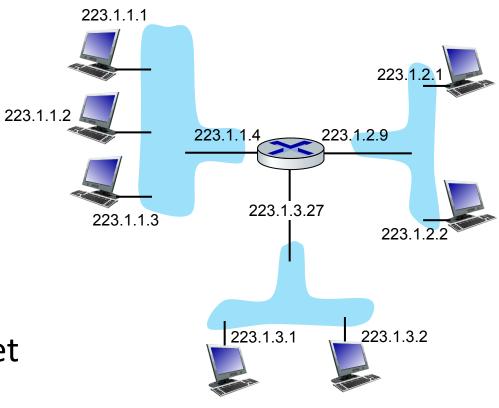
Subnets

•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

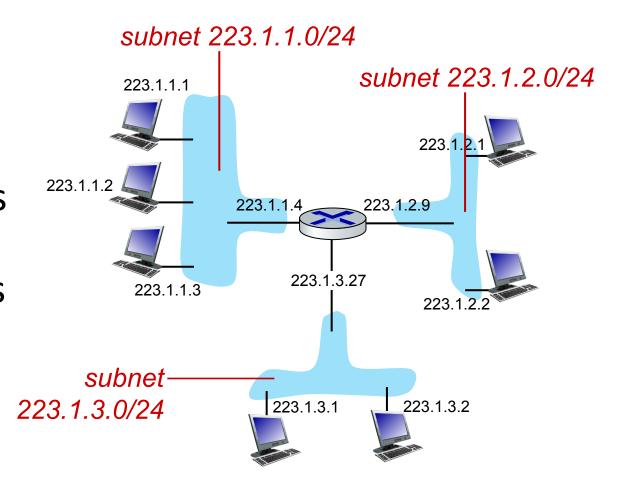


network consisting of 3 subnets

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*

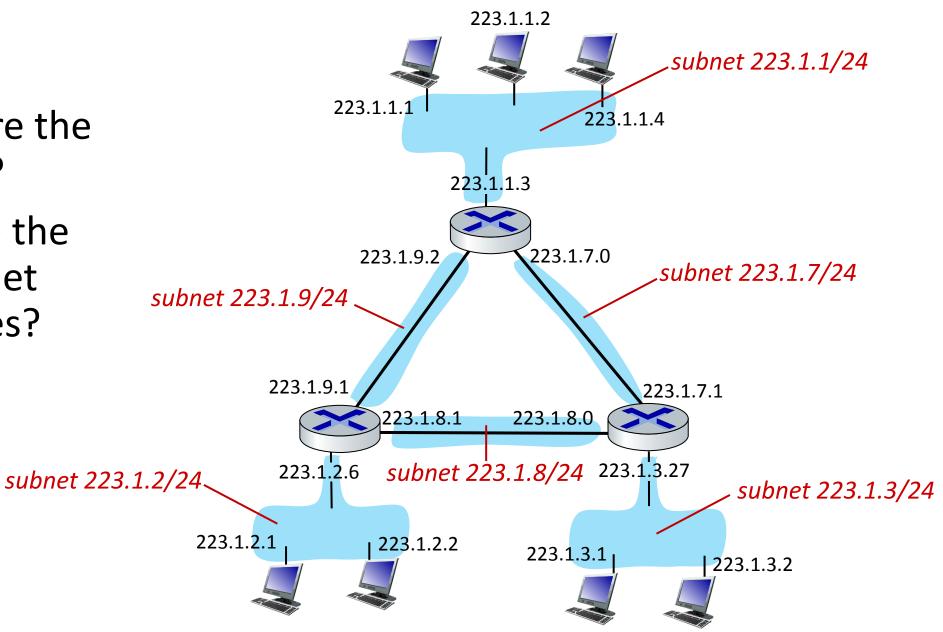


subnet mask: /24

