Chapter 4 Network Layer: Data Plane

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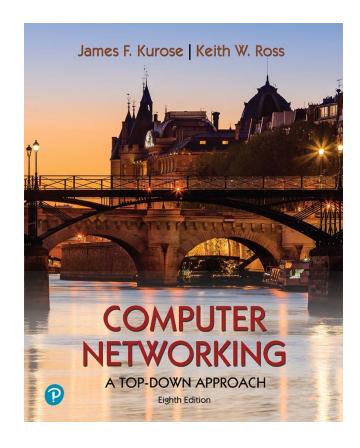
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Computer Networking: A Top-Down Approach

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

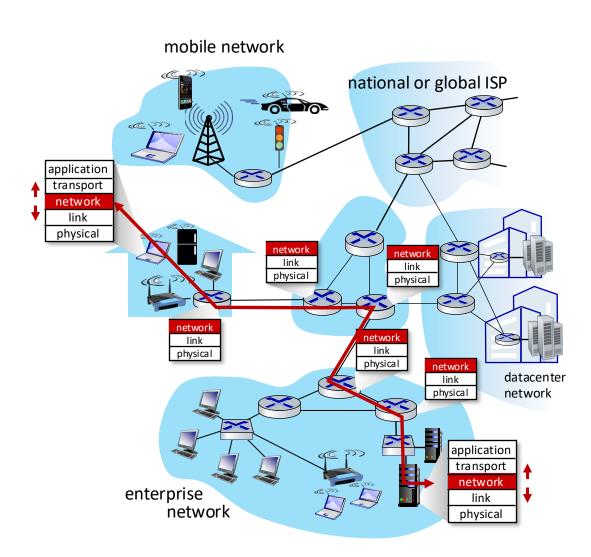
- Network layer: overview
 - data plane
 - control plane
- What's inside a router
 - input ports, switching, output ports
 - buffer management, scheduling
- IP: the Internet Protocol
 - datagram format
 - addressing
 - network address translation (NAT)
 - IPv6



- 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



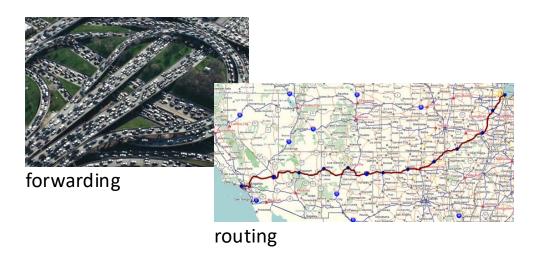
Two key network-layer functions

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

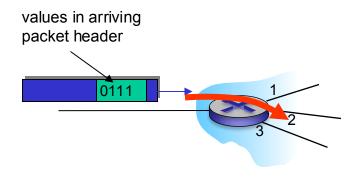
- forwarding: process of getting through single interchange
- routing: process of planning trip from source to destination



