

Chapter 4

Network Layer

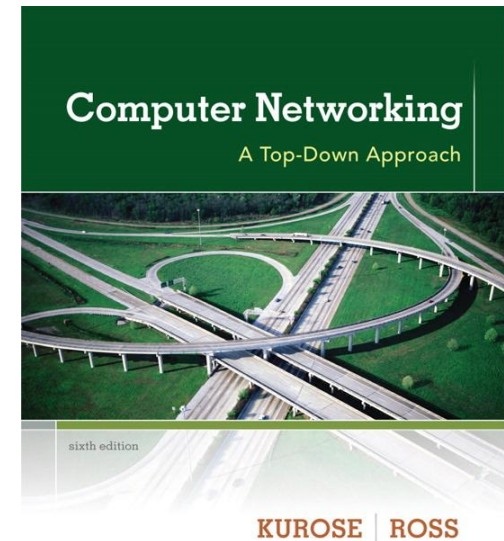
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**Computer
Networking: A Top
Down Approach**
6th edition
Jim Kurose, Keith Ross
Addison-Wesley
March 2012

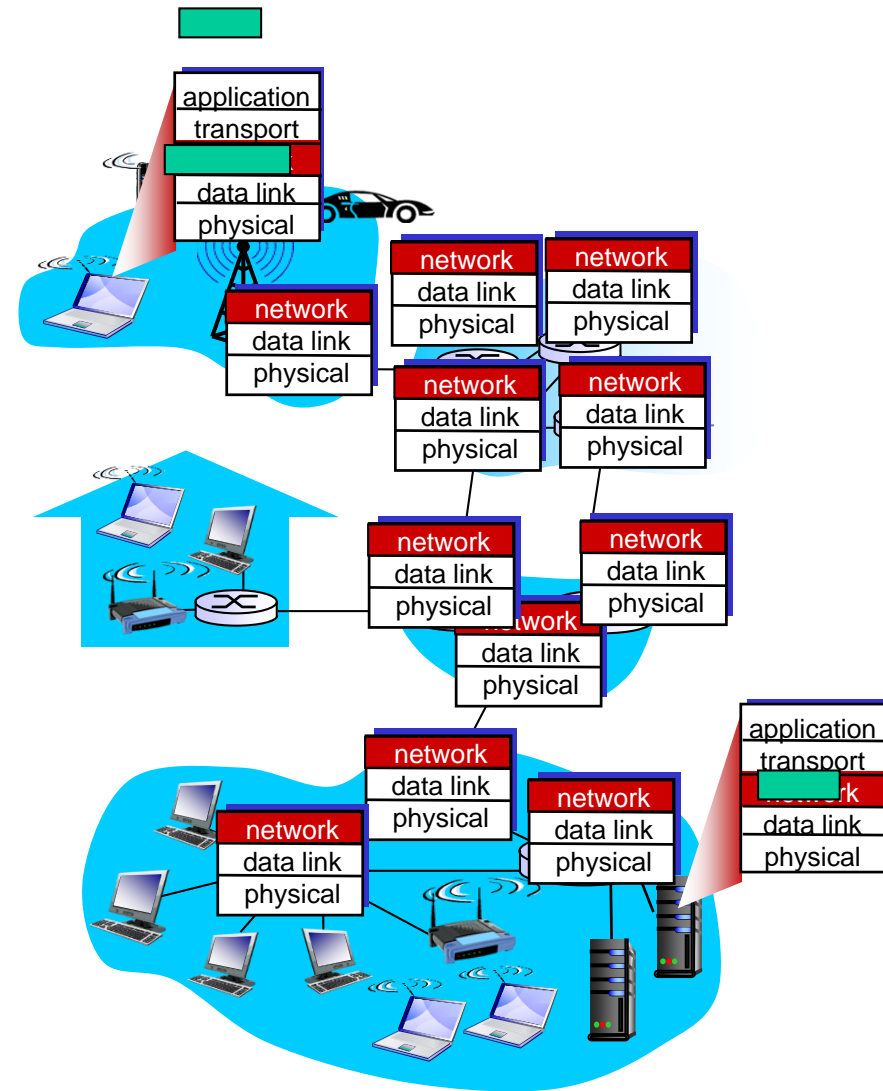
Chapter 4: network layer

chapter goals:

- ❖ understand principles behind network layer services:
 - network layer service models
 - forwarding versus routing
 - how a router works
 - routing (path selection)
 - broadcast, multicast
- ❖ instantiation, implementation in the Internet

Network layer

- ❖ transport segment from sending to receiving host
- ❖ on sending side encapsulates segments into datagrams
- ❖ on receiving side, delivers segments to transport layer
- ❖ network layer protocols in *every* host, router
- ❖ router examines header fields in all IP datagrams passing through it



Two key network-layer functions

❖ *forwarding*: move packets from router's input to appropriate router output

❖ *routing*: determine route taken by packets from source to dest.

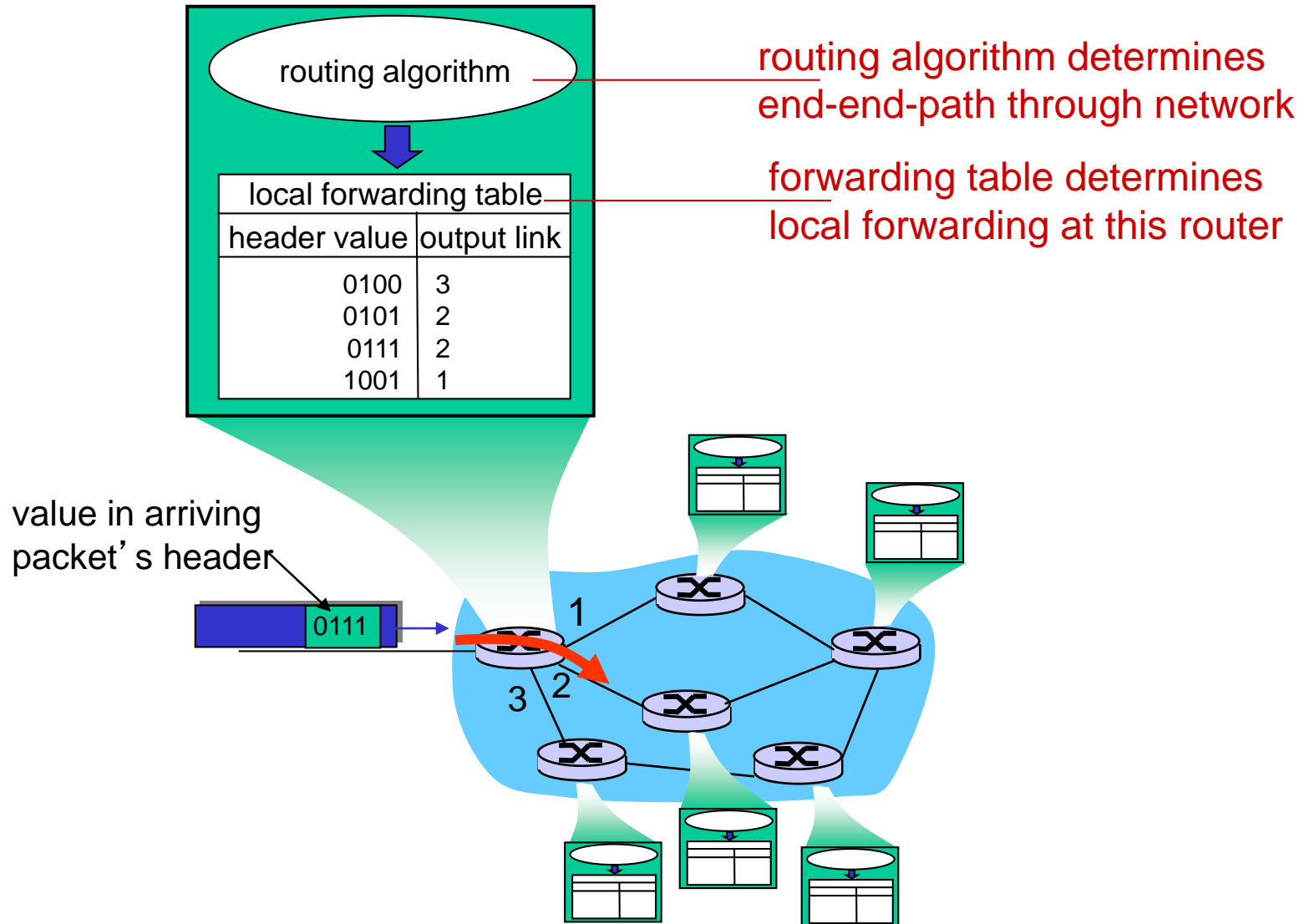
- *routing algorithms*

analogy:

❖ *routing*: process of planning trip from source to dest

❖ *forwarding*: process of getting through single interchange

Interplay between routing and forwarding



Connection setup

- ❖ 3rd important function in *some* network architectures:
 - ATM, frame relay, X.25
- ❖ before datagrams flow, two end hosts *and* intervening routers establish virtual connection
 - routers get involved
- ❖ network vs transport layer connection service:
 - *network*: between two hosts (may also involve intervening routers in case of VCs)
 - *transport*: between two processes

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 models:

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

Connection, connection-less service

- ❖ *datagram* network provides network-layer *connectionless* service
- ❖ *virtual-circuit* network provides network-layer *connection* service
- ❖ analogous to TCP/UDP connection-oriented / connectionless transport-layer services, but:
 - *service*: host-to-host
 - *no choice*: network provides one or the other
 - *implementation*: in network core

Virtual circuits

“source-to-dest path behaves much like telephone circuit”

- performance-wise
- network actions along source-to-dest path

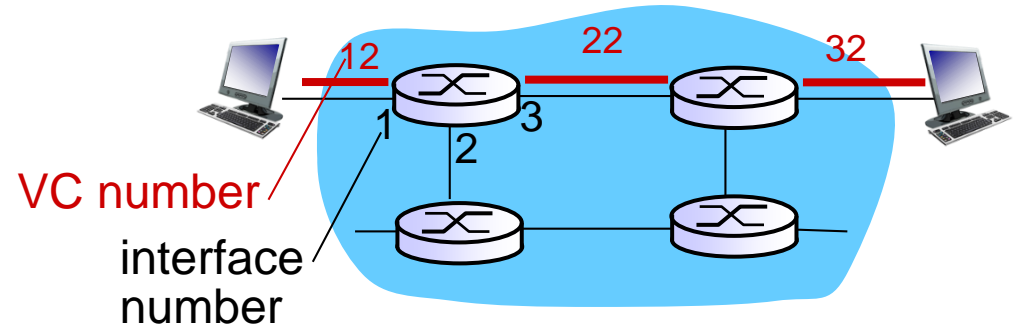
- ❖ call setup, teardown for each call *before* data can flow
- ❖ each packet carries VC identifier (not destination host address)
- ❖ every router on source-dest path maintains “state” for each passing connection
- ❖ link, router resources (bandwidth, buffers) may be *allocated* to VC (dedicated resources = predictable service)

VC implementation

a VC consists of:

1. *path* from source to destination
 2. *VC numbers*, one number for each link along path
 3. *entries in forwarding tables* in routers along path
- ❖ packet belonging to VC carries VC number (rather than dest address)
 - ❖ VC number can be changed on each link.
 - new VC number comes from forwarding table

VC forwarding table



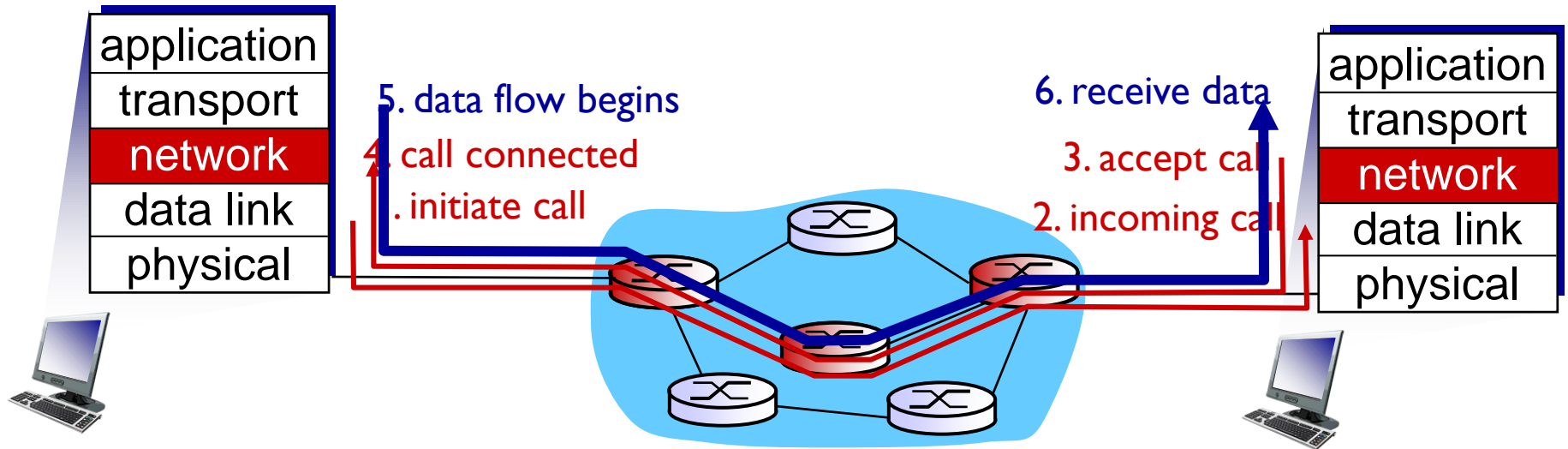
forwarding table in northwest router:

Incoming interface	Incoming VC #	Outgoing interface	Outgoing VC #
1	12	3	22
2	63	1	18
3	7	2	17
1	97	3	87
...

VC routers maintain connection state information!

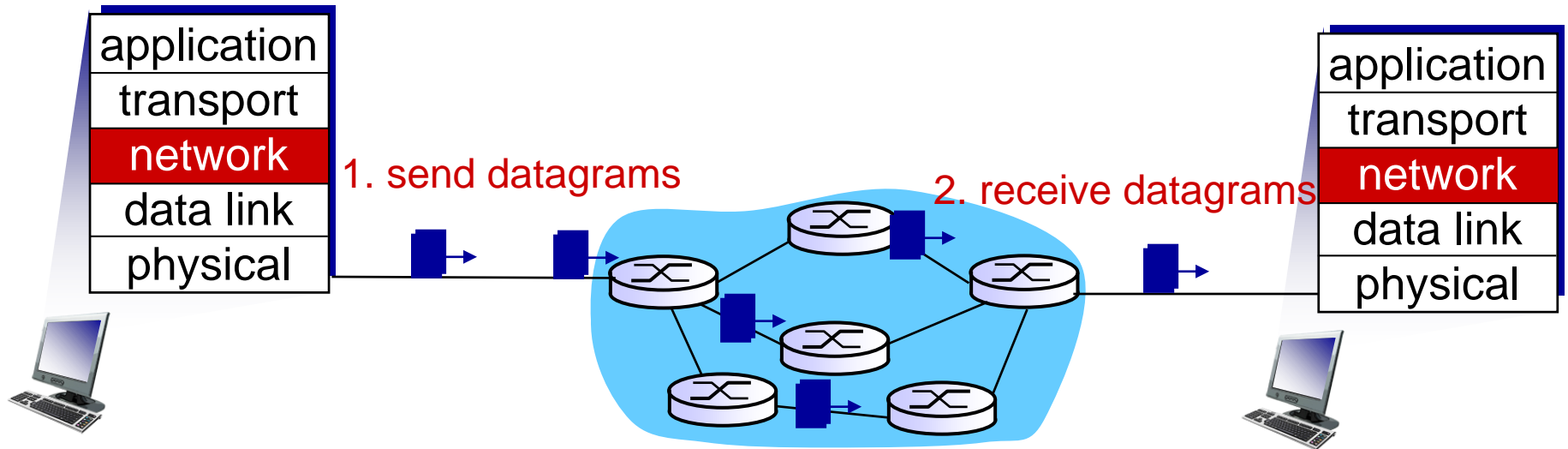
Virtual circuits: signaling protocols

- ❖ used to setup, maintain teardown VC
- ❖ used in ATM, frame-relay, X.25
- ❖ not used in today's Internet

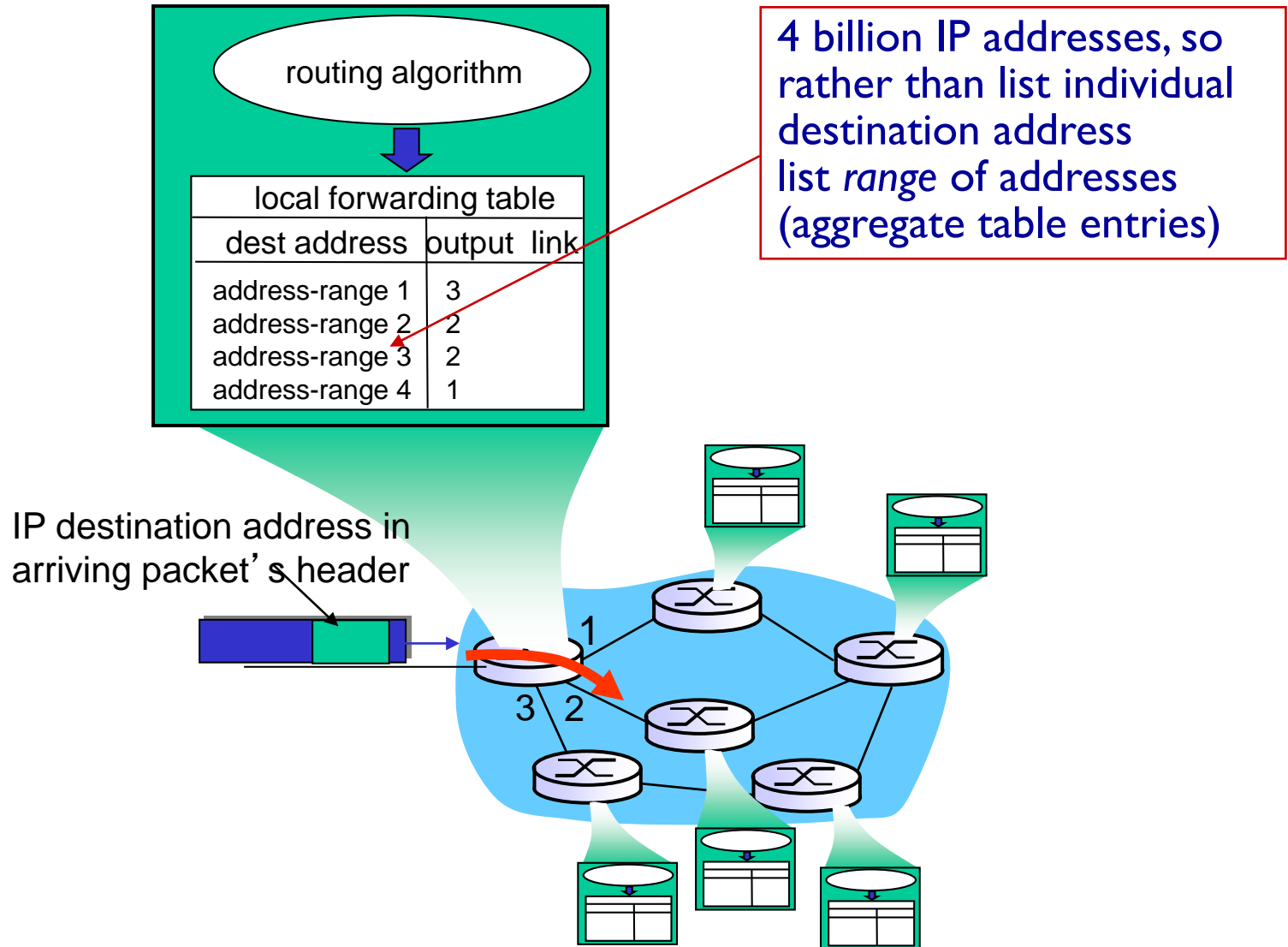


Datagram networks

- ❖ no call setup at network layer
- ❖ routers: no state about end-to-end connections
 - no network-level concept of “connection”
- ❖ packets forwarded using destination host address



Datagram forwarding table



Datagram forwarding table

Destination Address Range	Link Interface
11001000 00010111 00010000 00000000 through 11001000 00010111 00010111 11111111	0
11001000 00010111 00011000 00000000 through 11001000 00010111 00011000 11111111	1
11001000 00010111 00011001 00000000 through 11001000 00010111 00011111 11111111	2
otherwise	3

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

Longest prefix matching

longest prefix matching

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

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

examples:

DA: 11001000 00010111 00010110 10100001

which interface?

DA: 11001000 00010111 00011000 10101010

which interface?

Datagram or VC network: why?