(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



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

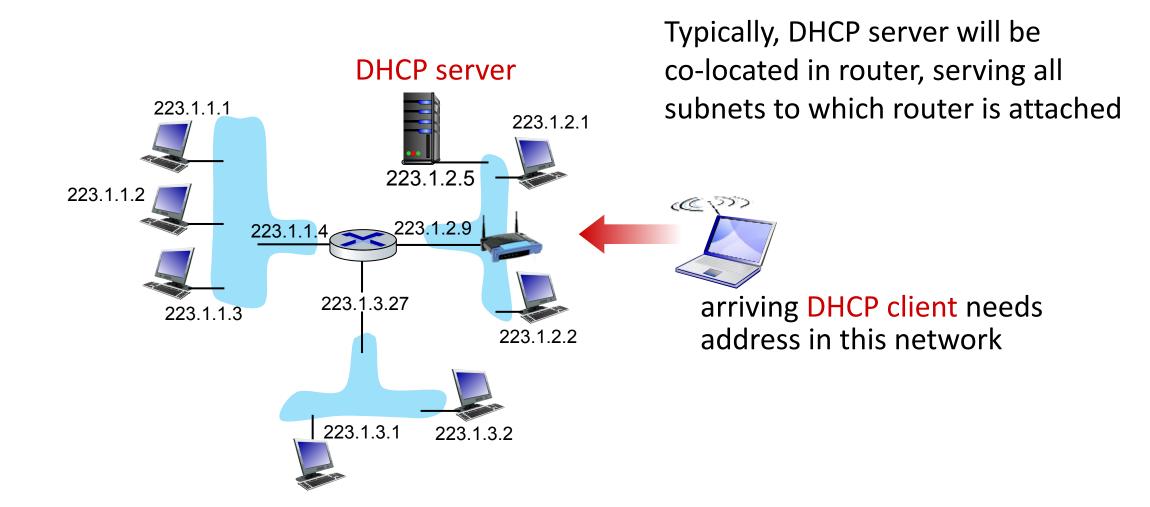
goal: host dynamically obtains IP address from network server when it "joins" network

- 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

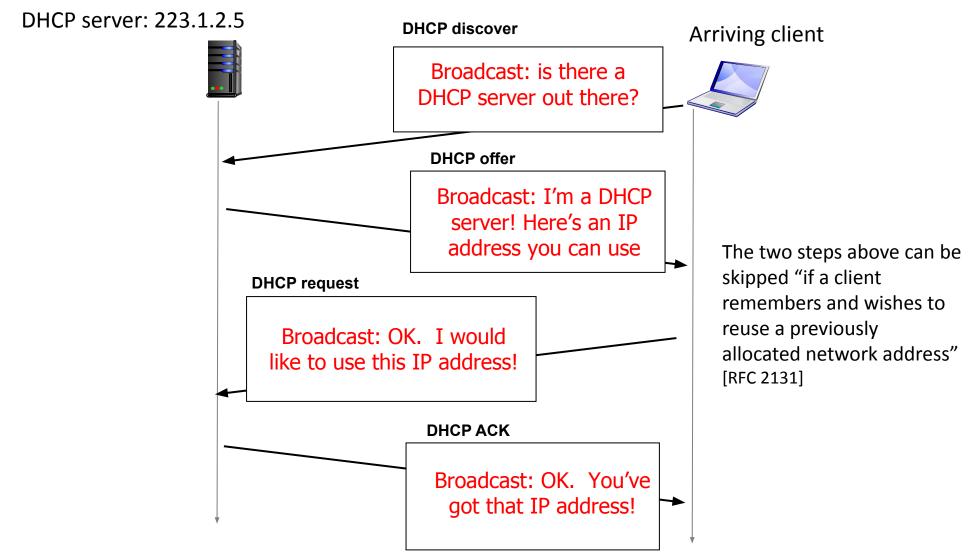
DHCP overview:

- 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

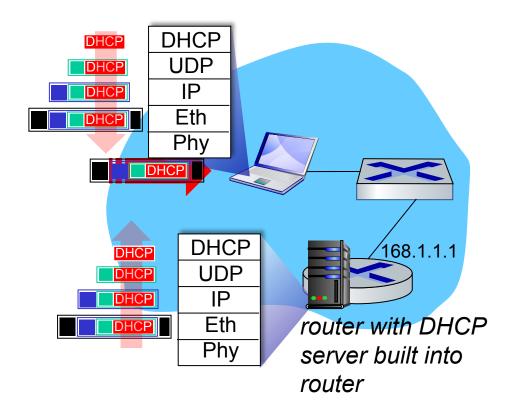


DHCP: more than IP addresses

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

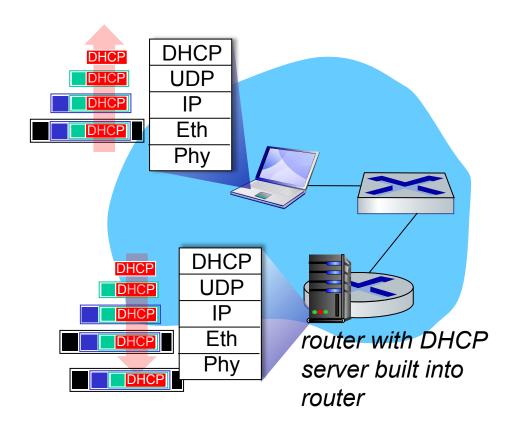
- address of first-hop router for client
- name and IP address of DNS sever
- network mask (indicating network versus host portion of address)

DHCP: example



- Connecting laptop will use DHCP to get IP address, address of first-hop 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: 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 <u>11001000 00010111 0001</u>0000 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

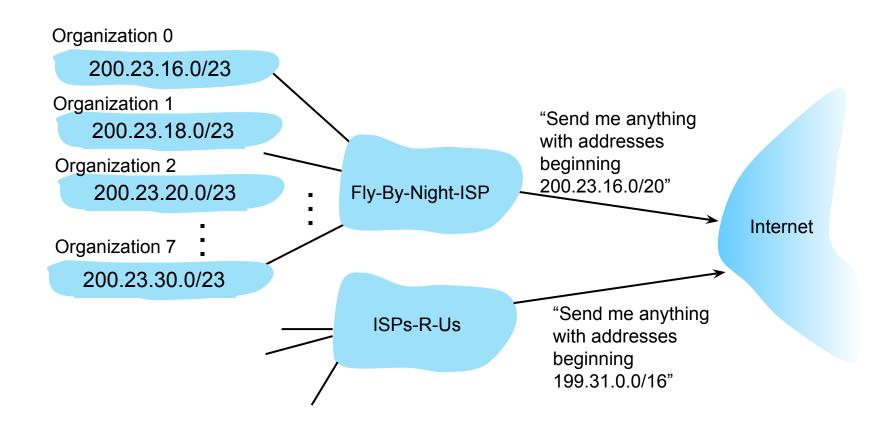
        Organization 1
        11001000 00010111 0001001
        00000000
        200.23.18.0/23