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



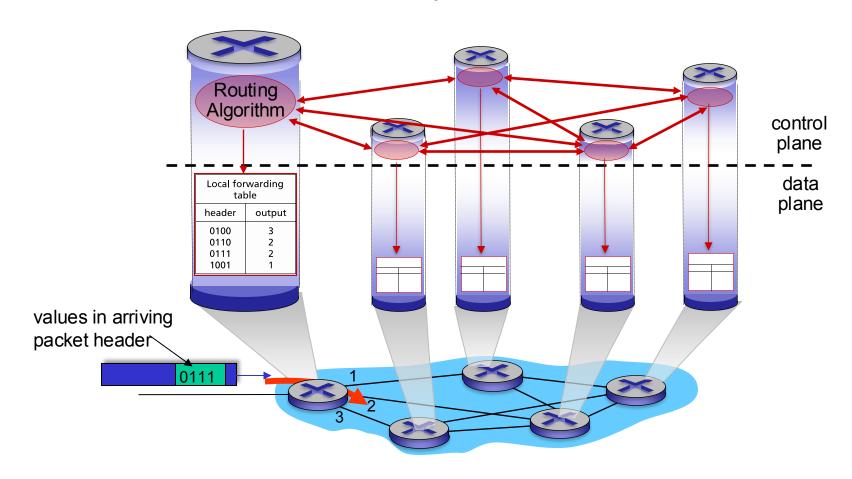
Control plane

- 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

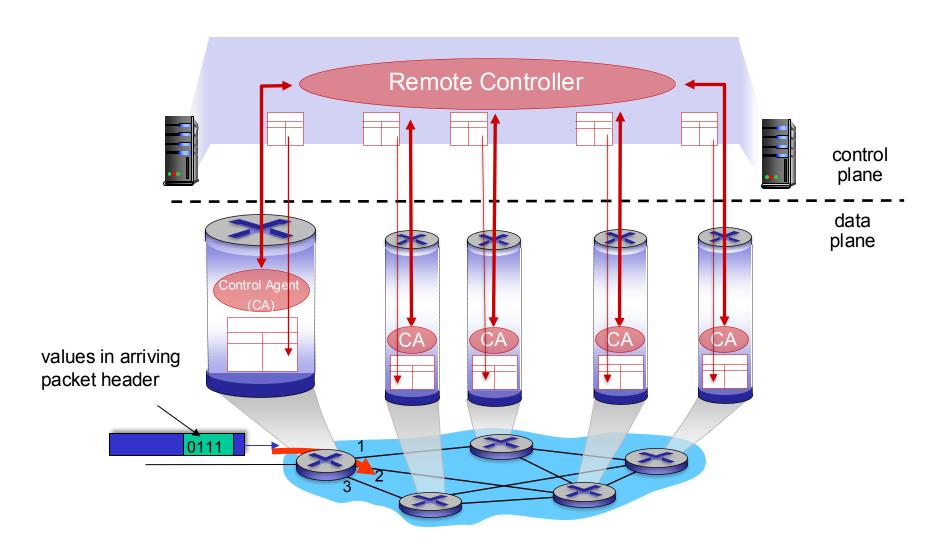
Per-router control plane

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 bounded delay?

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 Model	Quality of Service (QoS) Guarantees?				
			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 (data centers, 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

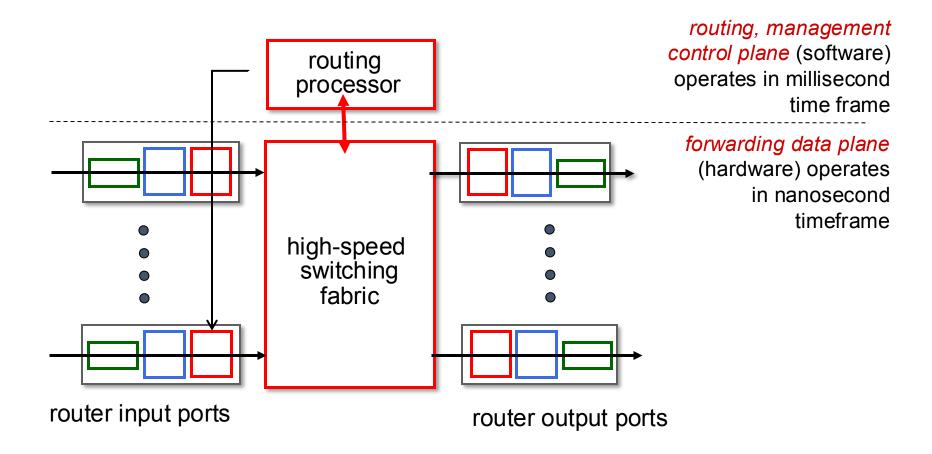
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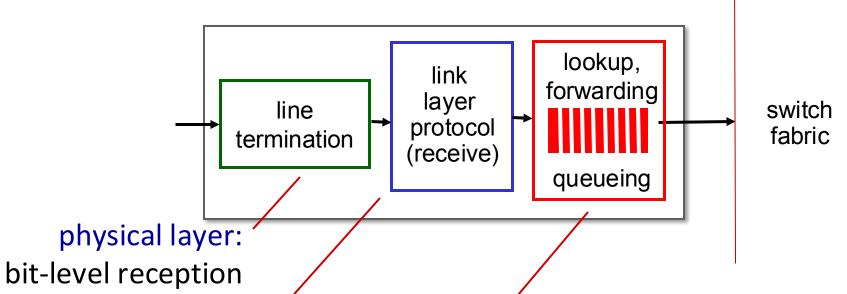
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 - Match+action
 - OpenFlow: match+action in action
- Middleboxes

Router architecture overview

high-level view of generic router architecture:



Input port functions



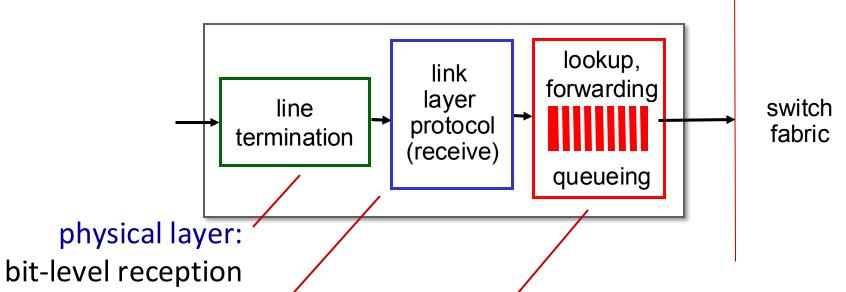
link layer:

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 (transmission rates)'
- 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

Destination Address Range	Link Interface
11001000 00010111 000 <mark>10000 00000000</mark> through	0
11001000 00010111 000 <mark>10111 11111111</mark>	
11001000 00010111 000 <mark>11000 00000000</mark> through	1
11001000 00010111 000 <mark>11000 11111111</mark>	
11001000 00010111 000 <mark>11001 00000000</mark> through	2
11001000 00010111 000 <mark>11111 11111111</mark>	
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 .	Link interface					
11001000	00010111	00010***	*****	0		
11001000	00010111	00011000	*****	1		
11001000	00010111	00011***	*****	2		
otherwise	otherwise					

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 Add	Link interface		
11001000 00	0010111 0001	0 * * * * * * * *	** 0
11001000 00	0111 0001	1000 ****	** 1
11001000 m	atch! 1 0001	1*** *****	** 2
otherwise			3

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.