Internet (datagram)

- ❖ data exchange among computers
 - “elastic” service, no strict timing req.
- ❖ many link types
 - different characteristics
 - uniform service difficult
- ❖ “smart” end systems (computers)
 - can adapt, perform control, error recovery
 - **simple inside network, complexity at “edge”**

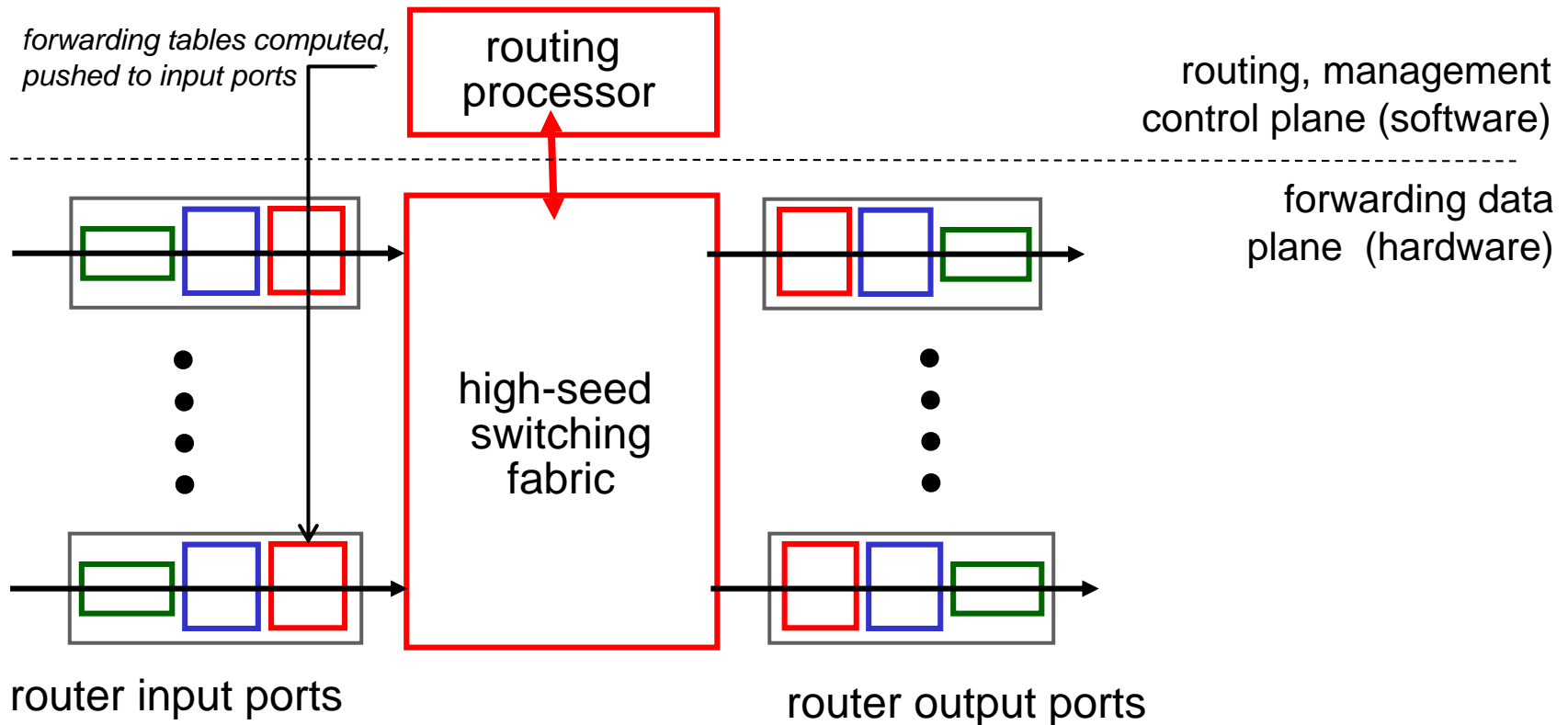
ATM (VC)

- ❖ evolved from telephony
- ❖ human conversation:
 - strict timing, reliability requirements
 - need for guaranteed service
- ❖ “dumb” end systems
 - telephones
 - **complexity inside network**

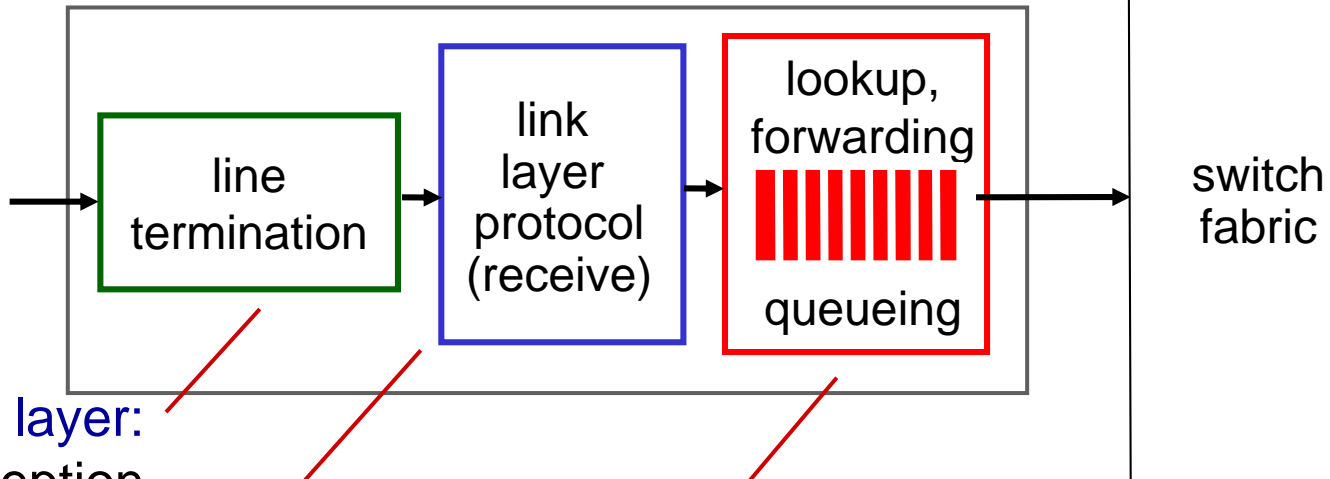
Router architecture overview

two key router functions:

- ❖ run routing algorithms/protocol (RIP, OSPF, BGP)
- ❖ *forwarding* datagrams from incoming to outgoing link



Input port functions



physical layer:
bit-level reception

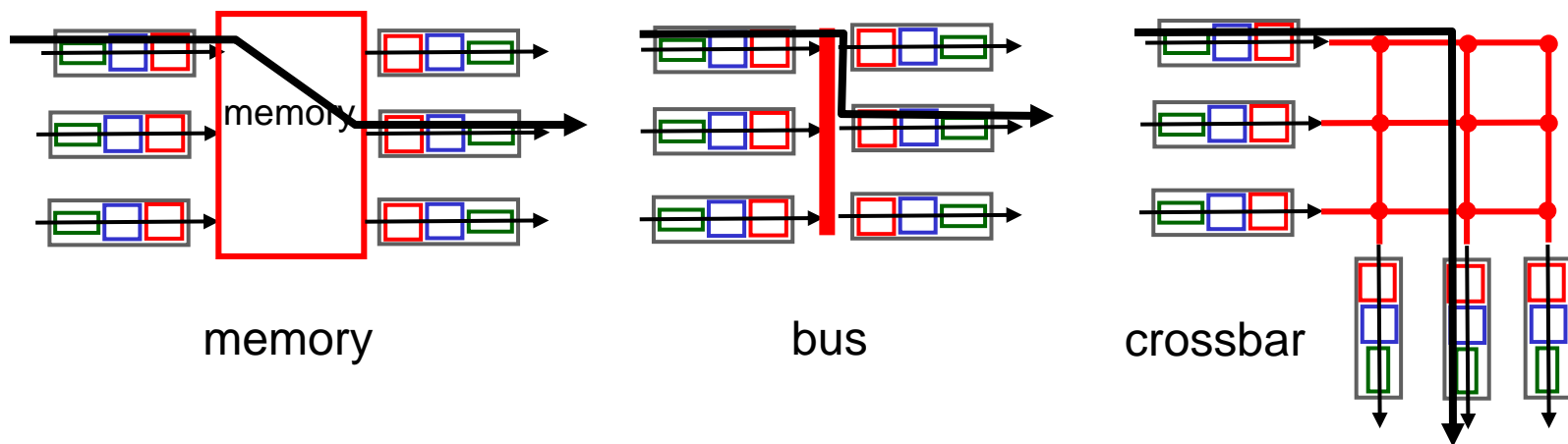
data link layer:
e.g., Ethernet
see chapter 5

decentralized switching:

- ❖ given datagram dest., lookup output port using forwarding table in input port memory (“*match plus action*”)
- ❖ goal: complete input port processing at ‘line speed’
- ❖ queuing: if datagrams arrive faster than forwarding rate into switch fabric

Switching fabrics

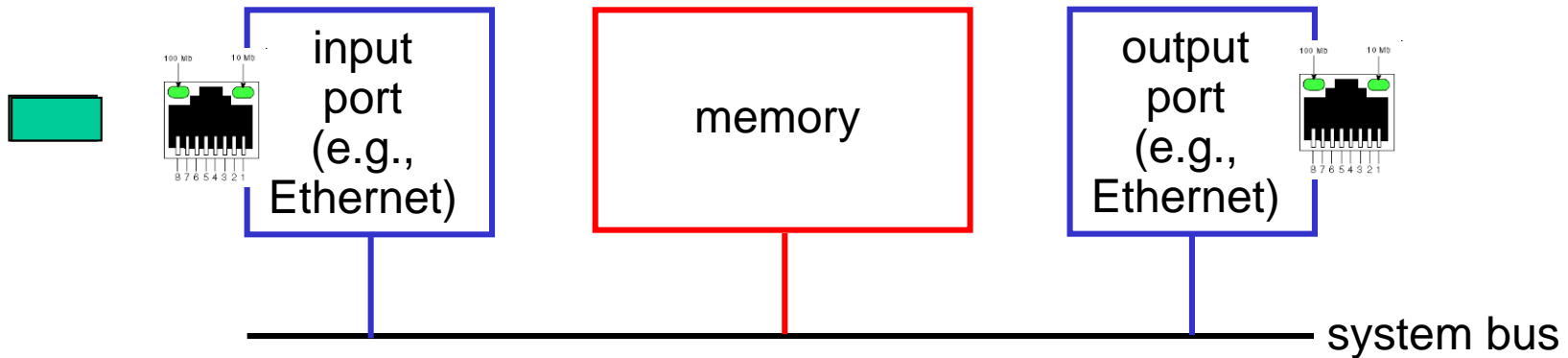
- ❖ transfer packet from input buffer to appropriate output buffer
- ❖ switching rate: rate at which packets can be transfer from inputs to outputs
 - often measured as multiple of input/output line rate
 - N inputs: switching rate N times line rate desirable
- ❖ three types of switching fabrics



Switching via memory

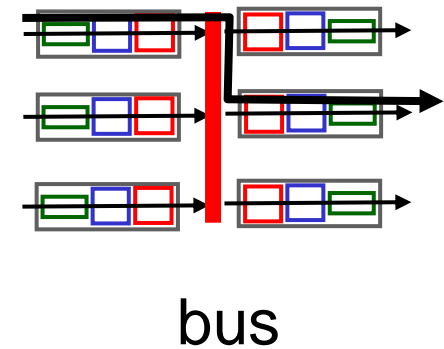
first generation routers:

- ❖ 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)



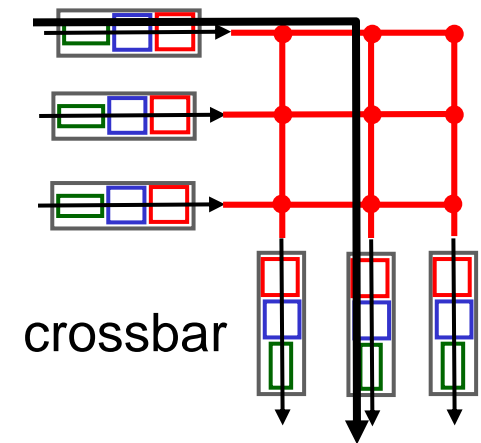
Switching via a bus

- ❖ 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 and enterprise routers

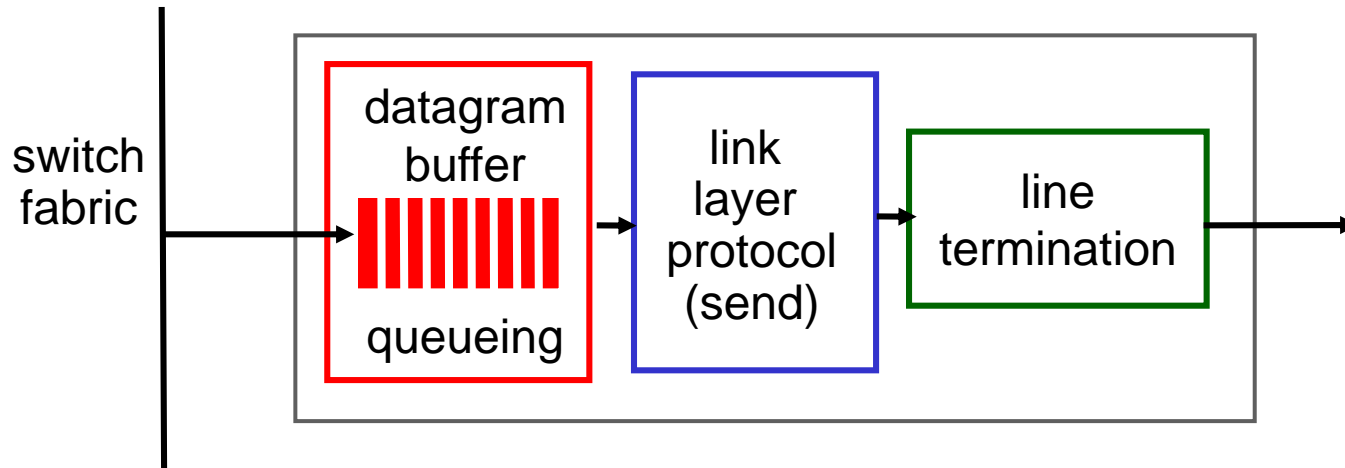


Switching via interconnection network

- ❖ overcome bus bandwidth limitations
- ❖ banyan networks, crossbar, other interconnection nets initially developed to connect processors in multiprocessor
- ❖ advanced design: fragmenting datagram into fixed length cells, switch cells through the fabric.
- ❖ Cisco I2000: switches 60 Gbps through the interconnection network

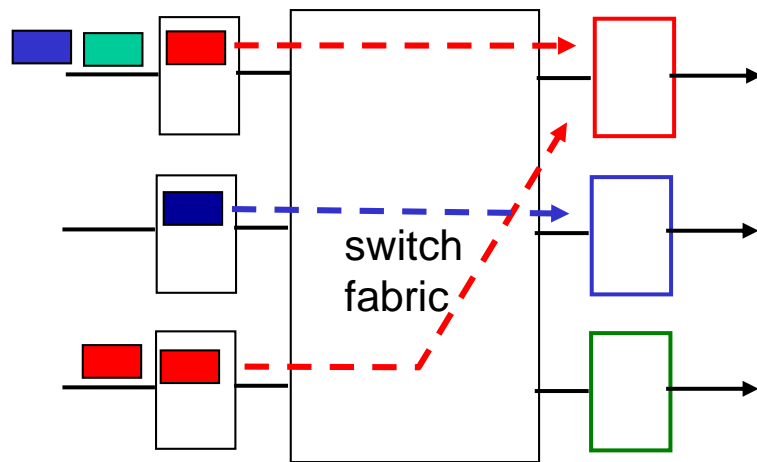


Output ports

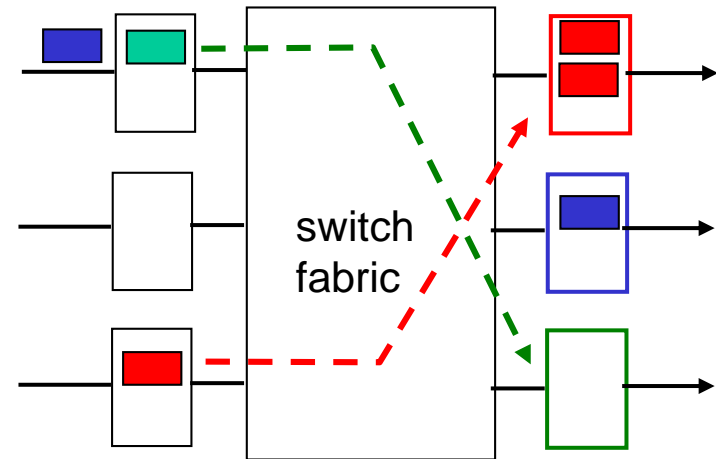


- ❖ *buffering* required when datagrams arrive from fabric faster than the transmission rate
- ❖ *scheduling discipline* chooses among queued datagrams for transmission

Output port queueing



at t , packets more
from input to output



one packet time later

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

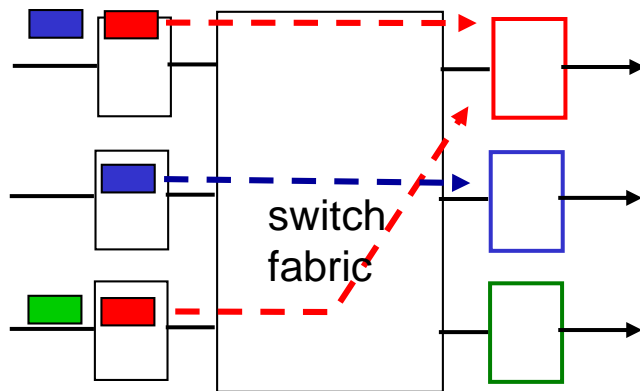
How much buffering?

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

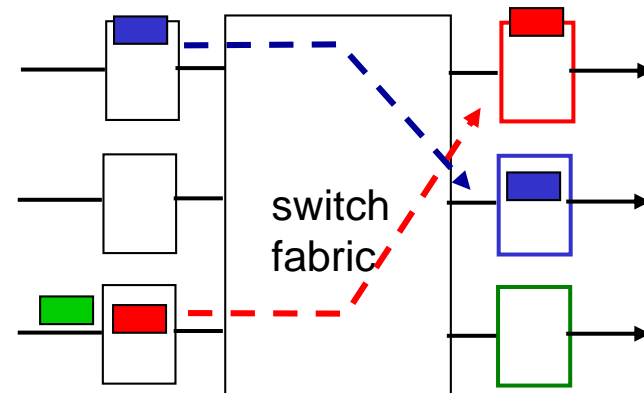
$$\frac{RTT \cdot C}{\sqrt{N}}$$

Input port queuing

- ❖ fabric slower than input ports combined -> queuing 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



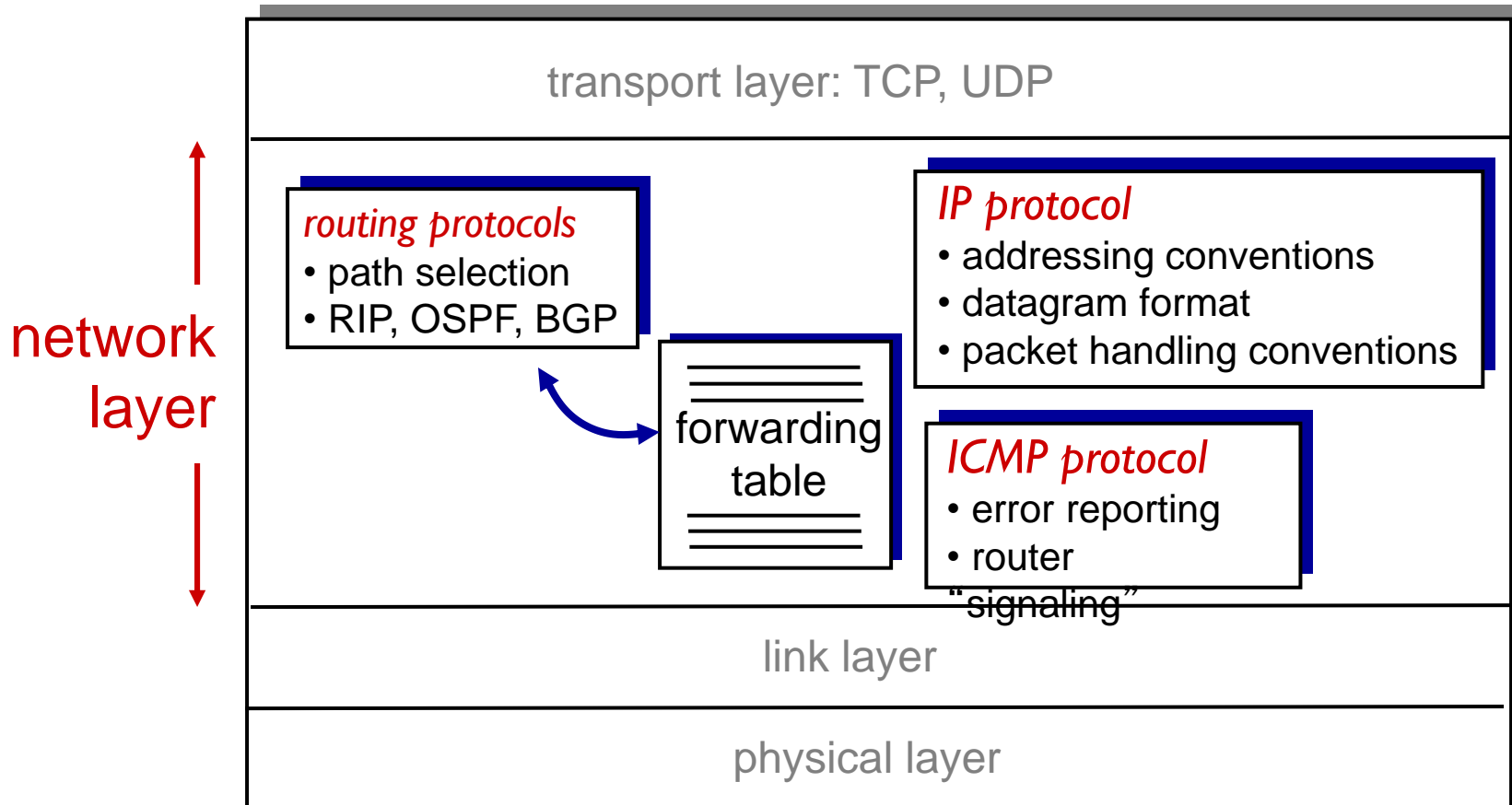
output port contention:
only one red datagram can be
transferred.
lower red packet is blocked