        Organization 2
        11001000 00010111 0001010
        00000000
        200.23.20.0/23
```

Organization 7 11001000 00010111 00011110 00000000 200.23.30.0/23

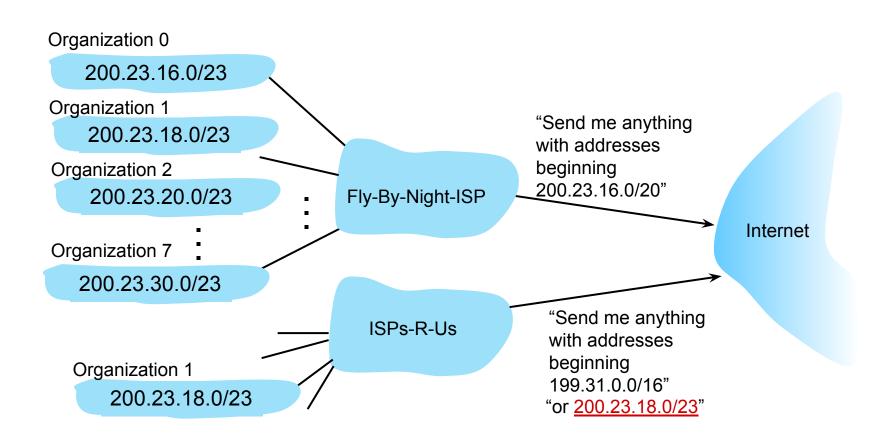
Hierarchical addressing: route

aggregationhierarchical addressing allows efficient advertisement of routing information:



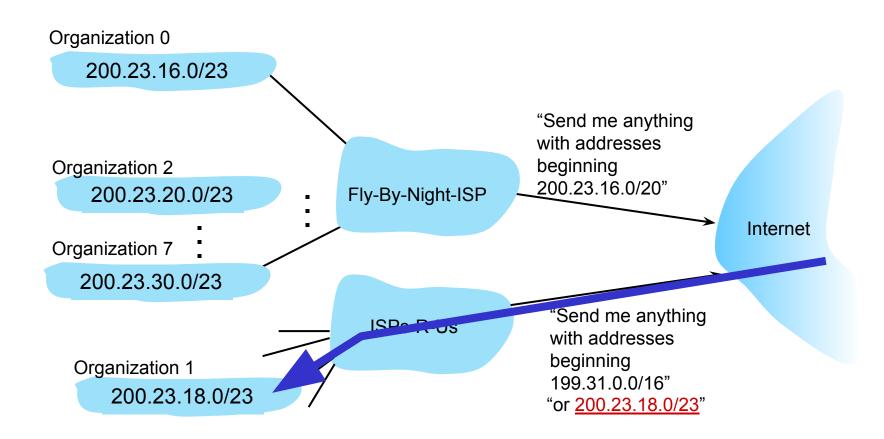
Hierarchical addressing: more specific routes

- 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
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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)

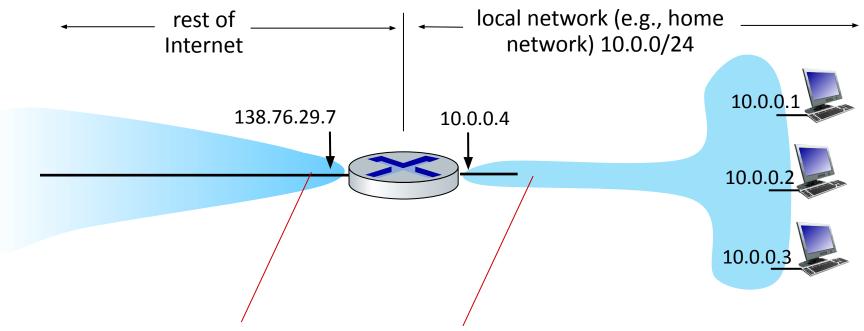
Network layer: "data plane" roadmap

- 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

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



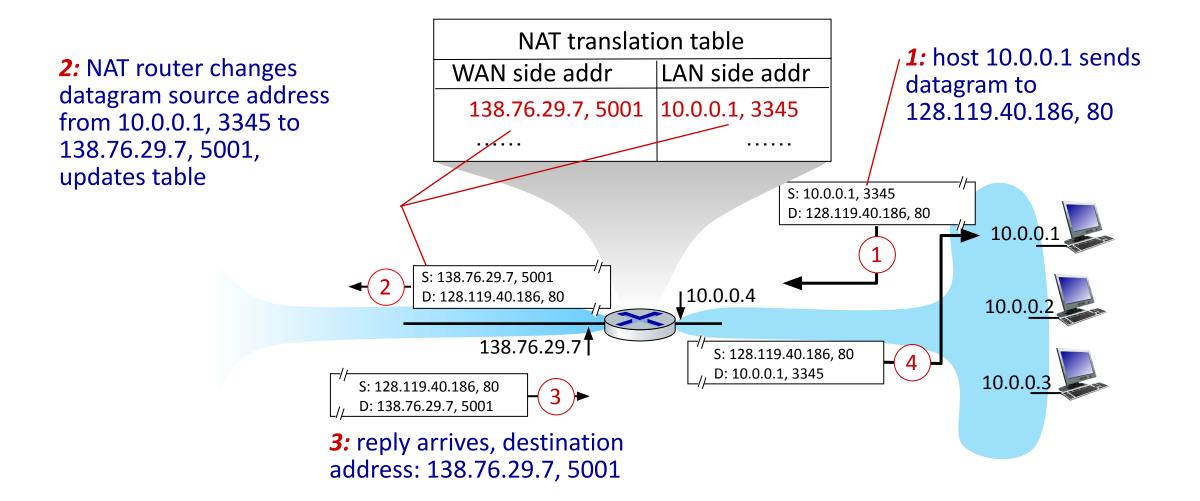
all datagrams leaving local network have same source NAT IP address: 138.76.29.7, but different source port numbers

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