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

examples:

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

longest prefix match

11001000

11001000

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

00010111

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

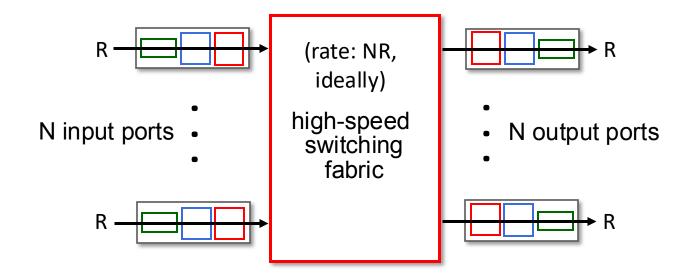
examples:

00011000 10101010 which interface?

which interface?

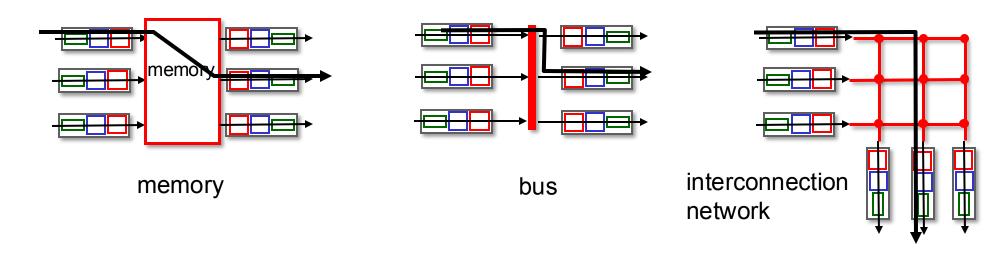
Switching fabrics

- transfer packet: input link -> 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
 - Suppose N inputs: Ideal switching rate >= N times the line rate



Switching fabrics

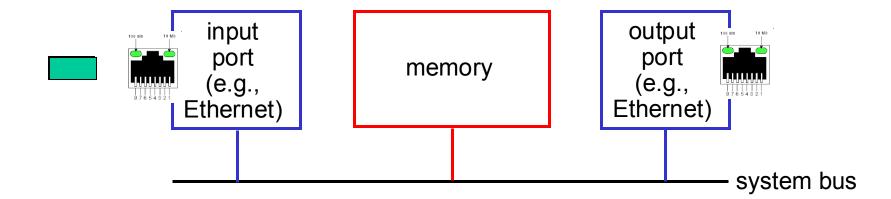
- transfer packet from input link to appropriate output link
- switching rate: rate at which packets can be transfer from inputs to outputs
 - often measured as multiple of input/output line rate
 - Suppose N inputs: Ideal switching rate >= N times the line rate
- three major types of switching fabrics:



Switching via memory

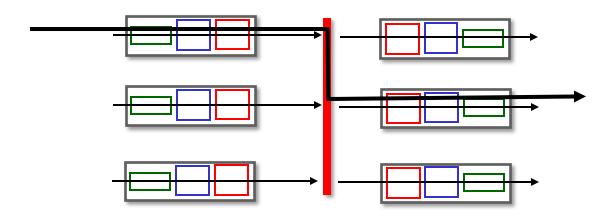
first generation routers:

- 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)



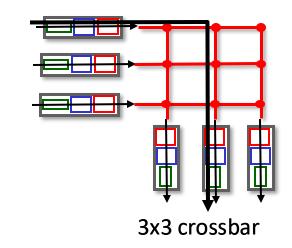
Switching via a bus

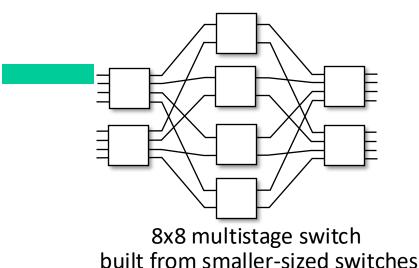
- datagram from input port memory to output port memory via a shared bus
- bus contention: switching speed limited by bus bandwidth
- 32 Gbps bus, Cisco 5600: sufficient speed for access routers (small local area and enterprise networks)



Switching via interconnection network

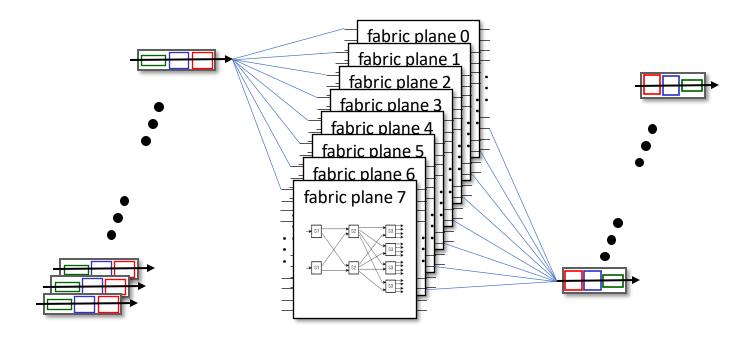
- Crossbar, Clos networks, other interconnection nets initially developed to connect processors in multiprocessor
- multistage switch: nxn switch from multiple stages of smaller switches
- exploiting parallelism:
 - fragment datagram into fixed length cells on entry
 - switch cells through the fabric,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