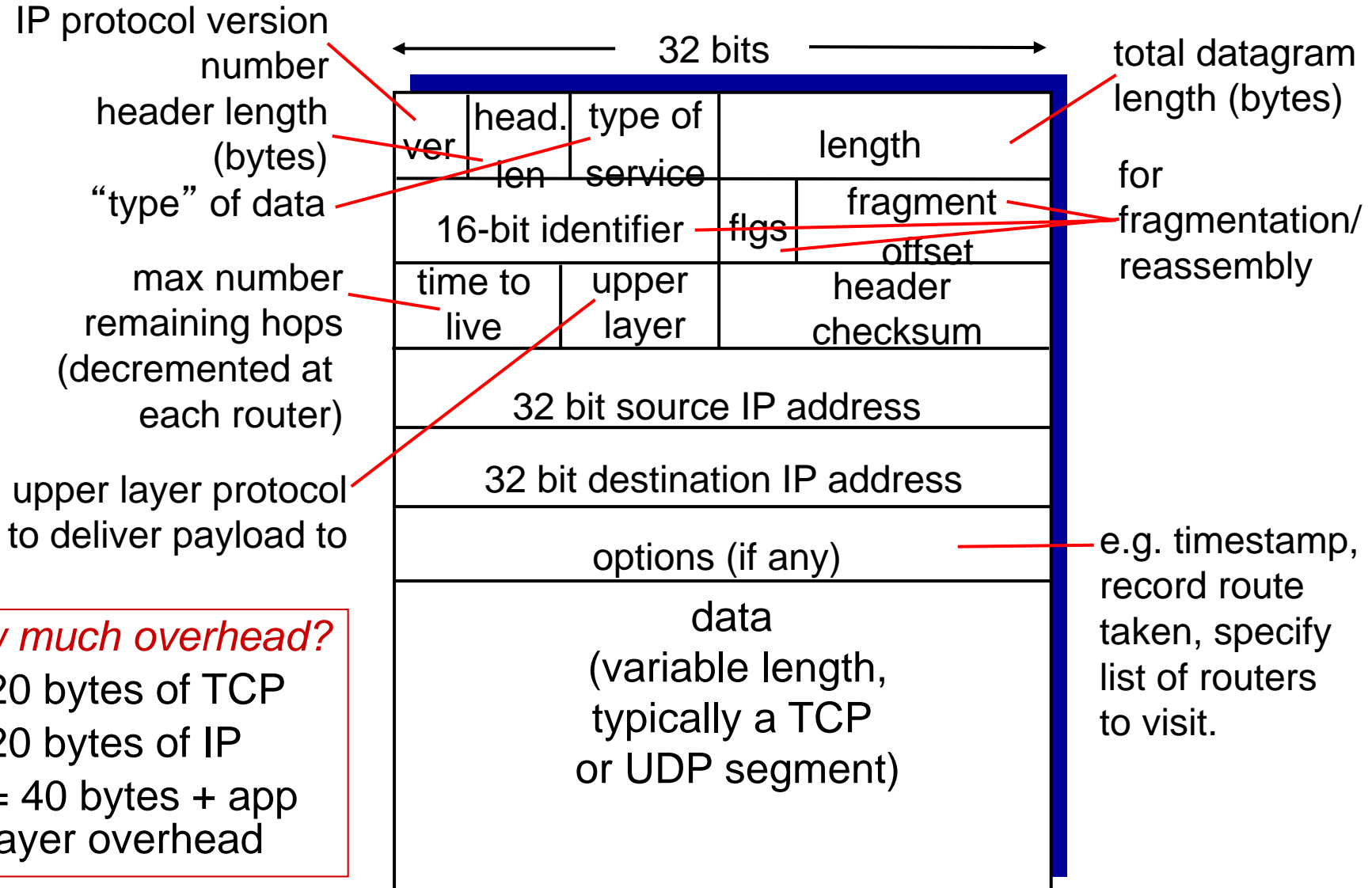
one packet time later:
green packet
experiences HOL
blocking

The Internet network layer

host, router network layer functions:



IP datagram format

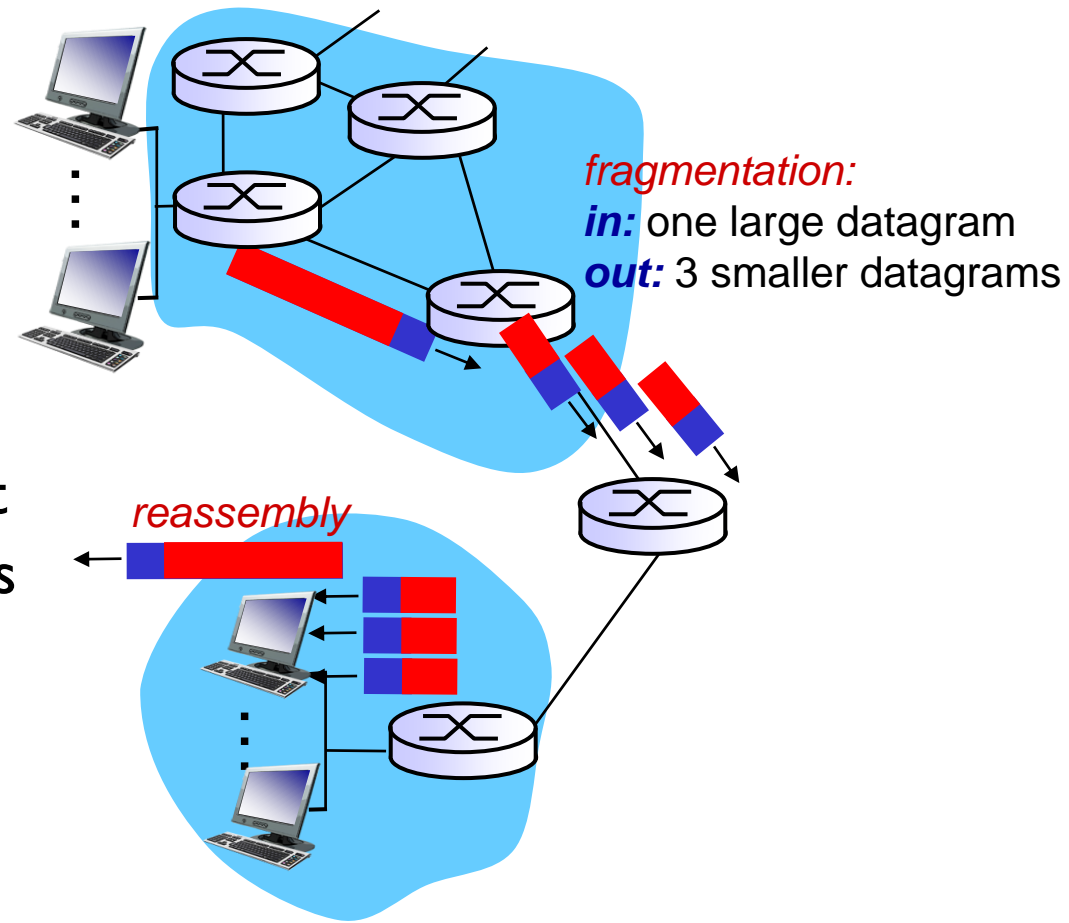


how much overhead?

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

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 final destination
 - IP header bits used to identify, order related fragments



IP fragmentation, reassembly

example:

- ❖ 4000 byte datagram
- ❖ MTU = 1500 bytes

	length	ID	fragflag	offset	
	=4000	=x	=0	=0	

one large datagram becomes several smaller datagrams

1480 bytes in data field

offset =
1480/8

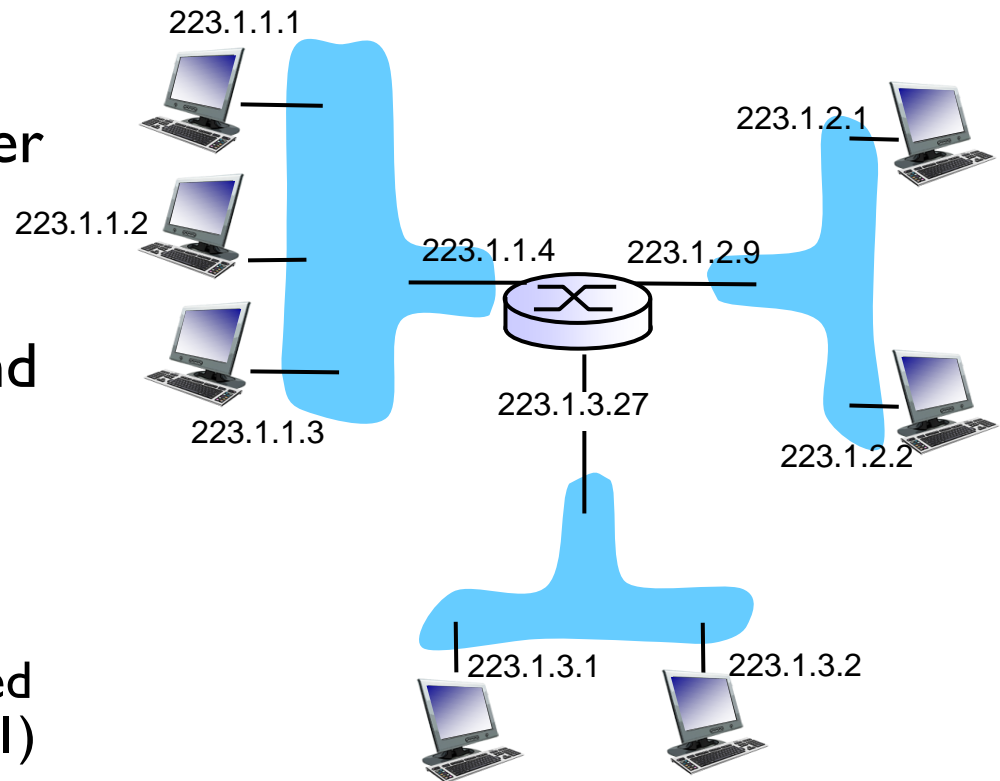
	length	ID	fragflag	offset	
	=1500	=x	=1	=0	

	length	ID	fragflag	offset	
	=1500	=x	=1	=185	

	length	ID	fragflag	offset	
	=1040	=x	=0	=370	

IP addressing: introduction

- ❖ **IP address:** 32-bit identifier for host, 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)
- ❖ **IP addresses associated with each interface**



$$223.1.1.1 = \underbrace{11011111}_{223} \underbrace{0000001}_{1} \underbrace{0000001}_{1} \underbrace{0000001}_{1}$$

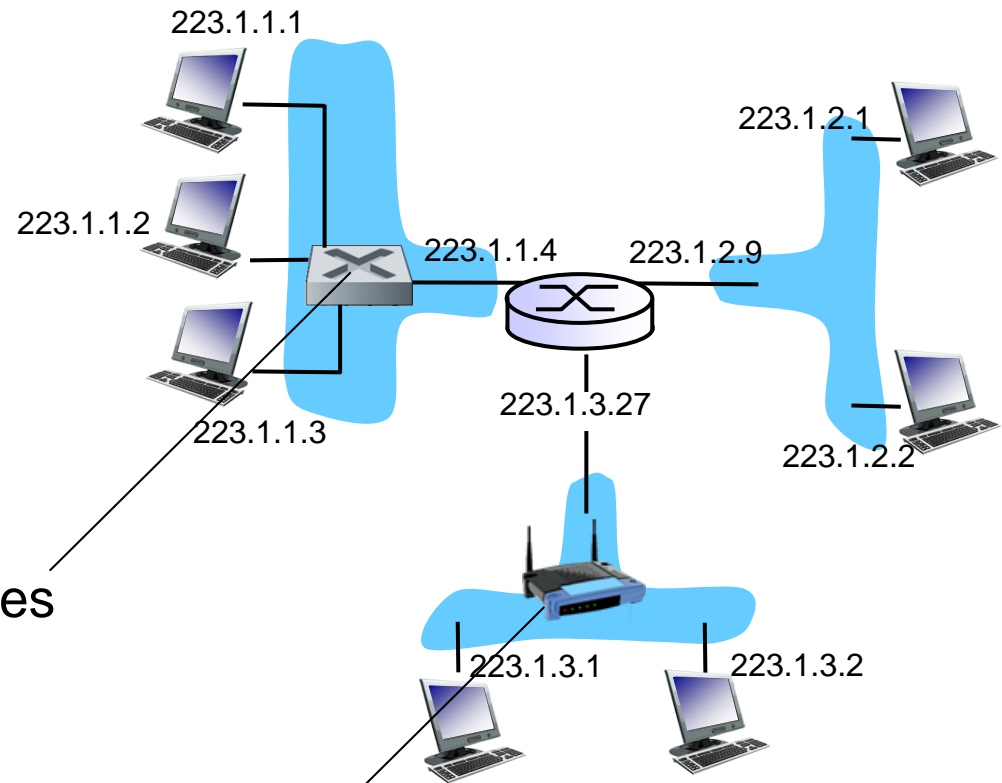
IP addressing: introduction

Q: how are interfaces actually connected?

A: we'll learn about that in chapter 5, 6.

A: wired Ethernet interfaces connected by Ethernet switches

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



A: wireless WiFi interfaces connected by WiFi base station

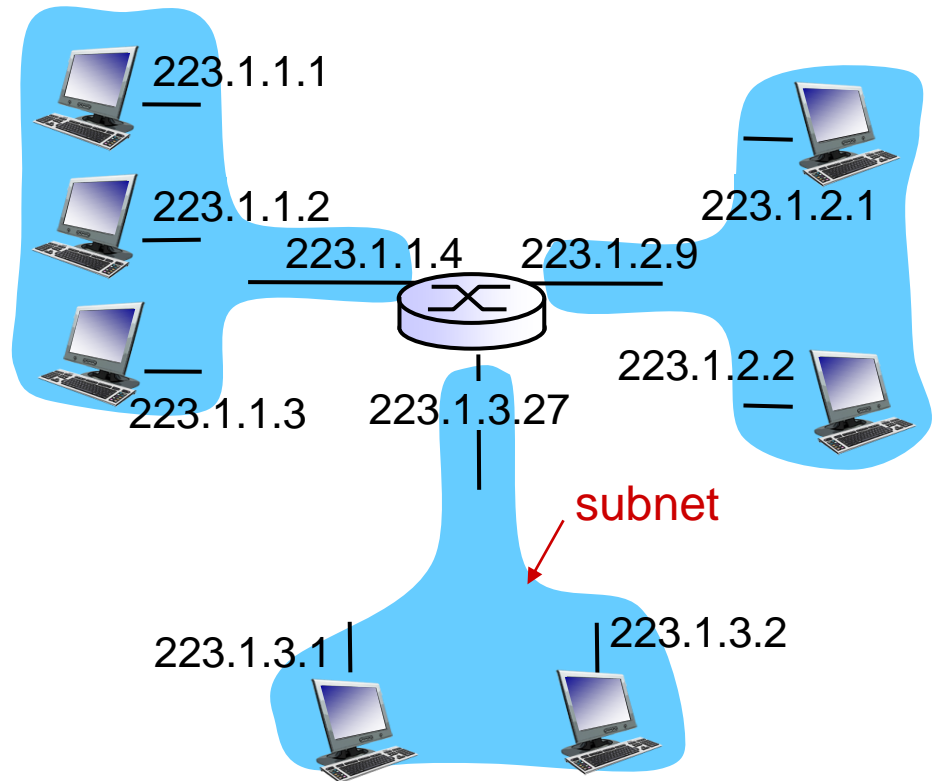
Subnets

❖ IP address:

- subnet part - high order bits
- host part - low order bits

❖ *what 's a subnet ?*

- device interfaces with same subnet part of IP address
- can physically reach each other *without intervening router*

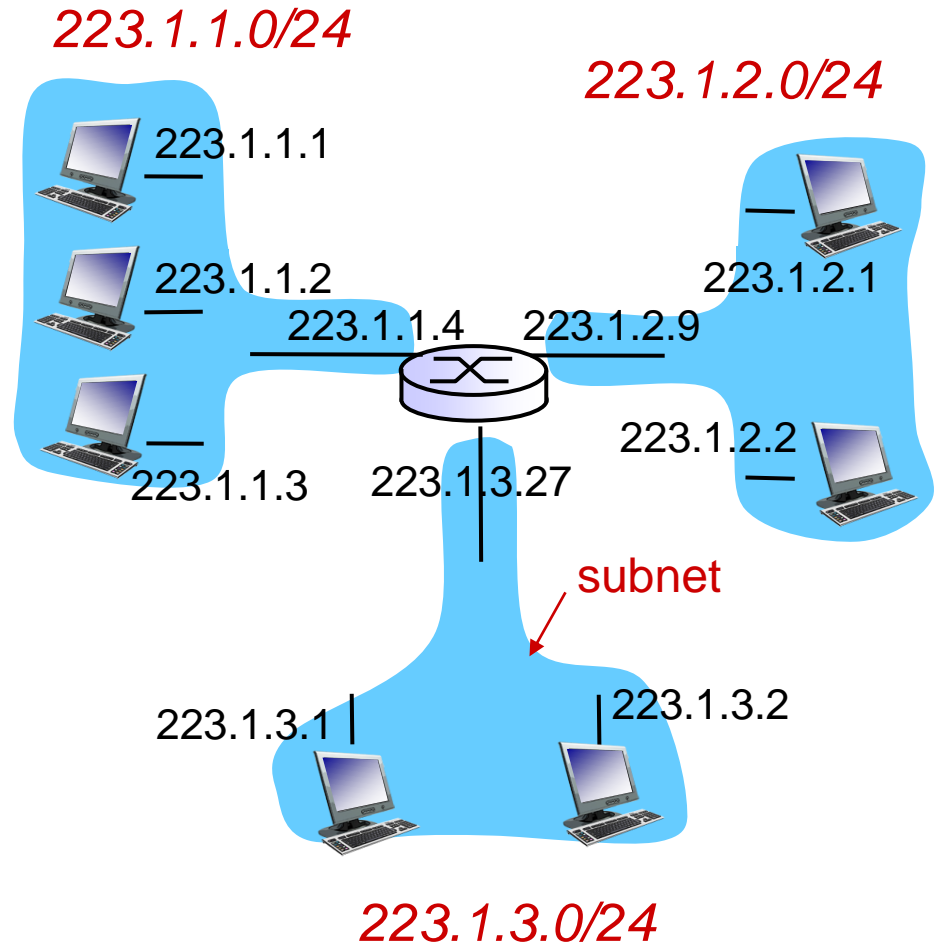


network consisting of 3 subnets

Subnets

recipe

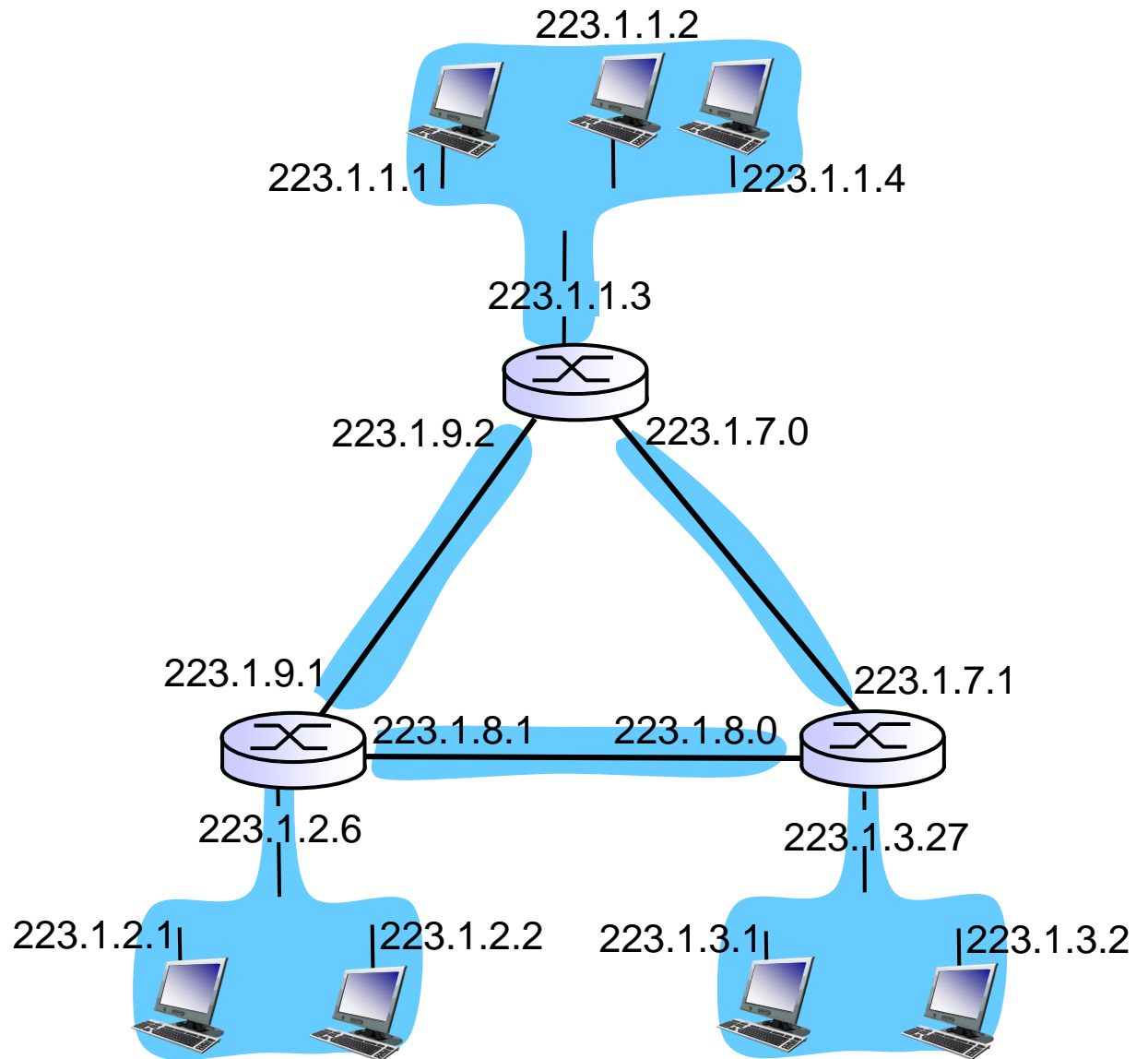
- ❖ to determine the subnets, detach each interface from its host or router, creating islands of isolated networks
- ❖ each isolated network is called a *subnet*



subnet mask: /24

Subnets

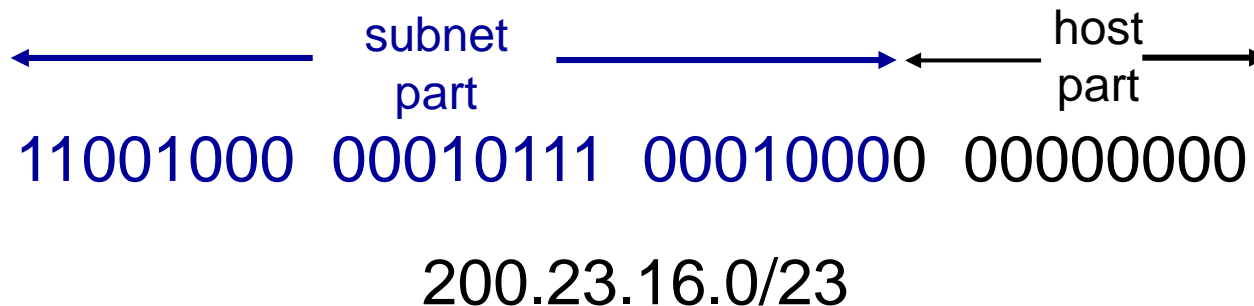
how many?