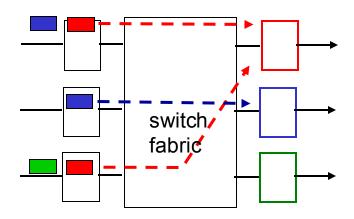
Switching via interconnection network

- 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

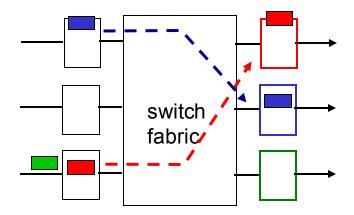


Input port queuing

- 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

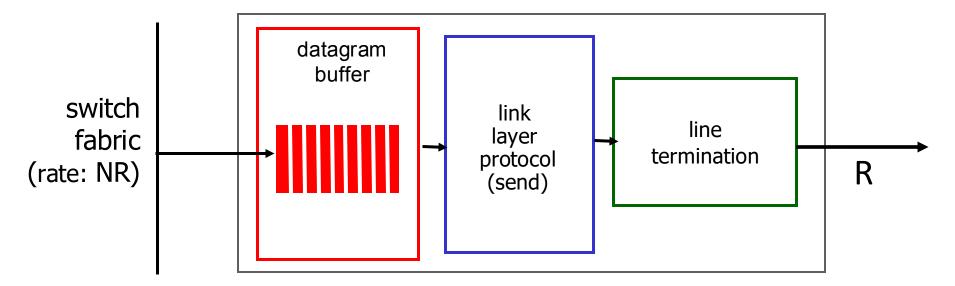


output port contention: at time t, only one red datagram can be transferred.

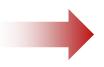


one packet time later: green packet experiences HOL blocking

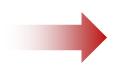
Output port queuing



- Buffering required when datagrams arrive from fabric faster than link transmission rate.
- Scheduling discipline chooses among queued datagrams for transmission.

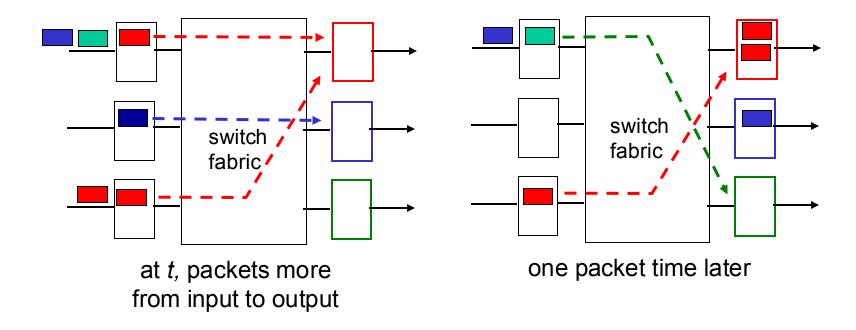


Datagrams can be lost due to congestion, lack of buffers



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!

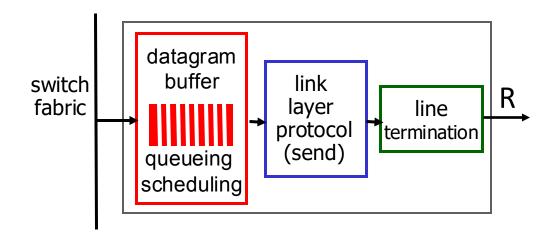
How much buffering?

- 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

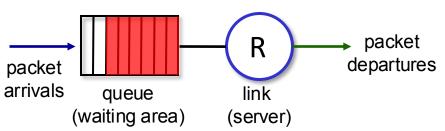
$$\frac{\mathsf{RTT} \cdot \mathsf{C}}{\sqrt{\mathsf{N}}}$$

- but too much buffering can increase delays (particularly in home routers)
 - long RTTs: poor performance for real-time 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

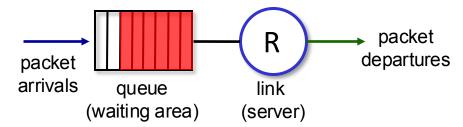
packet scheduling: deciding which
packet to send next on link

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

FCFS: packets transmitted in order of arrival to output port

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

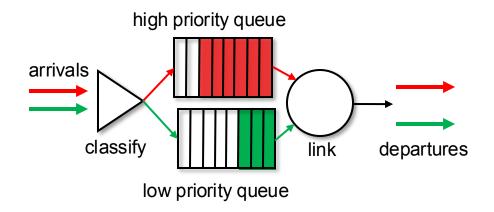
Abstraction: queue

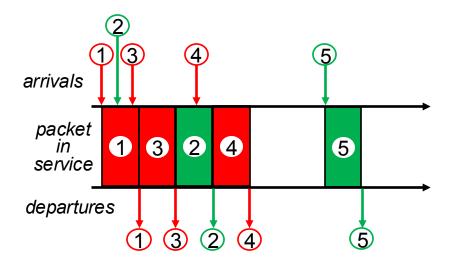


Scheduling policies: priority

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

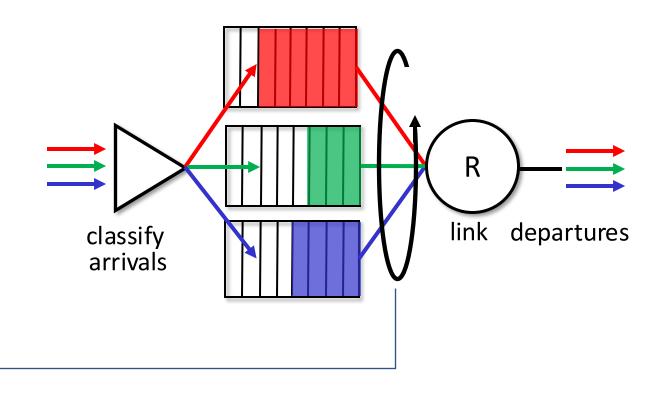




Scheduling policies: round robin

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

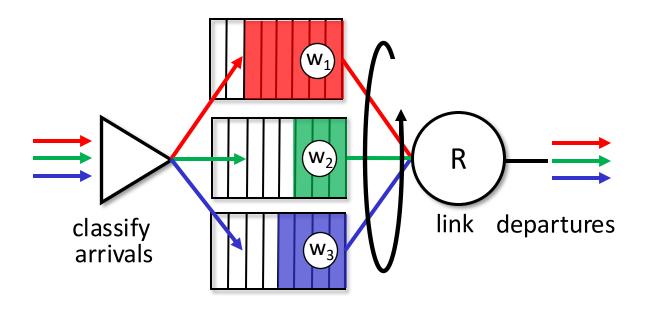


Scheduling policies: weighted fair queueing

Weighted Fair Queuing (WFQ):

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

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



Network layer: "data plane" roadmap

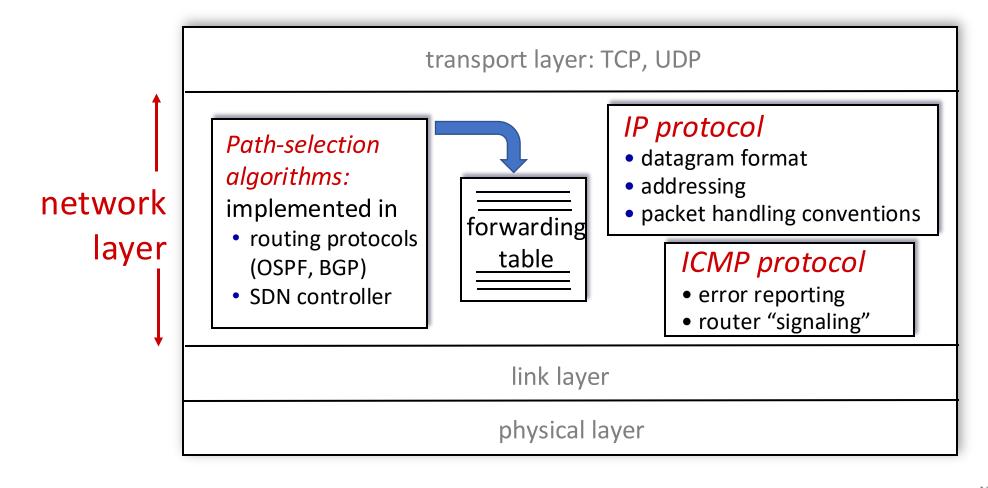
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- Generalized Forwarding, SDN
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Network Layer: Internet

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

TCP+IP

layer overhead for

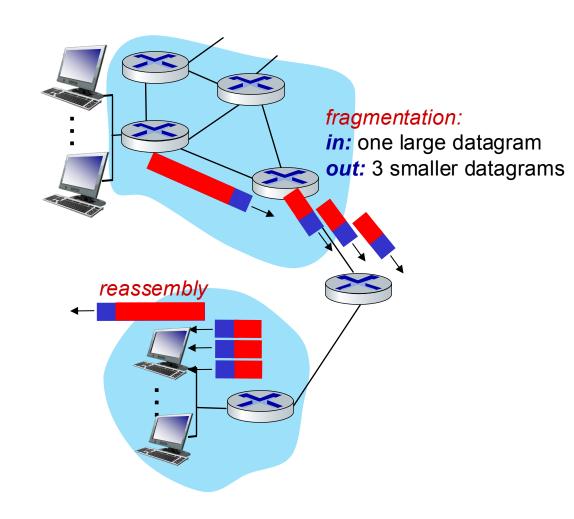
32 bits total datagram head. type of ver length length (bytes) service len fragment fragmentation/ 16-bit identifier | flgs offset reassembly time to upper header header checksum live layer checksum 32-bit source IP address source IP address destination IP address 32-bit destination IP address options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