IP addressing: CIDR

CIDR: Classless InterDomain Routing

- 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 a *host* get IP address?

- ❖ hard-coded by system admin in a file
 - 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

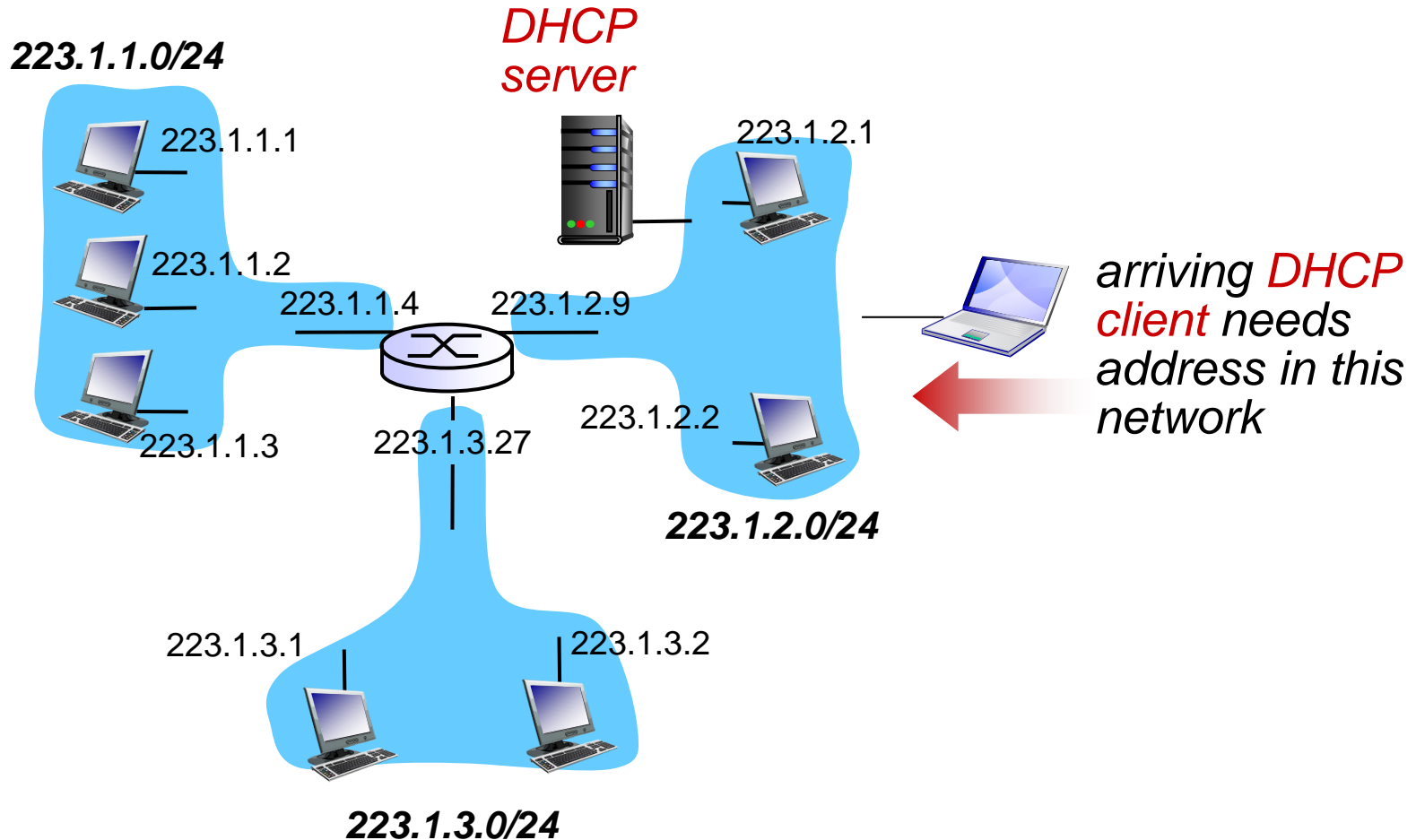
goal: allow host to *dynamically* obtain its IP address from network server when it joins network

- can renew its lease on address in use
- allows reuse of addresses (only hold address while connected/“on”)
- support for mobile users who want to join network (more shortly)

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 server: 223.1.2.5



DHCP discover

src : 0.0.0.0, 68
dest.: 255.255.255.255,67
yiaddr: 0.0.0.0
transaction ID: 654

arriving
client



DHCP offer

src: 223.1.2.5, 67
dest: 255.255.255.255, 68
yiaddr: 223.1.2.4
transaction ID: 654
lifetime: 3600 secs

DHCP request

src: 0.0.0.0, 68
dest.: 255.255.255.255, 67
yiaddr: 223.1.2.4
transaction ID: 655
lifetime: 3600 secs

DHCP ACK

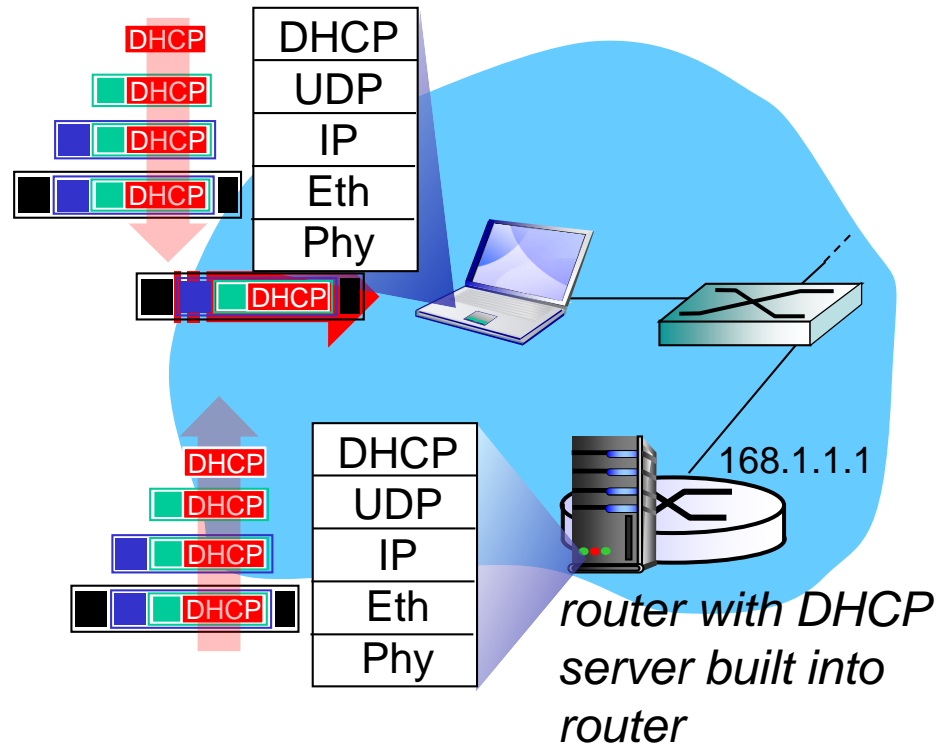
src: 223.1.2.5, 67
dest: 255.255.255.255, 68
yiaddr: 223.1.2.4
transaction ID: 655
lifetime: 3600 secs

DHCP: more than IP addresses

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

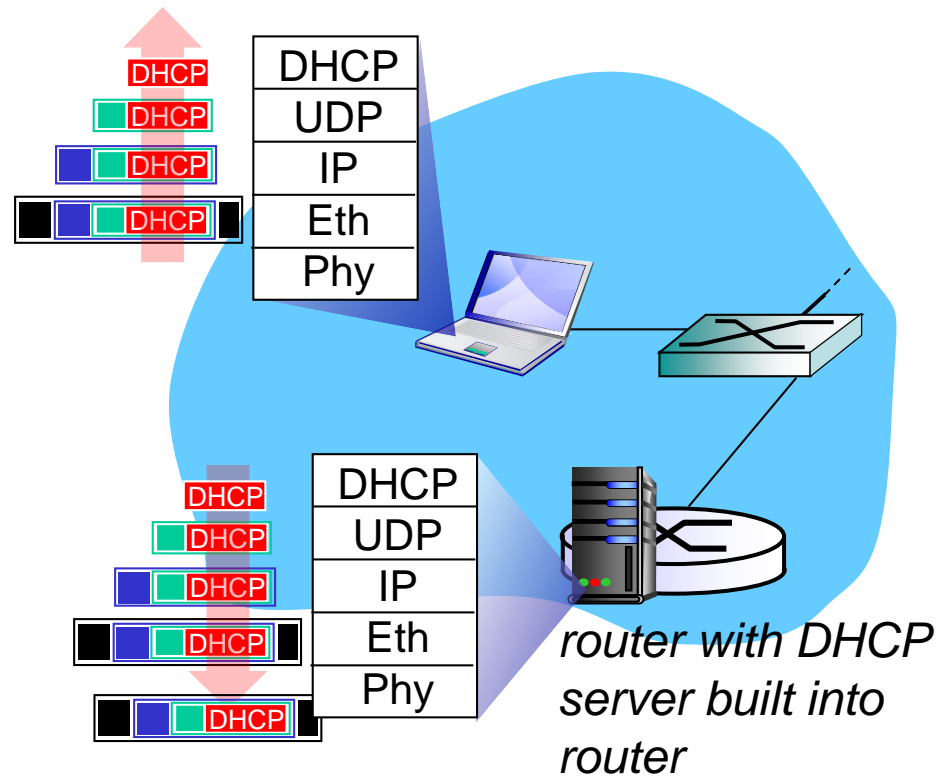
- address of first-hop router for client
- name and IP address of DNS sever
- network mask (indicating network versus host portion of address)

DHCP: example



- ❖ connecting laptop needs its IP address, addr of first-hop router, addr of DNS server: use DHCP
- ❖ DHCP request encapsulated in UDP, encapsulated in IP, encapsulated in 802.1 Ethernet
- ❖ Ethernet frame broadcast (dest: FFFFFFFF) on LAN, received at router running DHCP server
- ❖ Ethernet demuxed to IP demuxed, UDP demuxed to DHCP

DHCP: example



- ❖ DHCP server formulates DHCP ACK containing client's IP address, IP address of first-hop router for client, name & IP address of DNS server
- ❖ encapsulation of DHCP server, frame 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

DHCP: Wireshark output (home LAN)

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

Hardware type: Ethernet

Hardware address length: 6

Hops: 0

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

.....

request

Message type: **Boot Reply (2)**

Hardware type: Ethernet

Hardware address length: 6

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."

reply

IP addresses: how to get one?

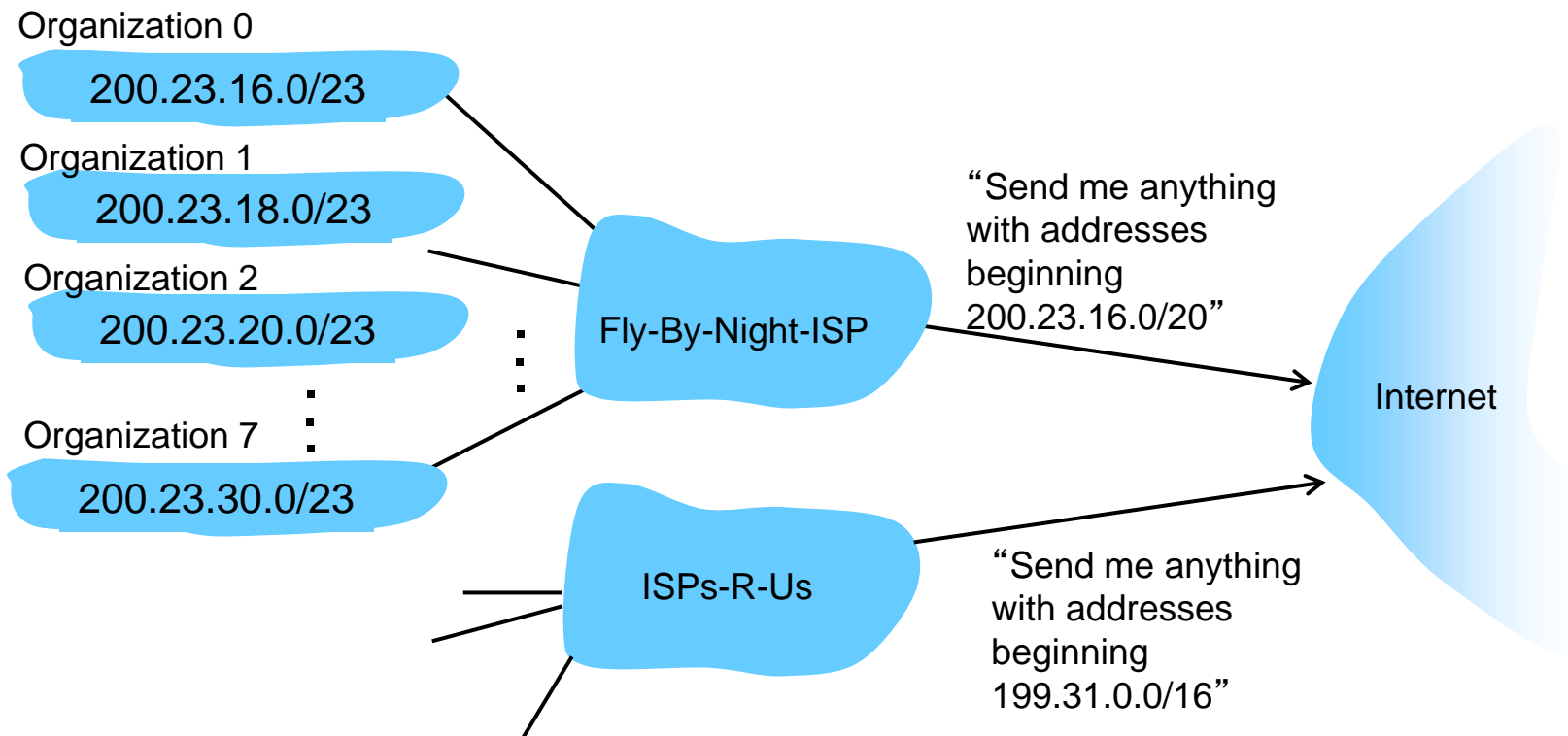
Q: how does *network* get subnet part of IP addr?

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

ISP's block	<u>11001000</u>	<u>00010111</u>	<u>00010000</u>	00000000	200.23.16.0/20
Organization 0	<u>11001000</u>	<u>00010111</u>	<u>00010000</u>	00000000	200.23.16.0/23
Organization 1	<u>11001000</u>	<u>00010111</u>	<u>00010010</u>	00000000	200.23.18.0/23
Organization 2	<u>11001000</u>	<u>00010111</u>	<u>00010100</u>	00000000	200.23.20.0/23
...
Organization 7	<u>11001000</u>	<u>00010111</u>	<u>00011110</u>	00000000	200.23.30.0/23

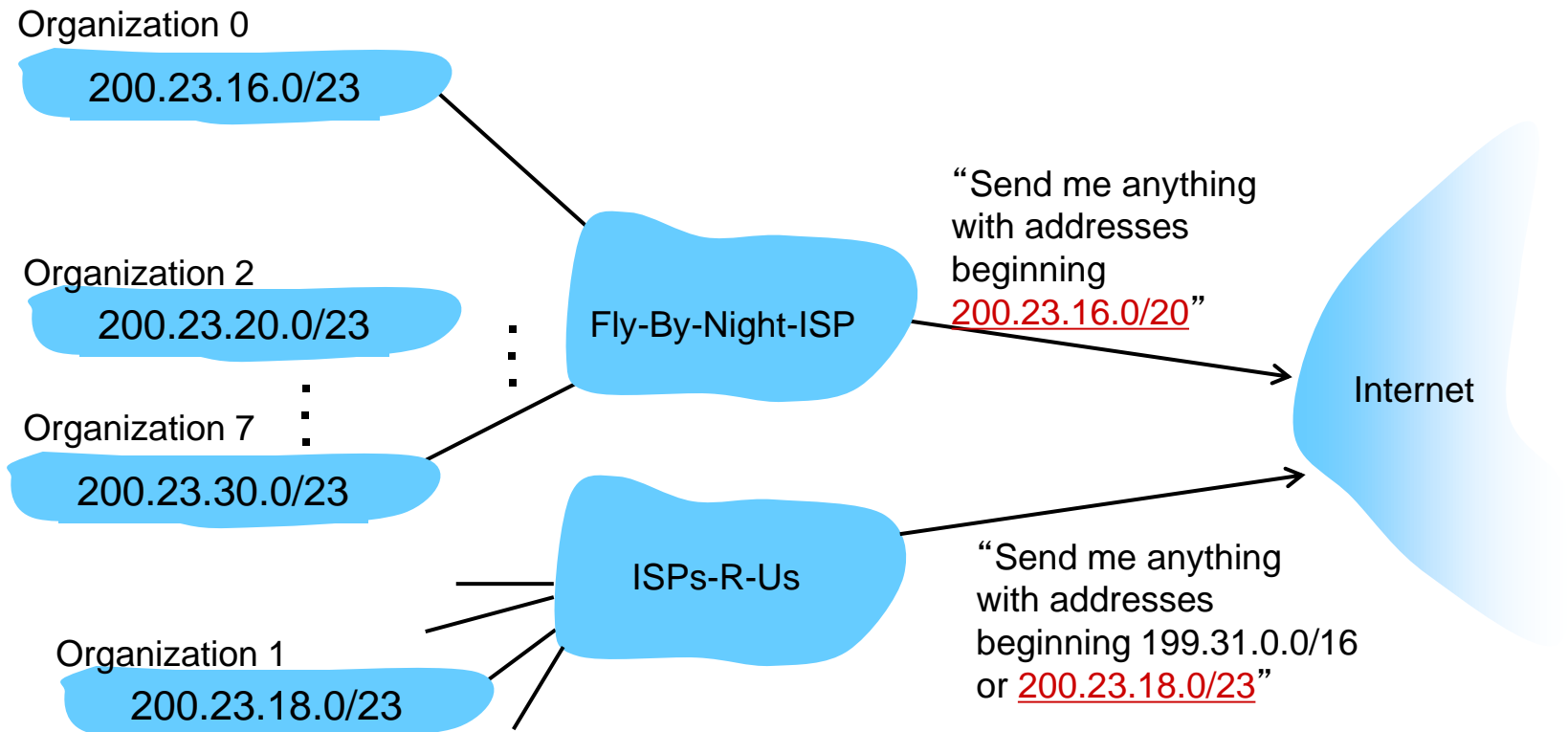
Hierarchical addressing: route aggregation

hierarchical addressing allows efficient advertisement of routing information:



Hierarchical addressing: more specific routes

ISPs-R-U's has a more specific route to Organization 1



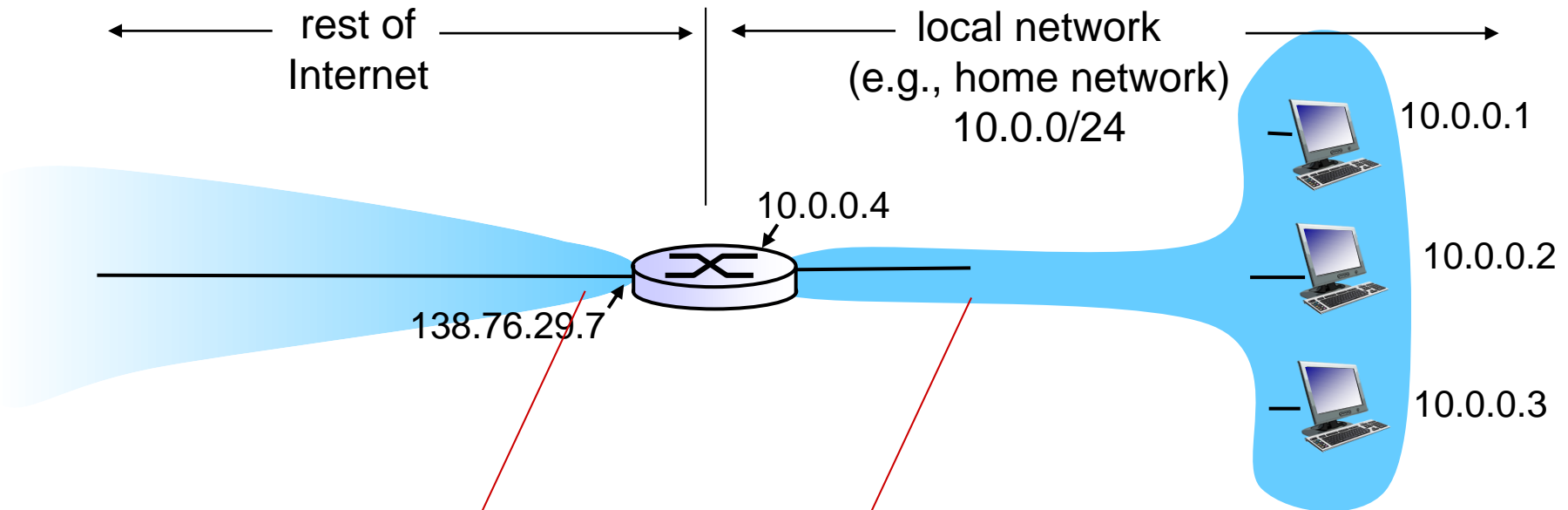
IP addressing: the last word...