Network Layer: 4-39

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



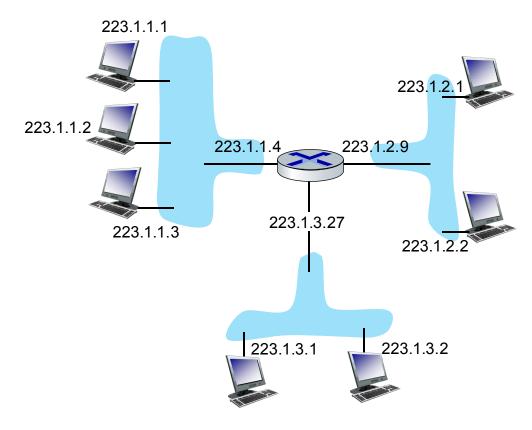
IP fragmentation/reassembly

example: fragflag length | ID offset =4000 | =x =0 =0 4000 byte datagram MTU = 1500 bytes one large datagram becomes several smaller datagrams Payload = 1500 - 20 = 1480 bytes 1480 bytes in length | ID fragflag offset data field =1500 =χ =0 =1 offset = length ID fragflag offset 1480/8 =1500 =χ =185 =1 length ID fragflag offset =1040 =x =370 =()

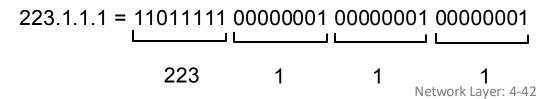
Payload = 4000 - 20 = 3980 bytes

IP addressing: introduction

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

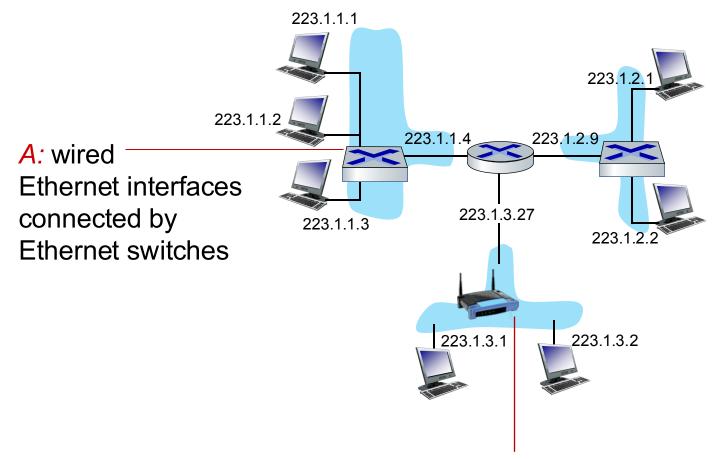


dotted-decimal IP address notation:



IP addressing: introduction

Q: how are interfaces actually connected?



A: wireless WiFi interfaces connected by WiFi base station

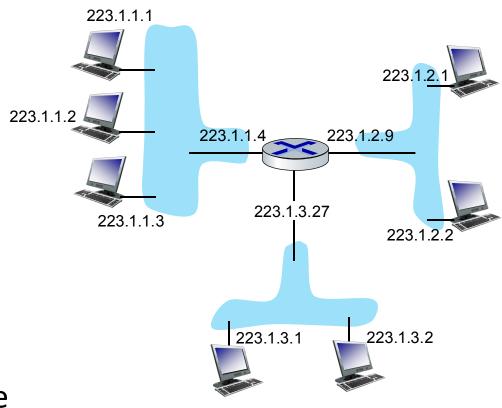
Subnets

■ What's a subnet?

- device interfaces with same subnet part of IP address
- 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



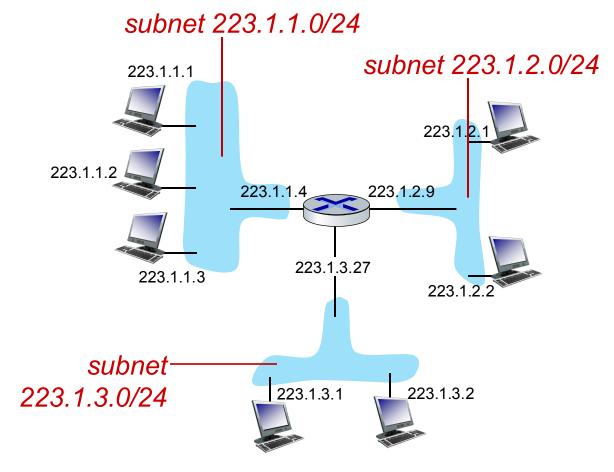
network consisting of 3 subnets

Subnets

Recipe:

 to determine the subnets, detach each interface from its host or router, creating islands of isolated networks