Q: how does an ISP get block of addresses?

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

- allocates addresses
- manages DNS
- assigns domain names, resolves disputes

NAT: network address translation



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

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

NAT: network address translation

motivation: local network uses just one IP address as far as outside world is concerned:

- range of addresses not needed from ISP: just one IP address for all devices
- can change addresses of devices in local network without notifying outside world
- can change ISP without changing addresses of devices in local network
- devices inside local net not explicitly addressable, visible by outside world (a security plus)

NAT: network address translation

implementation: NAT router must:

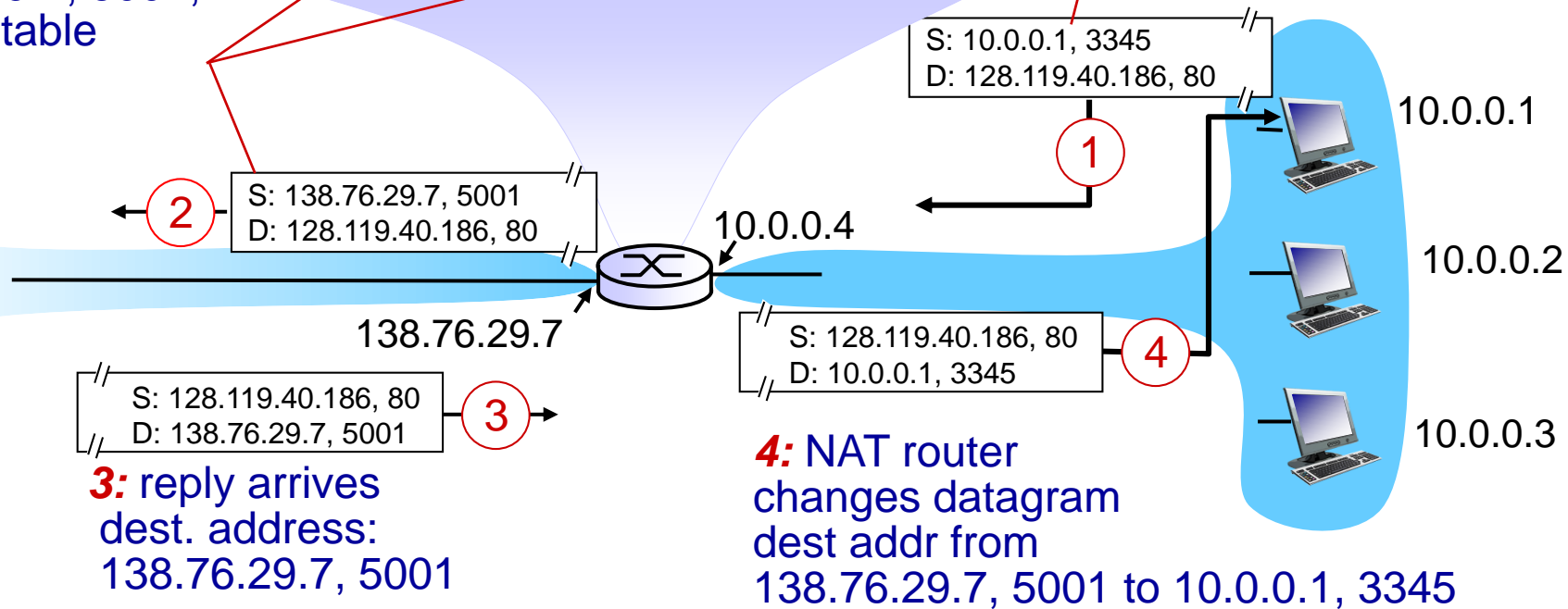
- *outgoing datagrams: replace* (source IP address, port #) of every outgoing datagram to (NAT IP address, new port #)
... remote clients/servers will respond using (NAT IP address, new port #) as destination addr
- *remember (in NAT translation table)* every (source IP address, port #) to (NAT IP address,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

NAT: network address translation

NAT translation table	
WAN side addr	LAN side addr
138.76.29.7, 5001	10.0.0.1, 3345
.....

2: NAT router changes datagram source addr from 10.0.0.1, 3345 to 138.76.29.7, 5001, updates table

1: host 10.0.0.1 sends datagram to 128.119.40.186, 80

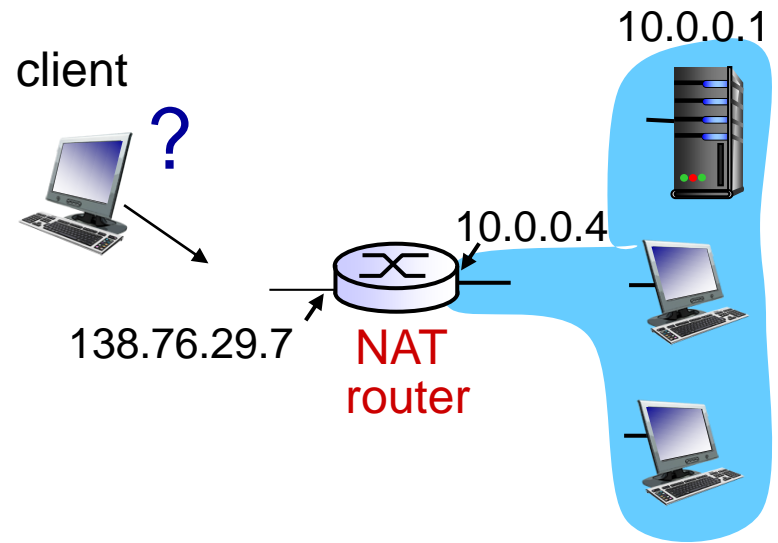


NAT: network address translation

- ❖ 16-bit port-number field:
 - 60,000 simultaneous connections with a single LAN-side address!
- ❖ NAT is controversial:
 - routers should only process up to layer 3
 - violates end-to-end argument
 - NAT possibility must be taken into account by app designers, e.g., P2P applications
 - address shortage should instead be solved by IPv6

NAT traversal problem

- ❖ client wants to connect to server with address 10.0.0.1
 - server address 10.0.0.1 local to LAN (client can't use it as destination addr)
 - only one externally visible NATed address: 138.76.29.7
- ❖ **solution 1:** statically configure NAT to forward incoming connection requests at given port to server
 - e.g., (138.76.29.7, port 2500) always forwarded to 10.0.0.1 port 25000

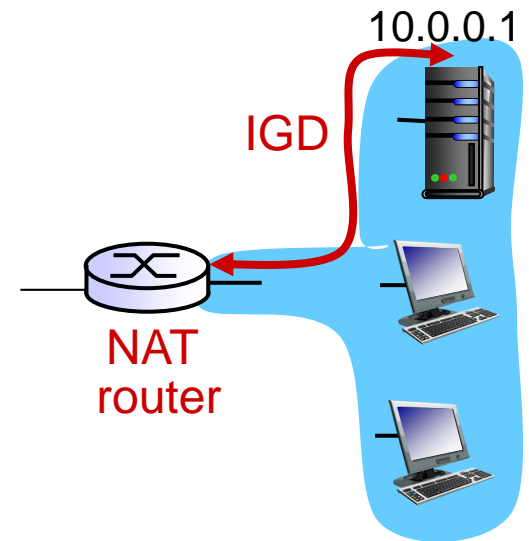


NAT traversal problem

❖ **solution 2:** Universal Plug and Play (UPnP) Internet Gateway Device (IGD) Protocol. Allows NATed host to:

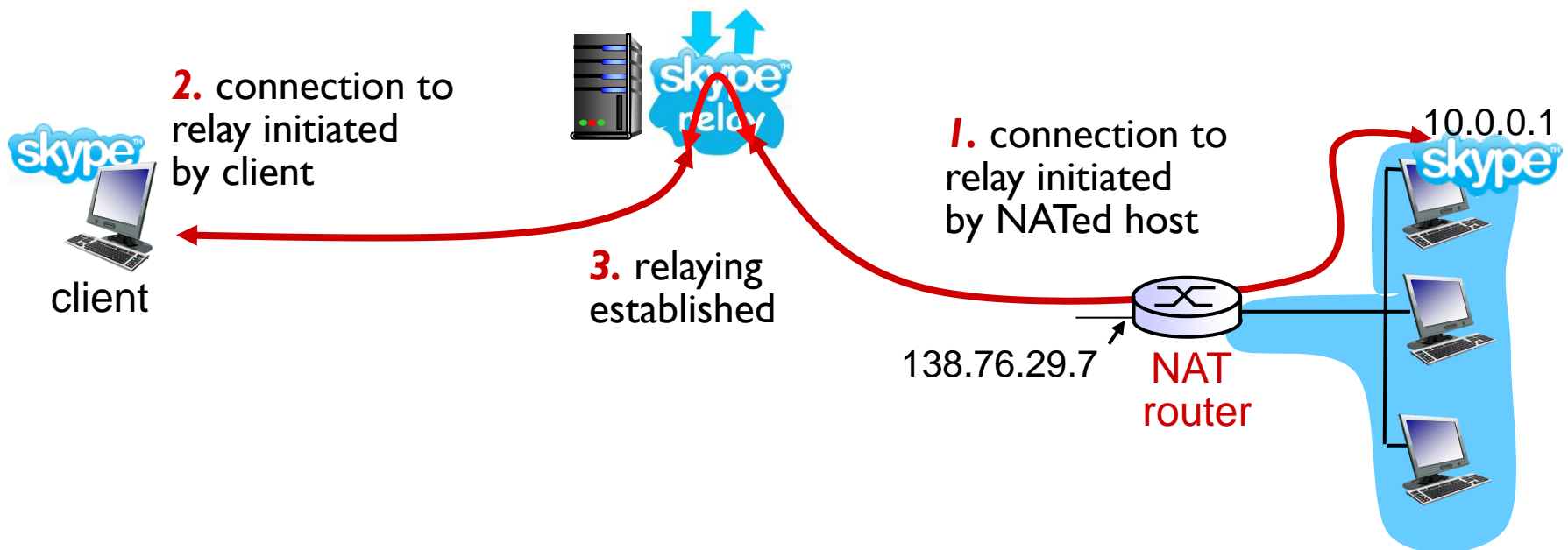
- ❖ learn public IP address (138.76.29.7)
- ❖ add/remove port mappings (with lease times)

i.e., automate static NAT port map configuration



NAT traversal problem

- ❖ **solution 3:** relaying (used in Skype)
 - NATed client establishes connection to relay
 - external client connects to relay
 - relay bridges packets between to connections



ICMP: internet control message protocol

- ❖ used by hosts & routers to communicate network-level information

- error reporting: unreachable host, network, port, protocol
- echo request/reply (used by ping)

- ❖ network-layer “above” IP:

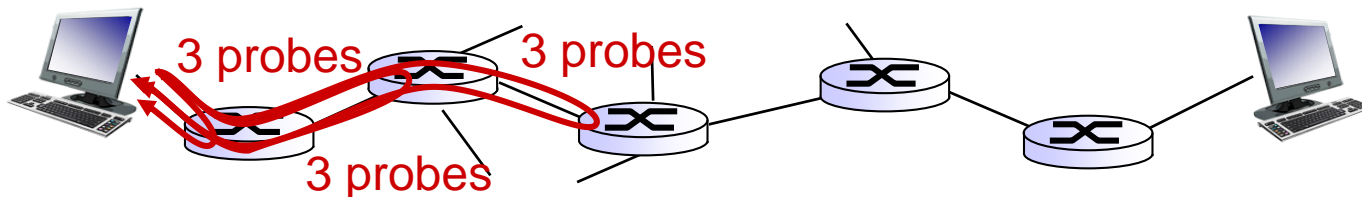
- ICMP msgs carried in IP datagrams

- ❖ **ICMP message:** type, code plus first 8 bytes of IP datagram causing error

<u>Type</u>	<u>Code</u>	<u>description</u>
0	0	echo reply (ping)
3	0	dest. network unreachable
3	1	dest host unreachable
3	2	dest protocol unreachable
3	3	dest port unreachable
3	6	dest network unknown
3	7	dest host unknown
4	0	source quench (congestion control - not used)
8	0	echo request (ping)
9	0	route advertisement
10	0	router discovery
11	0	TTL expired
12	0	bad IP header

Traceroute and ICMP

- ❖ source sends series of UDP segments to dest
 - first set has TTL = 1
 - second set has TTL=2, etc.
 - unlikely port number
 - ❖ when n th set of datagrams arrives to n th router:
 - router discards datagrams
 - and sends source ICMP messages (type 11, code 0)
 - ICMP messages includes name of router & IP address
 - ❖ when ICMP messages arrives, source records RTTs
- stopping criteria:*
- ❖ UDP segment eventually arrives at destination host
 - ❖ destination returns ICMP “port unreachable” message (type 3, code 3)
 - ❖ source stops



IPv6: motivation

- ❖ *initial motivation*: 32-bit address space soon to be completely allocated.
- ❖ additional motivation:
 - header format helps speed processing/forwarding
 - header changes to facilitate QoS

IPv6 datagram format:

- fixed-length 40 byte header
- no fragmentation allowed

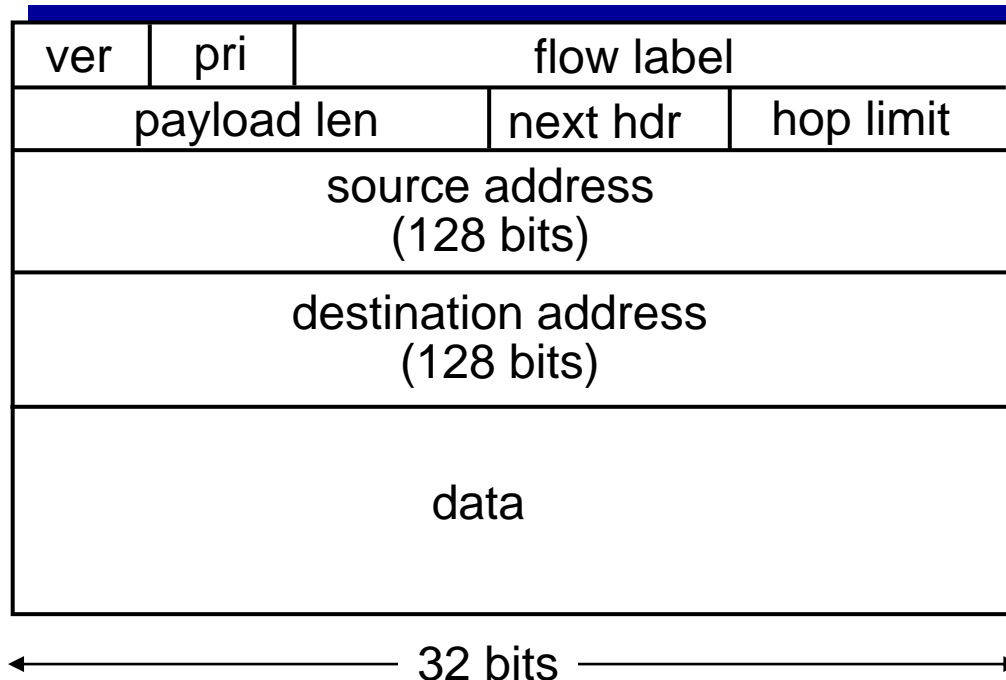
IPv6 datagram format

priority: identify priority among datagrams in flow

flow Label: identify datagrams in same “flow.”

(concept of “flow” not well defined).

next header: identify upper layer protocol for data

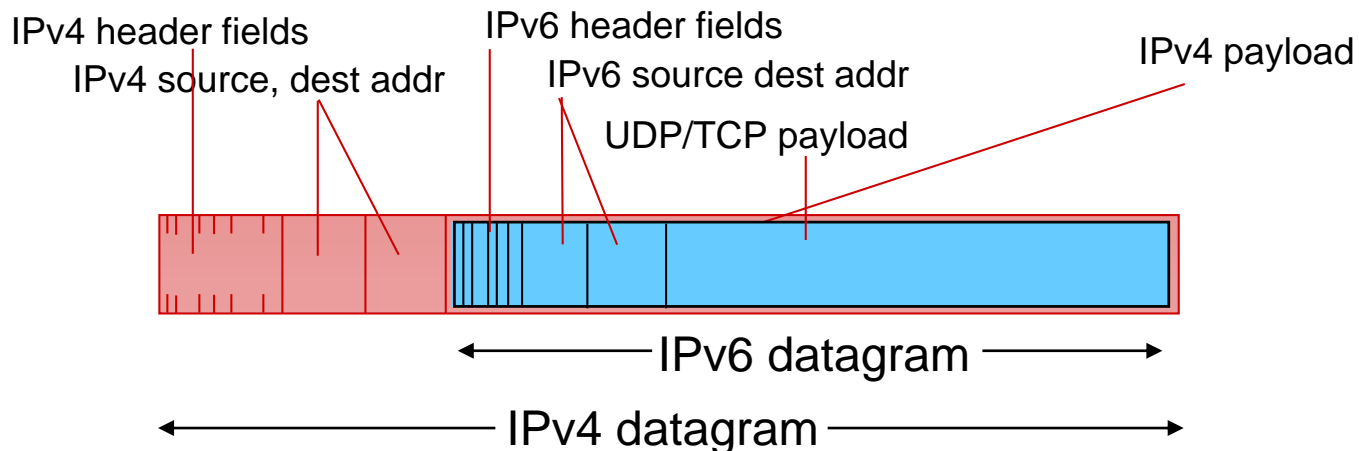


Other changes from IPv4

- ❖ *checksum*: removed entirely to reduce processing time at each hop
- ❖ *options*: allowed, but outside of header, indicated by “Next Header” field
- ❖ *ICMPv6*: new version of ICMP
 - additional message types, e.g. “Packet Too Big”
 - multicast group management functions

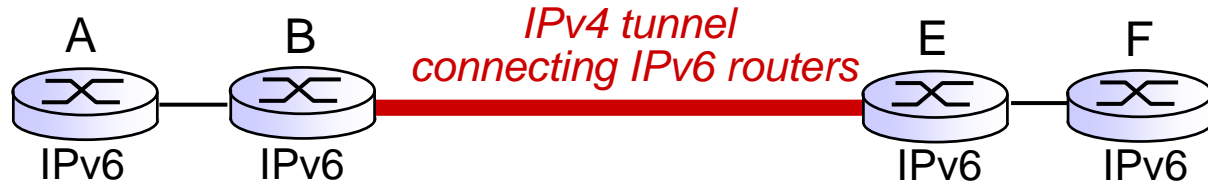
Transition from IPv4 to IPv6

- ❖ 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

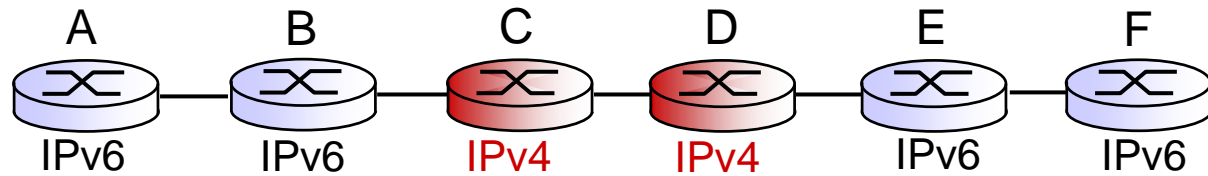


Tunneling

logical view:

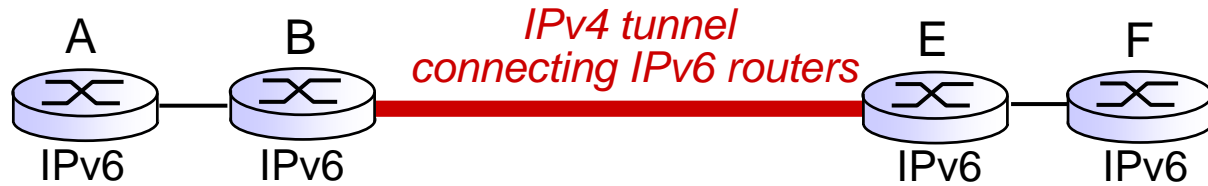


physical view:

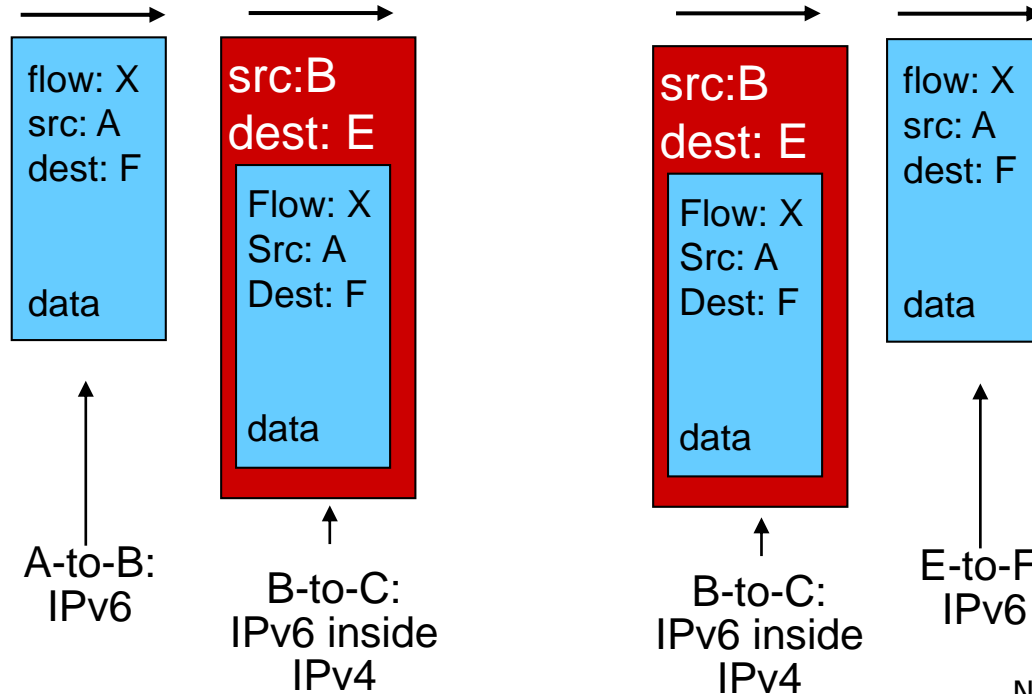
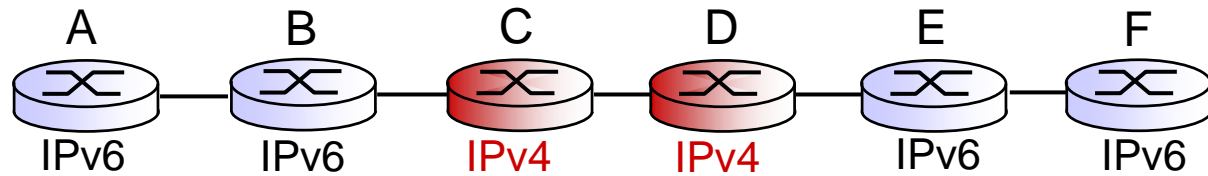


Tunneling

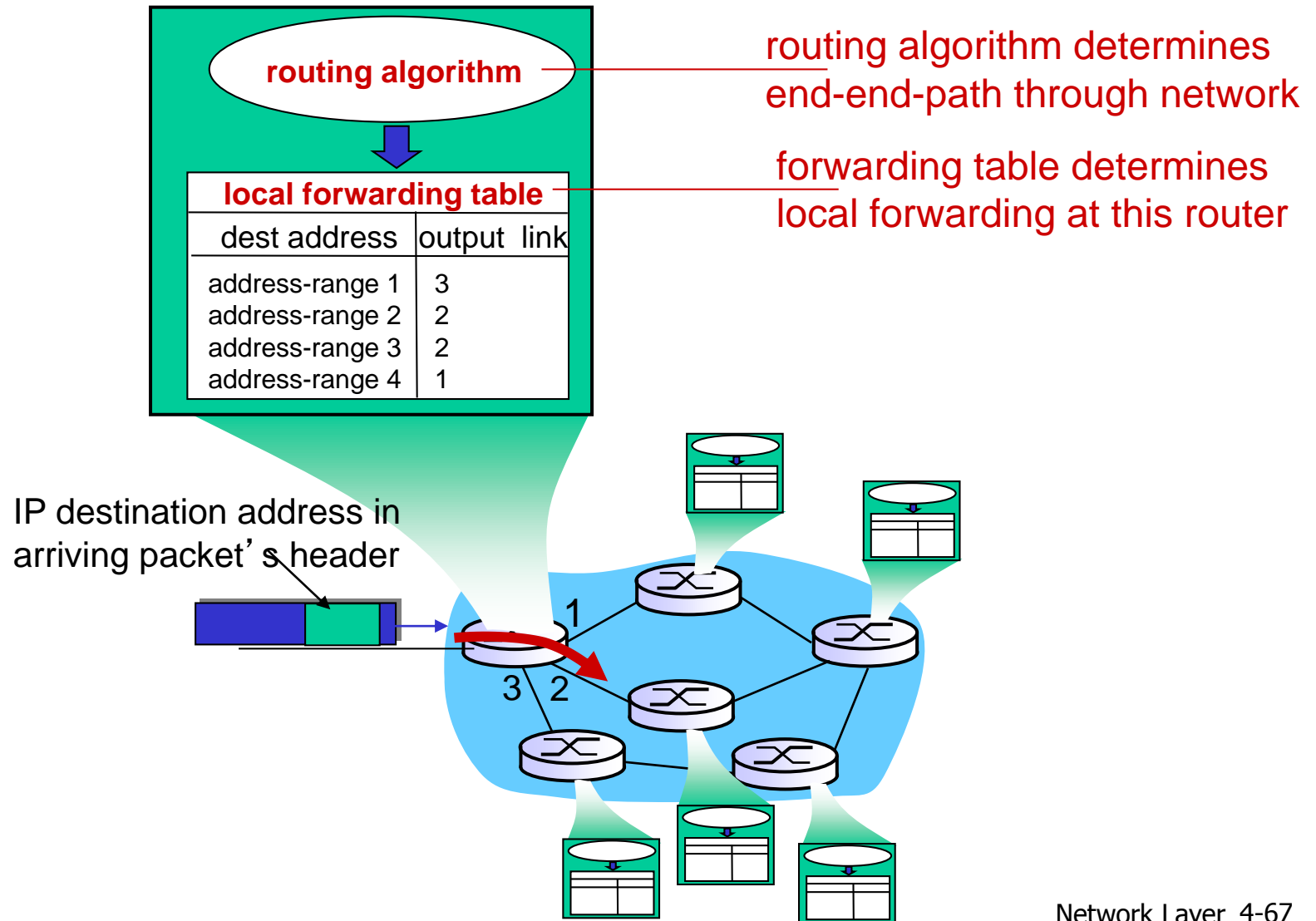
logical view:



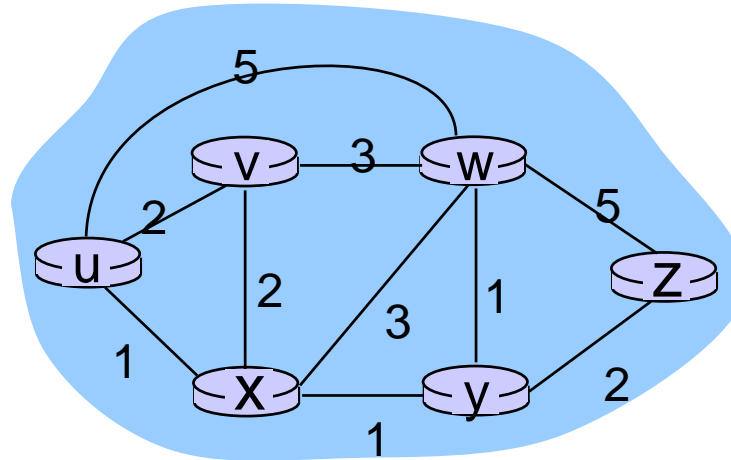
physical view:



Interplay between routing, forwarding



Graph abstraction



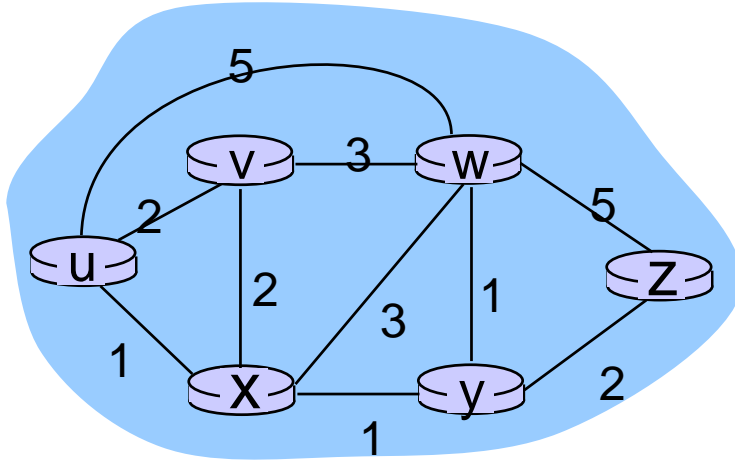
graph: $G = (N,E)$

$N = \text{set of routers} = \{ u, v, w, x, y, z \}$

$E = \text{set of links} = \{ (u,v), (u,x), (v,x), (v,w), (x,w), (x,y), (w,y), (w,z), (y,z) \}$

aside: graph abstraction is useful in other network contexts, e.g., P2P, where N is set of peers and E is set of TCP connections

Graph abstraction: costs



$c(x,x')$ = cost of link (x,x')
e.g., $c(w,z) = 5$

cost could always be 1, or
inversely related to bandwidth,
or inversely related to
congestion

cost of path $(x_1, x_2, x_3, \dots, x_p) = c(x_1, x_2) + c(x_2, x_3) + \dots + c(x_{p-1}, x_p)$

key question: what is the least-cost path between u and z ?
routing algorithm: algorithm that finds that least cost path

Routing algorithm classification

Q: global or decentralized information?

global:

- ❖ all routers have complete topology, link cost info
- ❖ “link state” algorithms

decentralized:

- ❖ router knows physically-connected neighbors, link costs to neighbors
- ❖ iterative process of computation, exchange of info with neighbors
- ❖ “distance vector” algorithms

Q: static or dynamic?

static:

- ❖ routes change slowly over time

dynamic:

- ❖ routes change more quickly
 - periodic update
 - in response to link cost changes

A Link-State Routing Algorithm

Dijkstra's algorithm

- ❖ net topology, link costs known to all nodes
 - accomplished via “link state broadcast”
 - all nodes have same info
- ❖ computes least cost paths from one node (‘source’) to all other nodes
 - gives *forwarding table* for that node
- ❖ iterative: after k iterations, know least cost path to k dest.'s

notation:

- ❖ $C(x,y)$: link cost from node x to y ; $= \infty$ if not direct neighbors
- ❖ $D(v)$: current value of cost of path from source to dest. v
- ❖ $p(v)$: predecessor node along path from source to v
- ❖ N' : set of nodes whose least cost path definitively known

Dijkstra's Algorithm

1 **Initialization:**

2 $N' = \{u\}$

3 for all nodes v

4 if v adjacent to u

5 then $D(v) = c(u,v)$

6 else $D(v) = \infty$

7

8 **Loop**

9 find w not in N' such that $D(w)$ is a minimum

10 add w to N'

11 update $D(v)$ for all v adjacent to w and not in N' :

12 **$D(v) = \min(D(v), D(w) + c(w,v))$**

13 /* new cost to v is either old cost to v or known

14 shortest path cost to w plus cost from w to v */

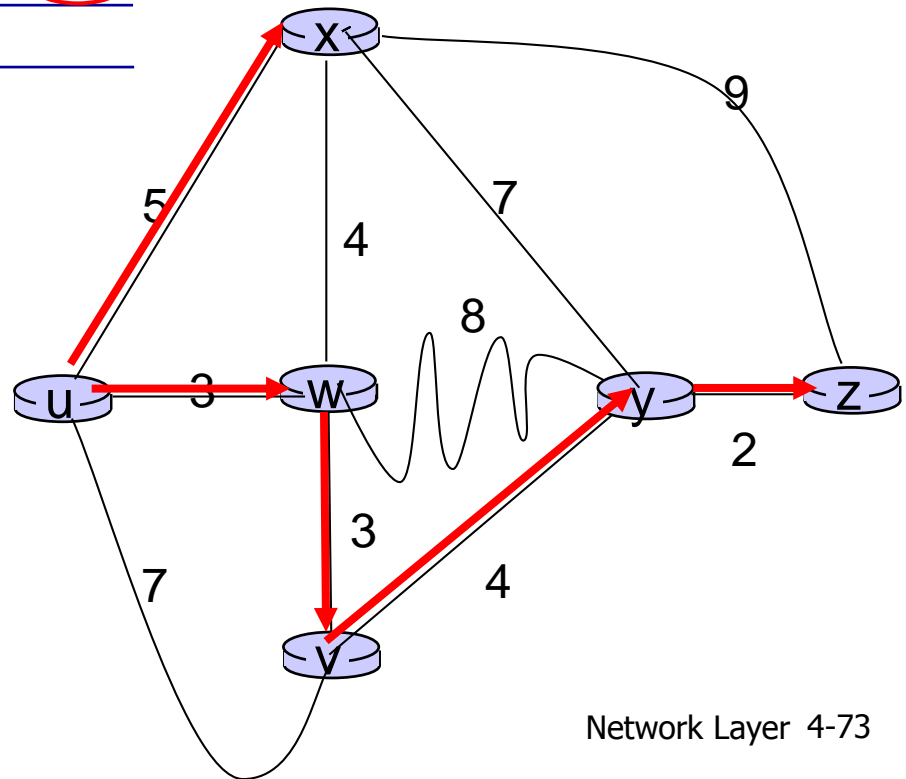
15 **until all nodes in N'**

Dijkstra's algorithm: example

Step	N'	D(v) p(v)	D(w) p(w)	D(x) p(x)	D(y) p(y)	D(z) p(z)
0	u	7,u	3,u	5,u	∞	∞
1	uw	6,w		5,u	11,w	∞
2	uwx	6,w			11,w	14,x
3	uwxv				10,v	14,x
4	uwxvy				12,y	
5	uwxvyz					

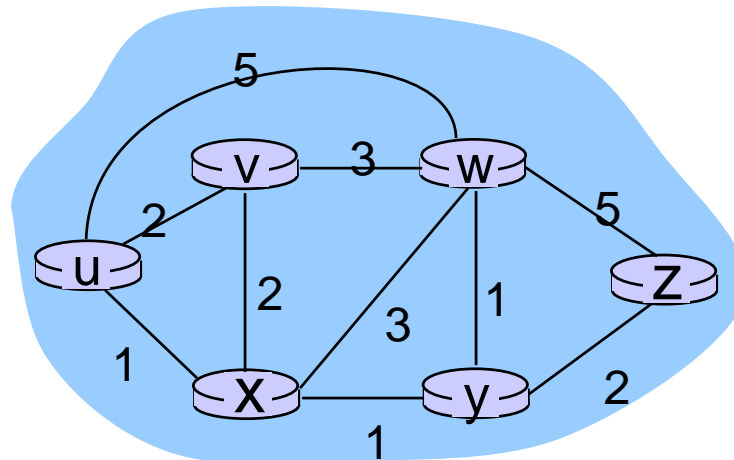
notes:

- ❖ construct shortest path tree by tracing predecessor nodes
- ❖ ties can exist (can be broken arbitrarily)



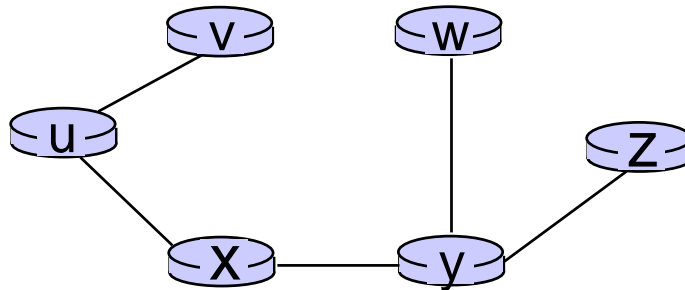
Dijkstra's algorithm: another example

Step	N'	D(v),p(v)	D(w),p(w)	D(x),p(x)	D(y),p(y)	D(z),p(z)
0	u	2,u	5,u	1,u	∞	∞
1	ux	2,u	4,x		2,x	∞
2	uxy	2,u	3,y			4,y
3	uxyv		3,y			4,y
4	uxyvw					4,y
5	uxyvwz					



Dijkstra's algorithm: example (2)

resulting shortest-path tree from u:



resulting forwarding table in u:

destination	link
v	(u,v)
x	(u,x)
y	(u,x)
w	(u,x)
z	(u,x)

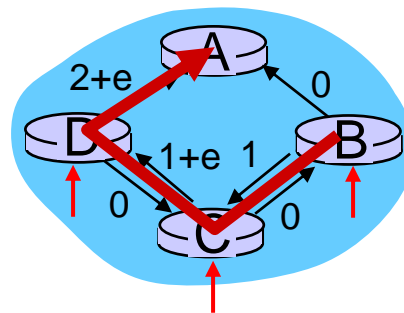
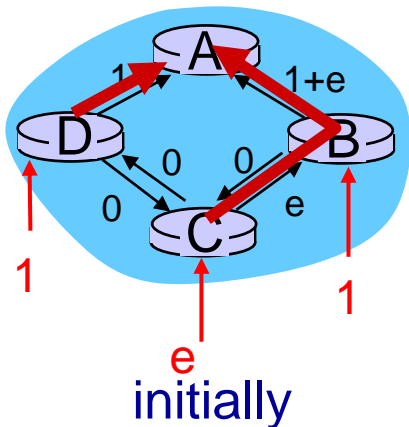
Dijkstra's algorithm, discussion

algorithm complexity: n nodes

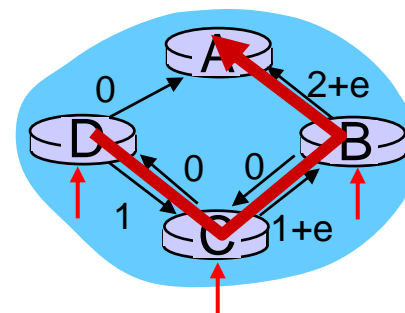
- ❖ each iteration: need to check all nodes, w, not in N
- ❖ $n(n+1)/2$ comparisons: $O(n^2)$
- ❖ more efficient implementations possible: $O(n \log n)$

oscillations possible:

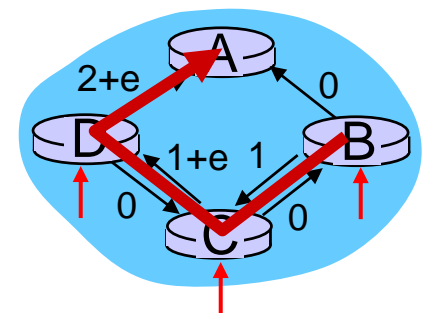
- ❖ e.g., support link cost equals amount of carried traffic:



given these costs,
find new routing....
resulting in new costs



given these costs,
find new routing....
resulting in new costs



given these costs,
find new routing....
resulting in new costs

Distance vector algorithm

Bellman-Ford equation (dynamic programming)

let

$d_x(y) :=$ cost of least-cost path from x to y

then

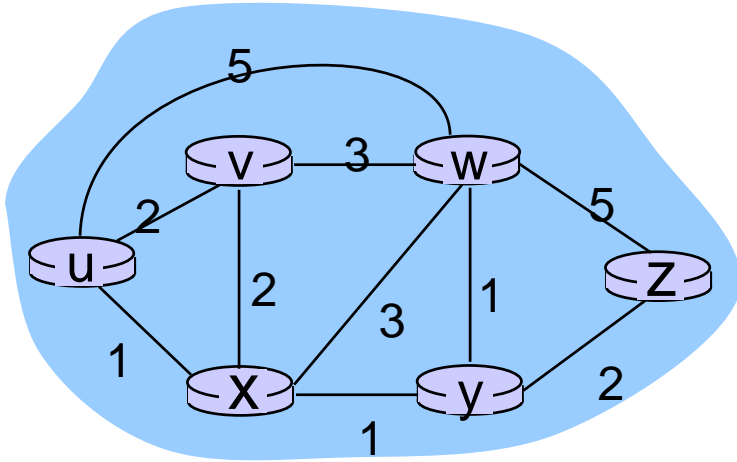
$$d_x(y) = \min_v \{ c(x,v) + d_v(y) \}$$

cost from neighbor v to destination y

cost to neighbor v

\min taken over all neighbors v of x

Bellman-Ford example



clearly, $d_v(z) = 5$, $d_x(z) = 3$, $d_w(z) = 3$

B-F equation says:

$$\begin{aligned} d_u(z) &= \min \{ c(u,v) + d_v(z), \\ &\quad c(u,x) + d_x(z), \\ &\quad c(u,w) + d_w(z) \} \\ &= \min \{ 2 + 5, \\ &\quad 1 + 3, \\ &\quad 5 + 3 \} = 4 \end{aligned}$$

node achieving minimum is next
hop in shortest path, used in forwarding table

Distance vector algorithm

- ❖ $D_x(y)$ = estimate of least cost from x to y
 - x maintains distance vector $\mathbf{D}_x = [D_x(y): y \in N]$
- ❖ node x :
 - knows cost to each neighbor v : $c(x,v)$
 - maintains its neighbors' distance vectors. For each neighbor v , x maintains $\mathbf{D}_v = [D_v(y): y \in N]$

Distance vector algorithm

key idea:

- ❖ from time-to-time, each node sends its own distance vector estimate to neighbors
- ❖ when x receives new DV estimate from neighbor, it updates its own DV using B-F equation:

$$D_x(y) \leftarrow \min_v \{c(x,v) + D_v(y)\} \text{ for each node } y \in N$$

- ❖ under minor, natural conditions, the estimate $D_x(y)$ converge to the actual least cost $d_x(y)$

Distance vector algorithm

iterative, asynchronous:

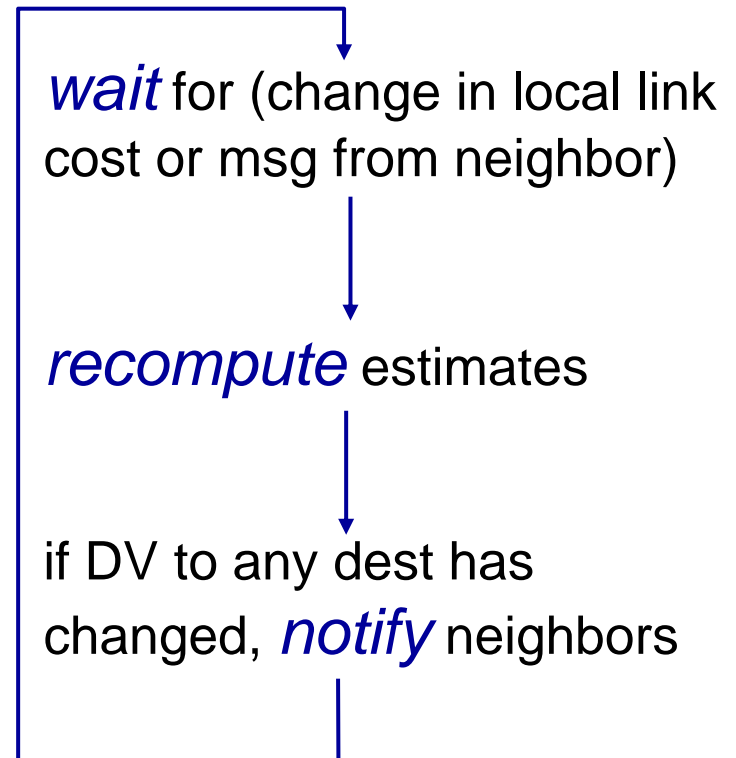
each local iteration
caused by:

- ❖ local link cost change
- ❖ DV update message from neighbor

distributed:

- ❖ each node notifies neighbors *only* when its DV changes
 - neighbors then notify their neighbors if necessary

each node:



$$D_x(y) = \min\{c(x,y) + D_y(y), c(x,z) + D_z(y)\}$$

$$= \min\{2+0, 7+1\} = 2$$

$$D_x(z) = \min\{c(x,y) + D_y(z), c(x,z) + D_z(z)\}$$

$$= \min\{2+1, 7+0\} = 3$$

node x table

		cost to		
		x	y	z
from	x	0	2	7
	y	∞	∞	∞
	z	∞	∞	∞

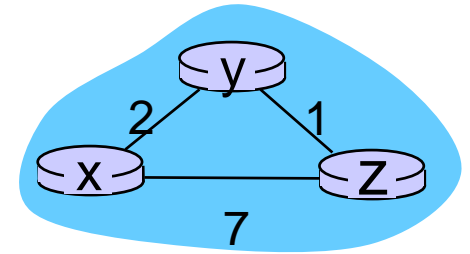
		cost to		
		x	y	z
from	x	0	2	3
	y	2	0	1
	z	7	1	0

node y table

		cost to		
		x	y	z
from	x	∞	∞	∞
	y	2	0	1
	z	∞	∞	∞

node z table

		cost to		
		x	y	z
from	x	∞	∞	∞
	y	∞	∞	∞
	z	7	1	0



time

$$D_x(y) = \min\{c(x,y) + D_y(y), c(x,z) + D_z(y)\}$$

$$= \min\{2+0, 7+1\} = 2$$

$$D_x(z) = \min\{c(x,y) + D_y(z), c(x,z) + D_z(z)\}$$

$$= \min\{2+1, 7+0\} = 3$$

node x table

		cost to		
		x	y	z
from	x	0	2	7
	y	∞	∞	∞
	z	∞	∞	∞

node y table

		cost to		
		x	y	z
from	x	∞	∞	∞
	y	2	0	1
	z	∞	∞	∞

node z table

		cost to		
		x	y	z
from	x	∞	∞	∞
	y	∞	∞	∞
	z	7	1	0

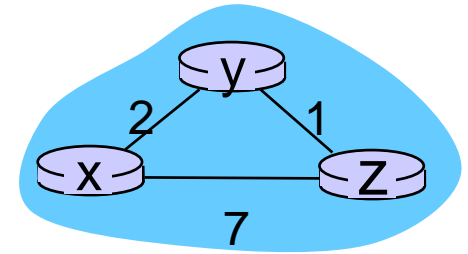
		cost to		
		x	y	z
from	x	0	2	3
	y	2	0	1
	z	7	1	0

		cost to		
		x	y	z
from	x	0	2	7
	y	2	0	1
	z	7	1	0

		cost to		
		x	y	z
from	x	0	2	3
	y	2	0	1
	z	3	1	0

		cost to		
		x	y	z
from	x	0	2	3
	y	2	0	1
	z	3	1	0

		cost to		
		x	y	z
from	x	0	2	3
	y	2	0	1
	z	3	1	0

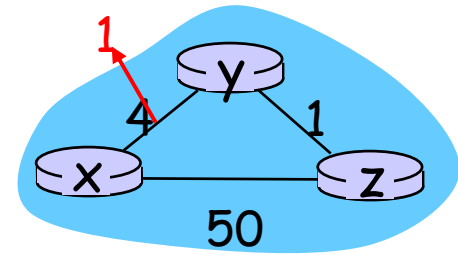


time →

Distance vector: link cost changes

link cost changes:

- ❖ node detects local link cost change
- ❖ updates routing info, recalculates distance vector
- ❖ if DV changes, notify neighbors



“good
news
travels
fast”

t_0 : y detects link-cost change, updates its DV, informs its neighbors.

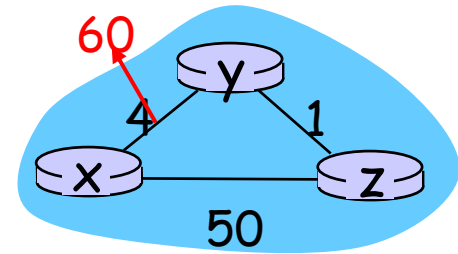
t_1 : z receives update from y , updates its table, computes new least cost to x , sends its neighbors its DV.

t_2 : y receives z 's update, updates its distance table. y 's least costs do *not* change, so y does *not* send a message to z .

Distance vector: link cost changes

link cost changes:

- ❖ node detects local link cost change
- ❖ *bad news travels slow* - “count to infinity” problem!
- ❖ 44 iterations before algorithm stabilizes: see text



poisoned reverse:

- ❖ If Z routes through Y to get to X :
 - Z tells Y its (Z' s) distance to X is infinite (so Y won' t route to X via Z)
- ❖ will this completely solve count to infinity problem?

Comparison of LS and DV algorithms

message complexity

- ❖ **LS:** with n nodes, E links, $O(nE)$ msgs sent
- ❖ **DV:** exchange between neighbors only
 - convergence time varies

speed of convergence

- ❖ **LS:** $O(n^2)$ algorithm requires $O(nE)$ msgs
 - may have oscillations
- ❖ **DV:** convergence time varies
 - may be routing loops
 - count-to-infinity problem

robustness: what happens if router malfunctions?

LS:

- node can advertise incorrect *link* cost
- each node computes only its own table

DV:

- DV node can advertise incorrect *path* cost
- each node's table used by others
 - error propagate thru network

Hierarchical routing

our routing study thus far - idealization

- ❖ all routers identical

- ❖ network “flat”

... *not* true in practice

scale: with 600 million destinations:

- ❖ can't store all dest's in routing tables!

- ❖ routing table exchange would swamp links!

administrative autonomy

- ❖ internet = network of networks

- ❖ each network admin may want to control routing in its own network

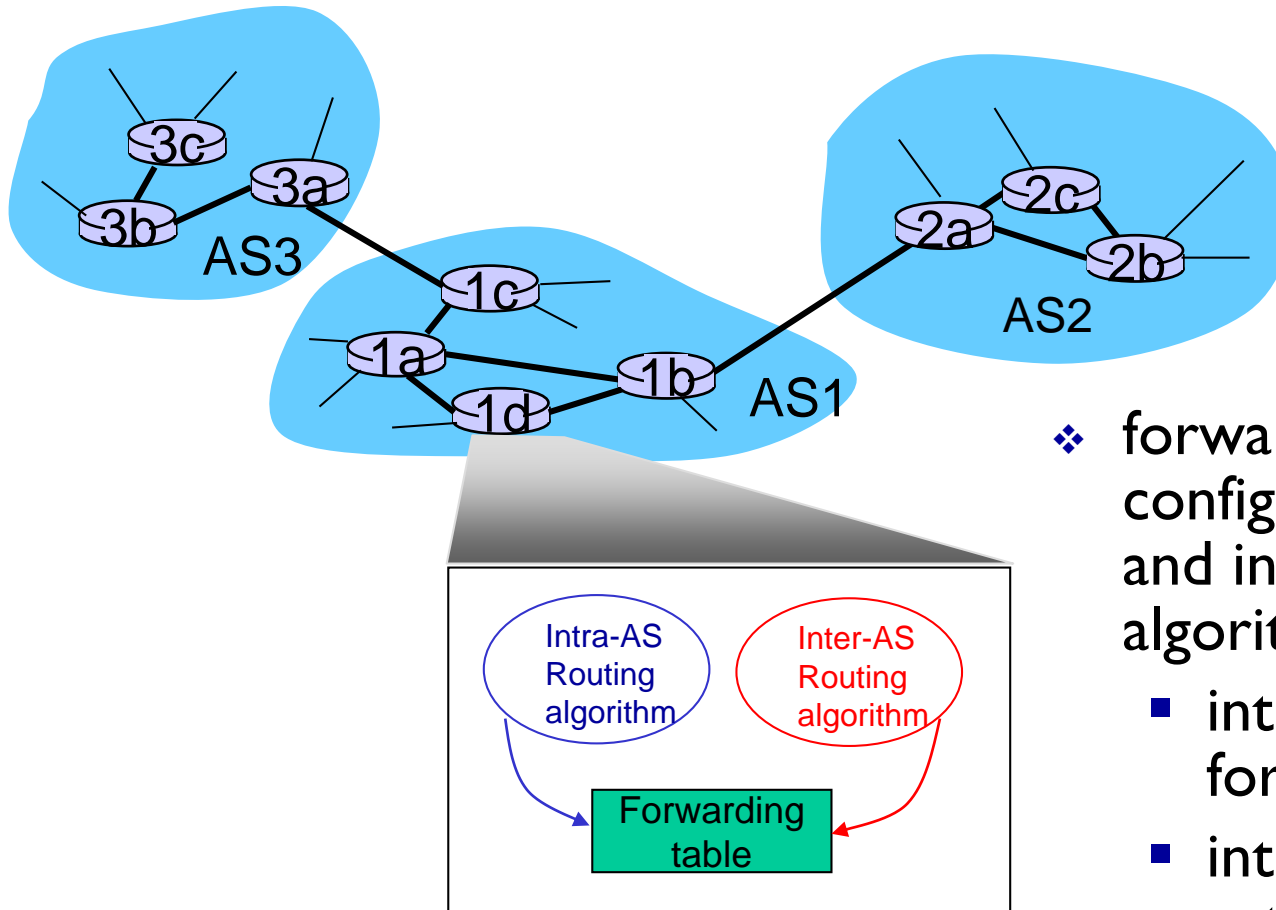
Hierarchical routing

- ❖ aggregate routers into regions, “**autonomous systems**” (AS)
- ❖ routers in same AS run same routing protocol
 - “**intra-AS**” routing protocol
 - routers in different AS can run different intra-AS routing protocol

gateway router:

- ❖ at “edge” of its own AS
- ❖ has link to router in another AS

Interconnected ASes



- ❖ forwarding table configured by both intra- and inter-AS routing algorithm
 - intra-AS sets entries for internal dests
 - inter-AS & intra-AS sets entries for external dests

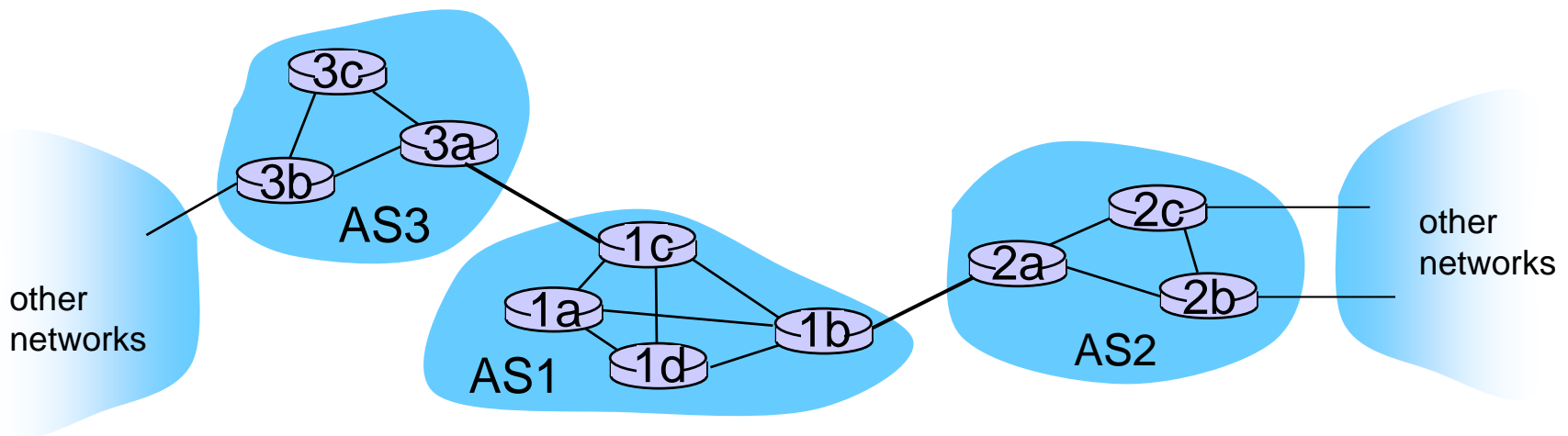
Inter-AS tasks

- ❖ suppose router in AS1 receives datagram destined outside of AS1:
 - router should forward packet to gateway router, but which one?

AS1 must:

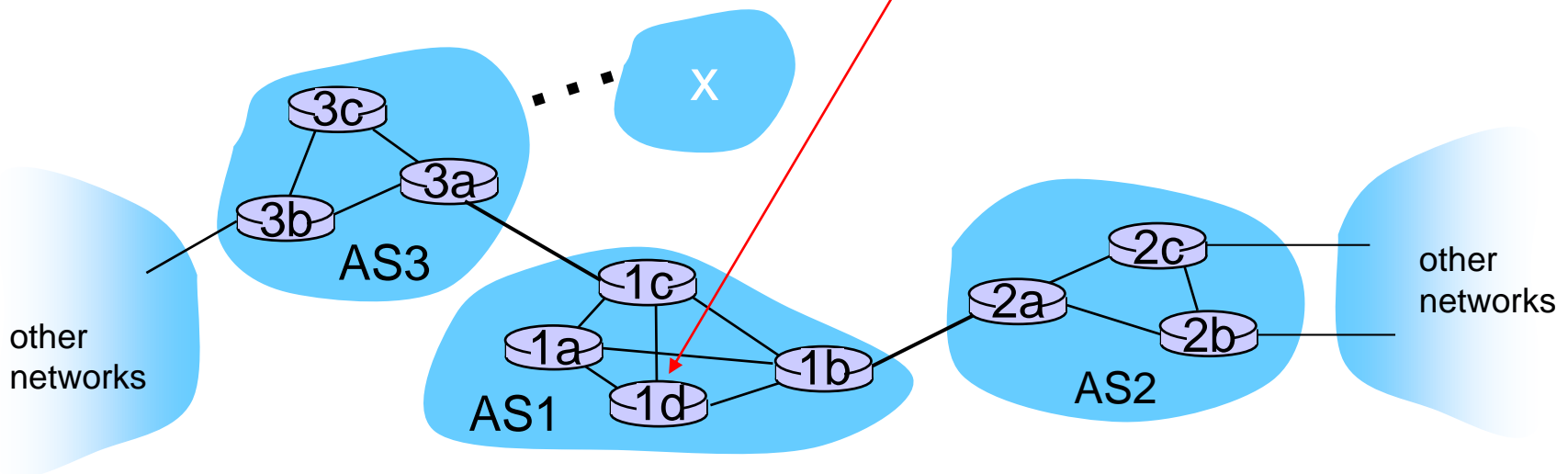
1. learn which dests are reachable through AS2, which through AS3
2. propagate this reachability info to all routers in AS1

job of inter-AS routing!



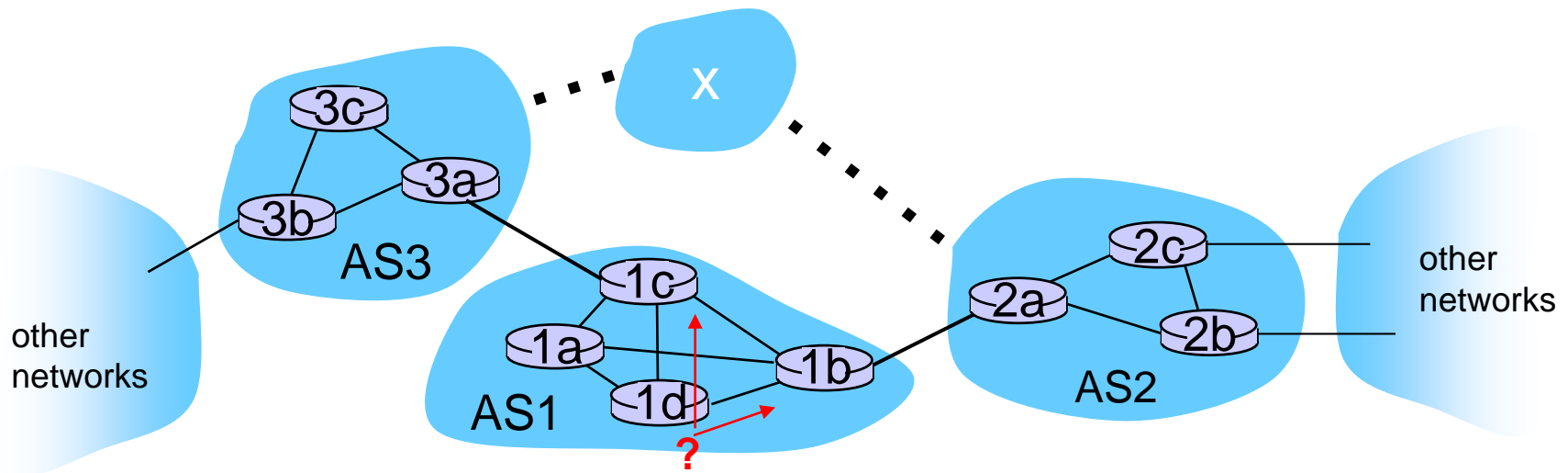
Example: setting forwarding table in router 1d

- ❖ suppose AS1 learns (via inter-AS protocol) that subnet x reachable via AS3 (gateway 1c), but not via AS2
 - inter-AS protocol propagates reachability info to all internal routers
- ❖ router 1d determines from intra-AS routing info that its interface l is on the least cost path to 1c
 - installs forwarding table entry (x, l)



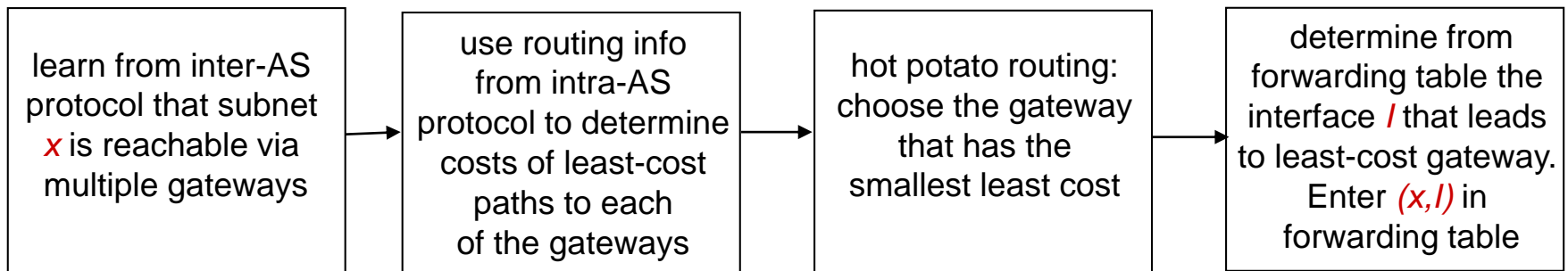
Example: choosing among multiple ASes

- ❖ now suppose AS1 learns from inter-AS protocol that subnet **x** is reachable from AS3 *and* from AS2.
- ❖ to configure forwarding table, router 1d must determine which gateway it should forward packets towards for dest **x**
 - this is also job of inter-AS routing protocol!



Example: choosing among multiple ASes

- ❖ now suppose AS1 learns from inter-AS protocol that subnet x is reachable from AS3 *and* from AS2.
- ❖ to configure forwarding table, router Id must determine towards which gateway it should forward packets for dest x
 - this is also job of inter-AS routing protocol!
- ❖ *hot potato routing: send* packet towards closest of two routers.