each isolated network is called a subnet



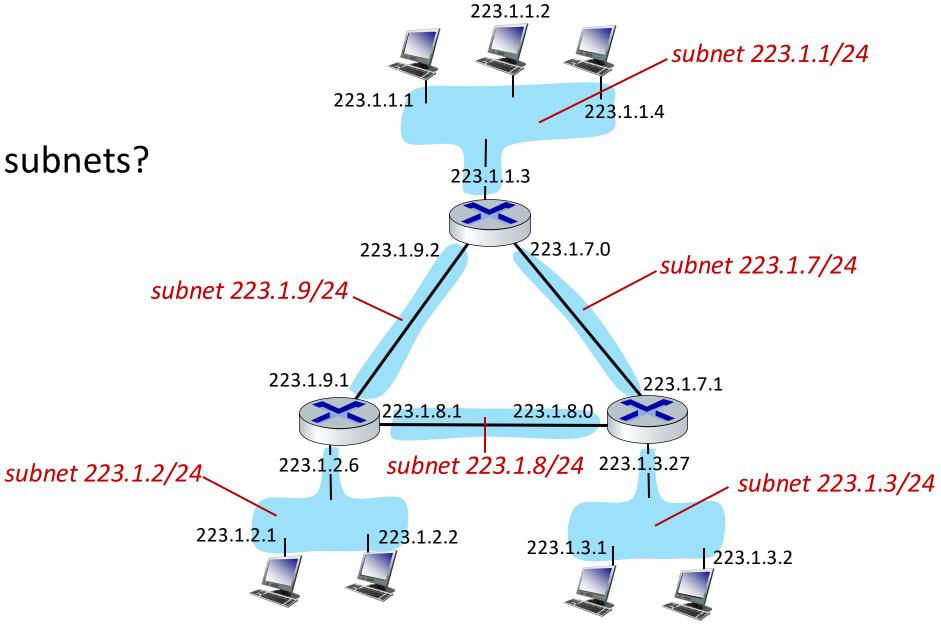
what are the /24 subnet addresses?

subnet mask: /24

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

Subnets

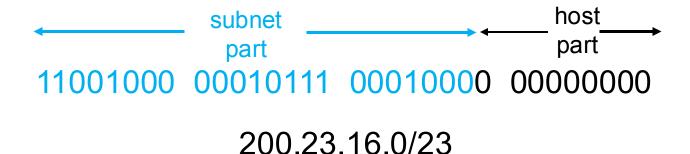
where are the subnets?



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?

Q: How does *host* get IP address?

- hard-coded by sysadmin in config file
 - e.g., Windows: control-panel->network->configuration->tcp/ip->properties
 - UNIX: /etc/rc.config)
- DHCP: Dynamic Host Configuration Protocol: dynamically get address from as server
 - "plug-and-play"

DHCP: Dynamic Host Configuration Protocol

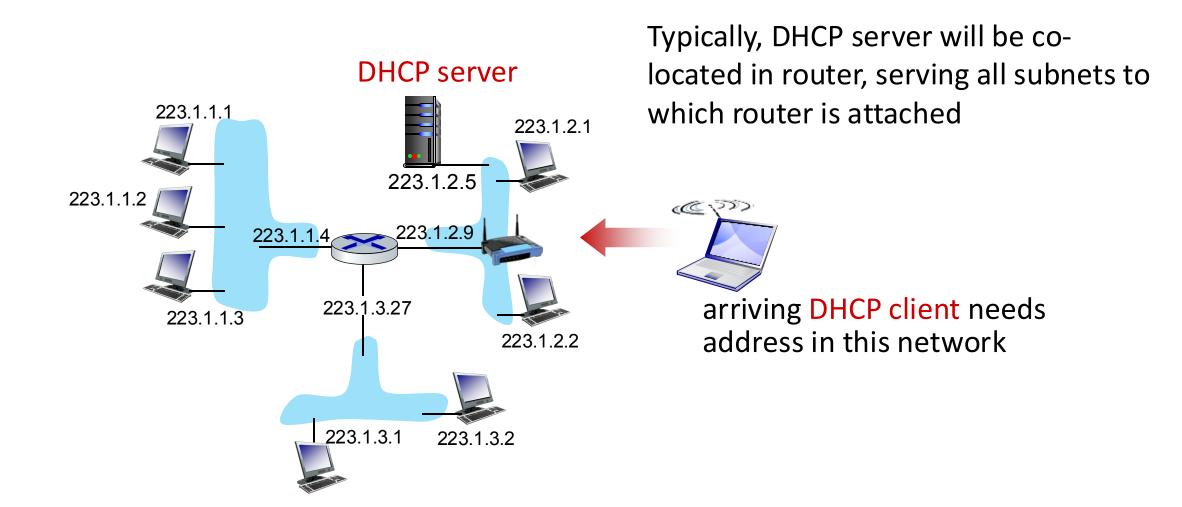
goal: host dynamically obtains IP address from network server when it "joins" network

- 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

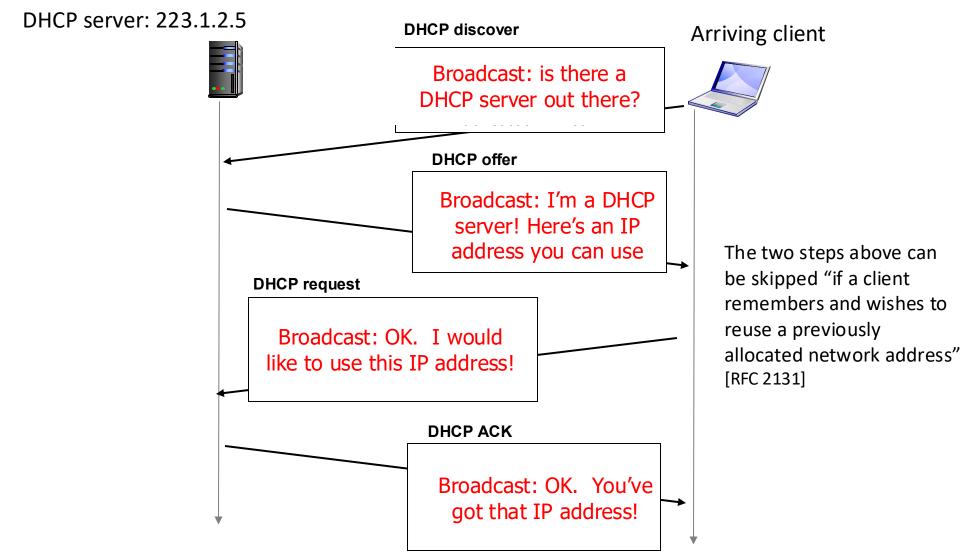
DHCP overview:

- 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)

IP addresses: how to get one?

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

A: gets allocated portion of its 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 00010000 00000000
        200.23.16.0/23

        Organization 1
        11001000 00010111 00010010 00000000
        200.23.16.0/23

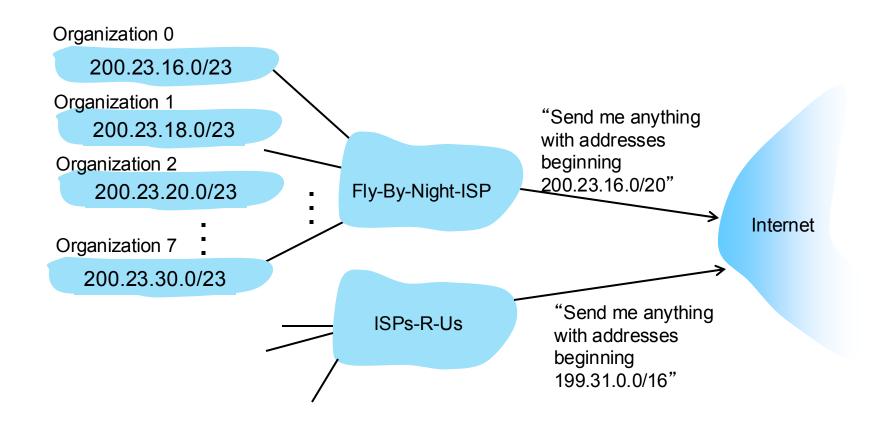
        Organization 2
        11001000 00010111 00010100 00000000
        200.23.20.0/23

        ...
        ...
        ...

        Organization 7
        11001000 00010111 00011110 00000000
        200.23.30.0/23
```

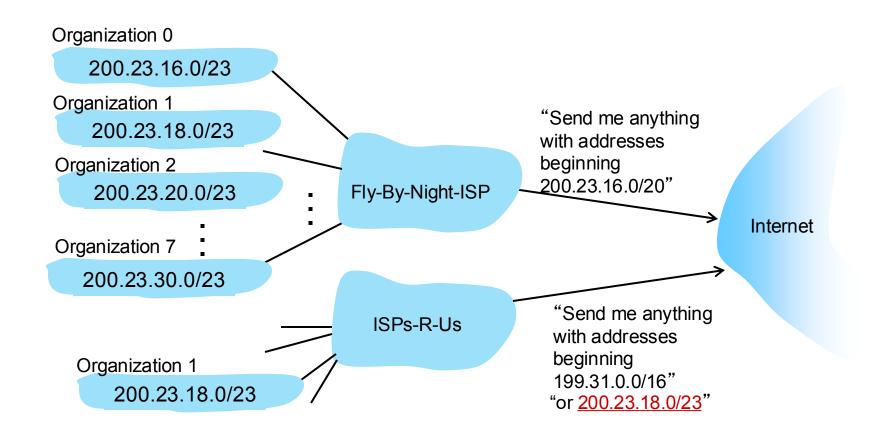
Hierarchical addressing: route aggregation

hierarchical addressing allows efficient advertisement of routing information:



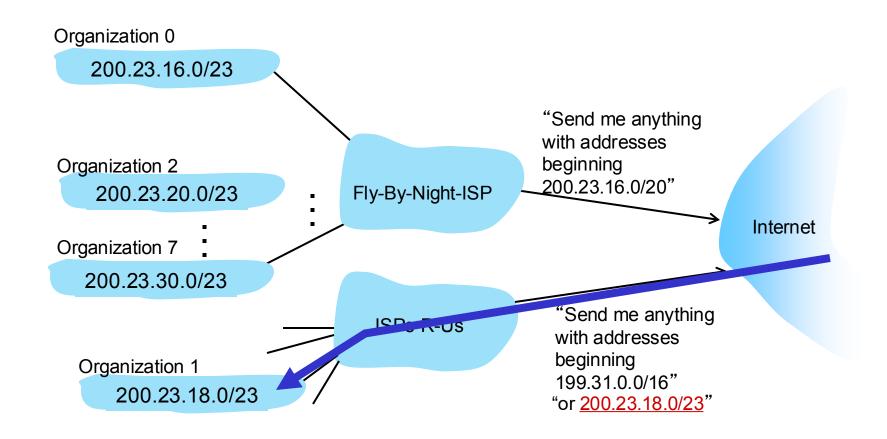
Hierarchical addressing: more specific routes

- 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)