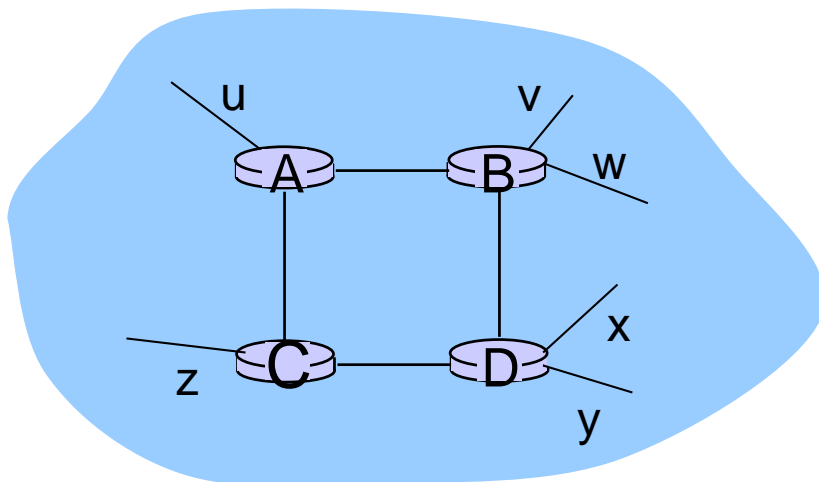


Intra-AS Routing

- ❖ also known as *interior gateway protocols (IGP)*
- ❖ most common intra-AS routing protocols:
 - RIP: Routing Information Protocol
 - OSPF: Open Shortest Path First
 - IGRP: Interior Gateway Routing Protocol (Cisco proprietary)

RIP (Routing Information Protocol)

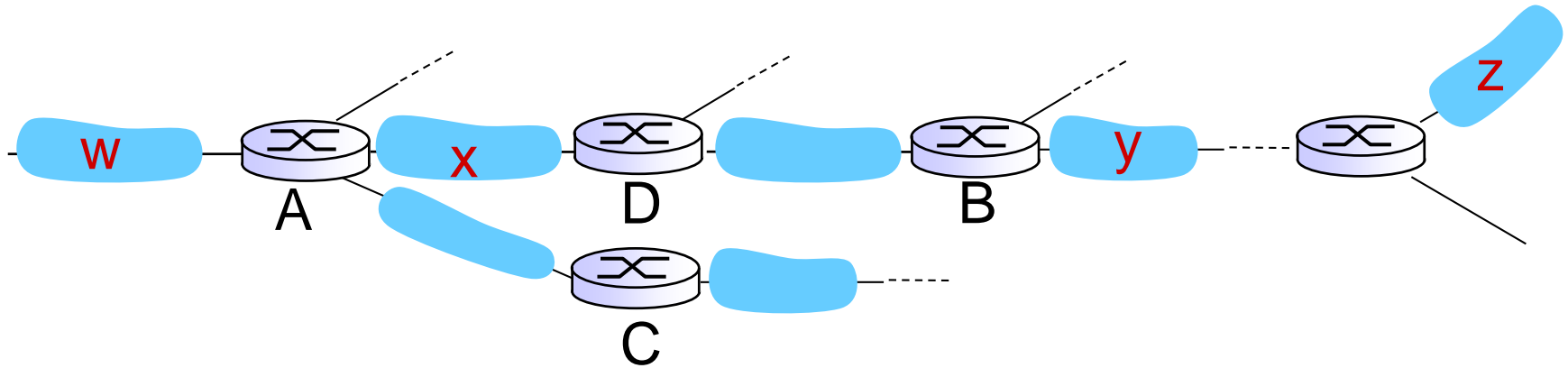
- ❖ included in BSD-UNIX distribution in 1982
- ❖ distance vector algorithm
 - distance metric: # hops (max = 15 hops), each link has cost 1
 - DVs exchanged with neighbors every 30 sec in response message (aka **advertisement**)
 - each advertisement: list of up to 25 destination **subnets** (in IP addressing sense)



from router A to destination **subnets**:

<u>subnet</u>	<u>hops</u>
u	1
v	2
w	2
x	3
y	3
z	2

RIP: example



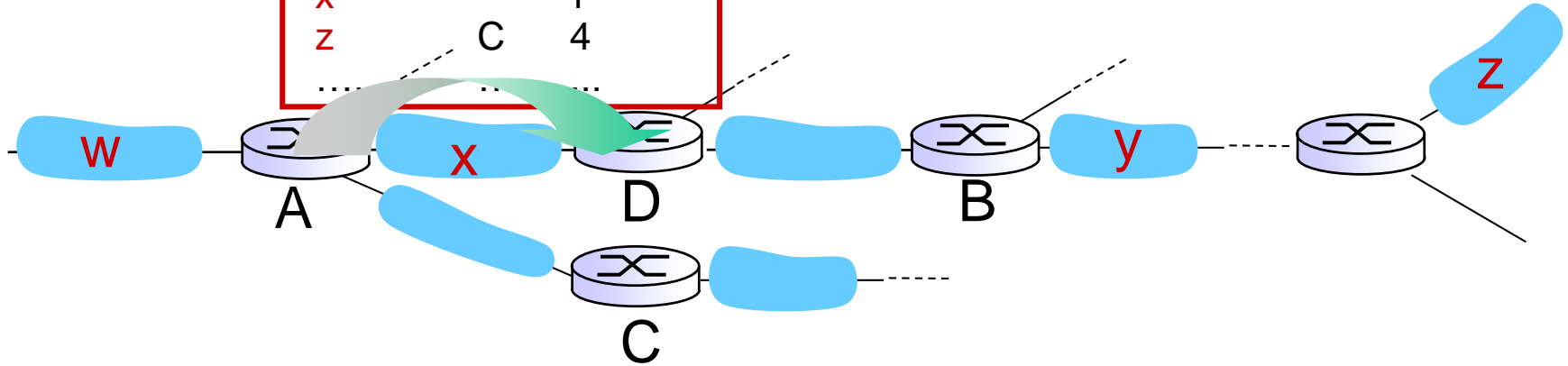
routing table in router D

destination subnet	next router	# hops to dest
W	A	2
y	B	2
Z	B	7
X	--	1
....

RIP: example

A-to-D advertisement

dest	next hops	hops
W	-	1
X	-	1
Z	C	4
.....



routing table in router D

destination subnet	next router	# hops to dest
W	A	2
y	B	2
Z	B → A	7 → 5
X	--	1
.....

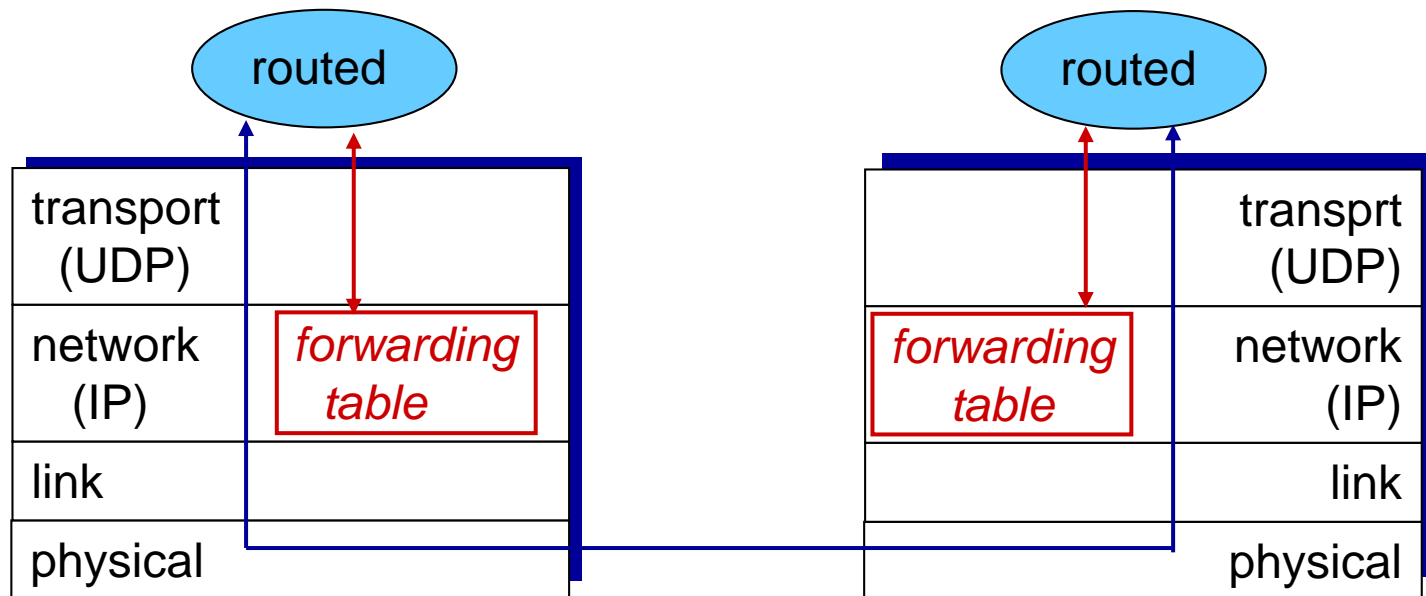
RIP: link failure, recovery

if no advertisement heard after 180 sec -->
neighbor/link declared dead

- routes via neighbor invalidated
- new advertisements sent to neighbors
- neighbors in turn send out new advertisements (if tables changed)
- link failure info quickly (?) propagates to entire net
- *poison reverse* used to prevent ping-pong loops (infinite distance = 16 hops)

RIP table processing

- ❖ RIP routing tables managed by *application-level* process called route-d (daemon)
- ❖ advertisements sent in UDP packets, periodically repeated



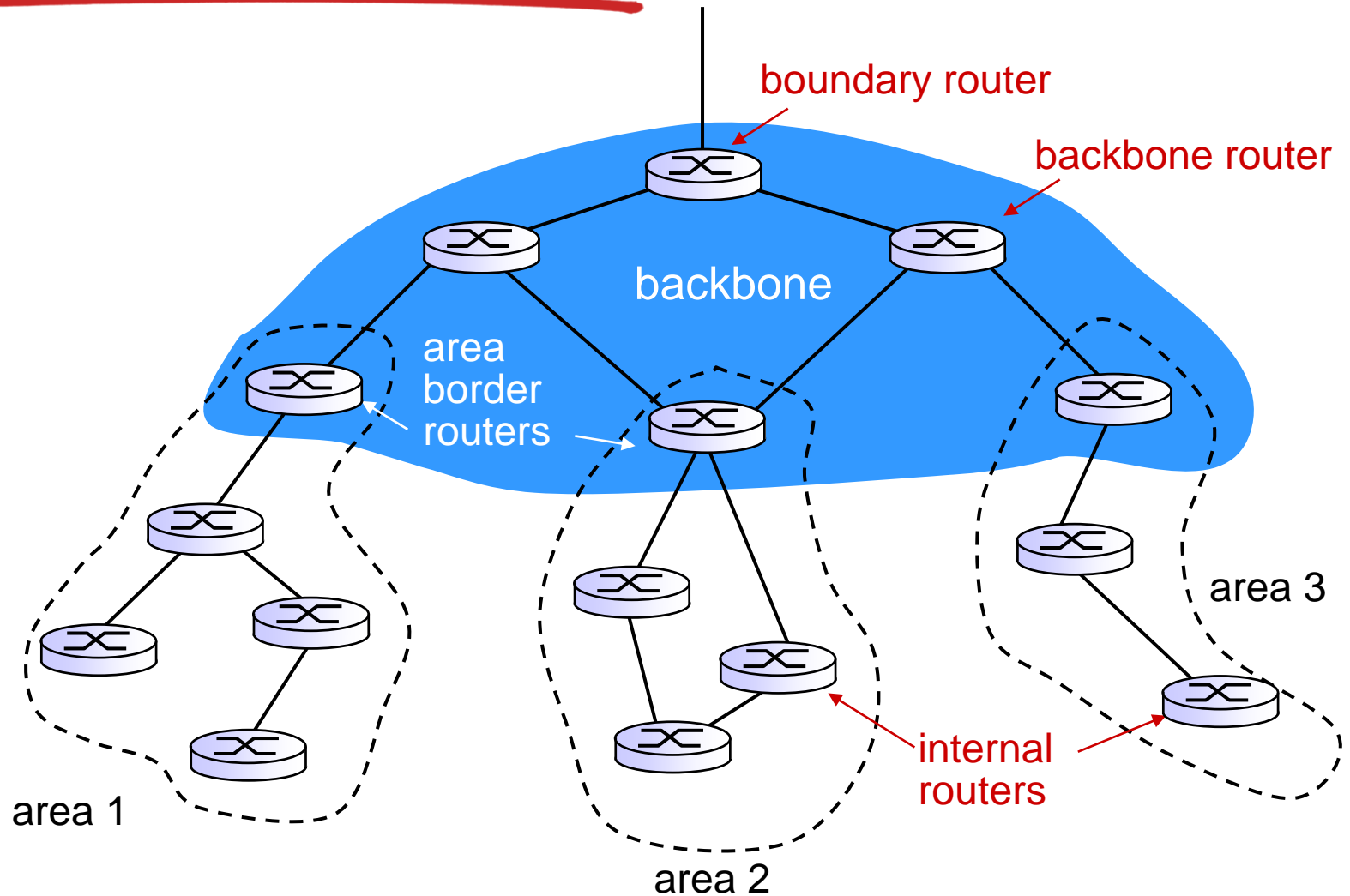
OSPF (Open Shortest Path First)

- ❖ “open”: publicly available
- ❖ uses link state algorithm
 - LS packet dissemination
 - topology map at each node
 - route computation using Dijkstra’s algorithm
- ❖ OSPF advertisement carries one entry per neighbor
- ❖ advertisements flooded to *entire* AS
 - carried in OSPF messages directly over IP (rather than TCP or UDP)
- ❖ *IS-IS routing* protocol: nearly identical to OSPF

OSPF “advanced” features (not in RIP)

- ❖ **security**: all OSPF messages authenticated (to prevent malicious intrusion)
- ❖ **multiple** same-cost **paths** allowed (only one path in RIP)
- ❖ for each link, multiple cost metrics for different **TOS** (e.g., satellite link cost set “low” for best effort ToS; high for real time ToS)
- ❖ integrated uni- and **multicast** support:
 - Multicast OSPF (MOSPF) uses same topology data base as OSPF
- ❖ **hierarchical** OSPF in large domains.

Hierarchical OSPF



Hierarchical OSPF

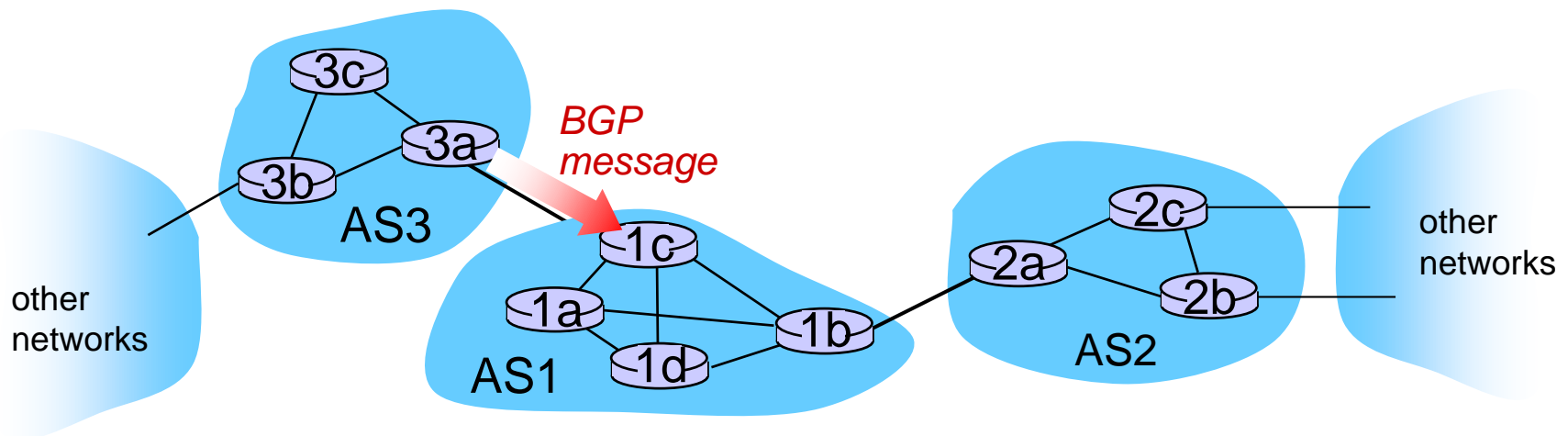
- ❖ *two-level hierarchy*: local area, backbone.
 - link-state advertisements only in area
 - each nodes has detailed area topology; only know direction (shortest path) to nets in other areas.
- ❖ *area border routers*: “summarize” distances to nets in own area, advertise to other Area Border routers.
- ❖ *backbone routers*: run OSPF routing limited to backbone.
- ❖ *boundary routers*: connect to other AS' s.

Internet inter-AS routing: BGP

- ❖ **BGP (Border Gateway Protocol):** *the de facto inter-domain routing protocol*
 - “glue that holds the Internet together”
- ❖ BGP provides each AS a means to:
 - **eBGP:** obtain subnet reachability information from neighboring ASs.
 - **iBGP:** propagate reachability information to all AS-internal routers.
 - determine “good” routes to other networks based on reachability information and policy.
- ❖ allows subnet to advertise its existence to rest of Internet: *“I am here”*

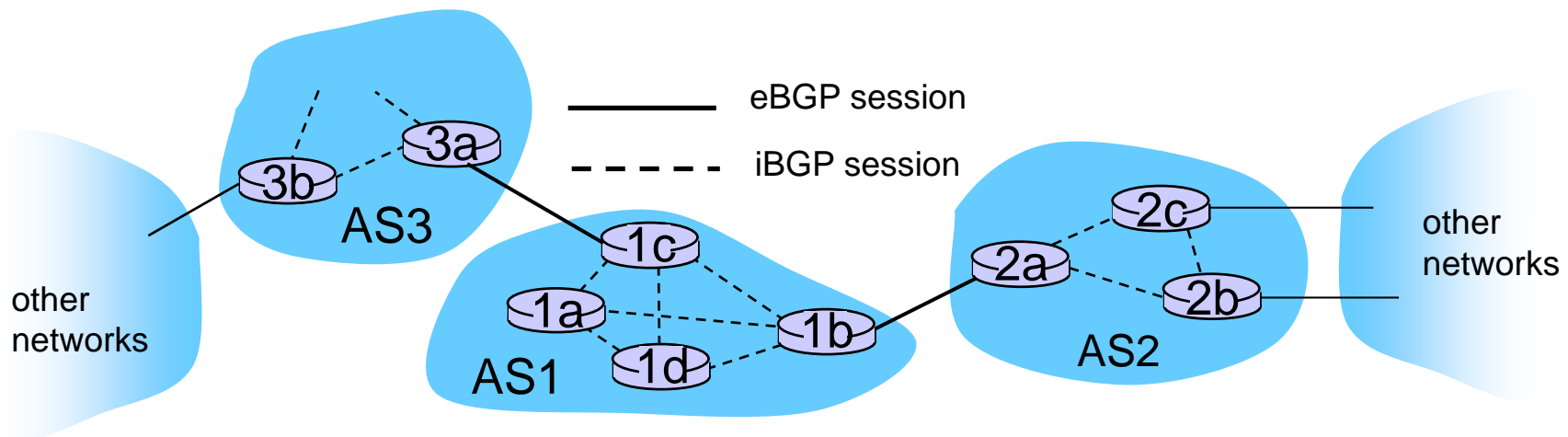
BGP basics

- ❖ **BGP session:** two BGP routers (“peers”) exchange BGP messages:
 - advertising *paths* to different destination network prefixes (“path vector” protocol)
 - exchanged over semi-permanent TCP connections
- ❖ when AS3 advertises a prefix to AS1:
 - AS3 *promises* it will forward datagrams towards that prefix
 - AS3 can aggregate prefixes in its advertisement



BGP basics: distributing path information

- ❖ using eBGP session between 3a and 1c, AS3 sends prefix reachability info to AS1.
 - 1c can then use iBGP to distribute new prefix info to all routers in AS1
 - 1b can then re-advertise new reachability info to AS2 over 1b-to-2a eBGP session
- ❖ when router learns of new prefix, it creates entry for prefix in its forwarding table.



Path attributes and BGP routes

- ❖ advertised prefix includes BGP attributes
 - prefix + attributes = “route”
- ❖ two important attributes:
 - **AS-PATH**: contains ASs through which prefix advertisement has passed: e.g., AS 67, AS 17
 - **NEXT-HOP**: indicates specific internal-AS router to next-hop AS. (may be multiple links from current AS to next-hop-AS)
- ❖ gateway router receiving route advertisement uses **import policy** to accept/decline
 - e.g., never route through AS x
 - *policy-based* routing

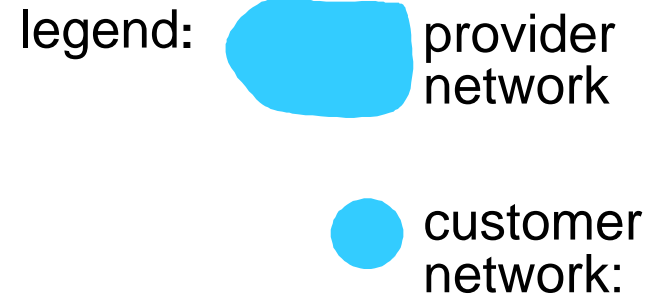
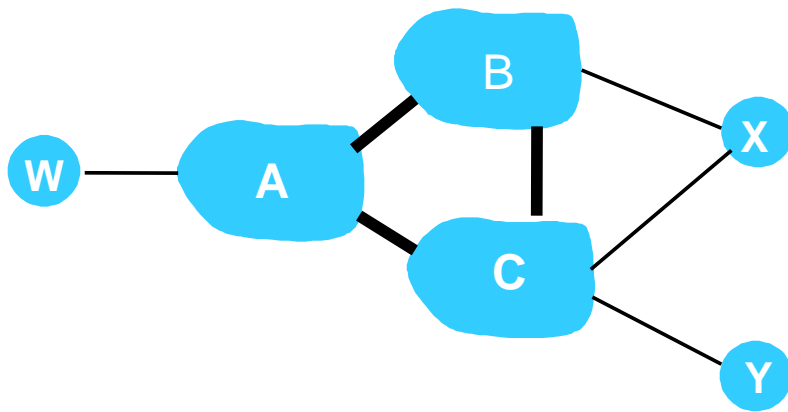
BGP route selection

- ❖ router may learn about more than 1 route to destination AS, selects route based on:
 1. local preference value attribute: policy decision
 2. shortest AS-PATH
 3. closest NEXT-HOP router: hot potato routing
 4. additional criteria

BGP messages

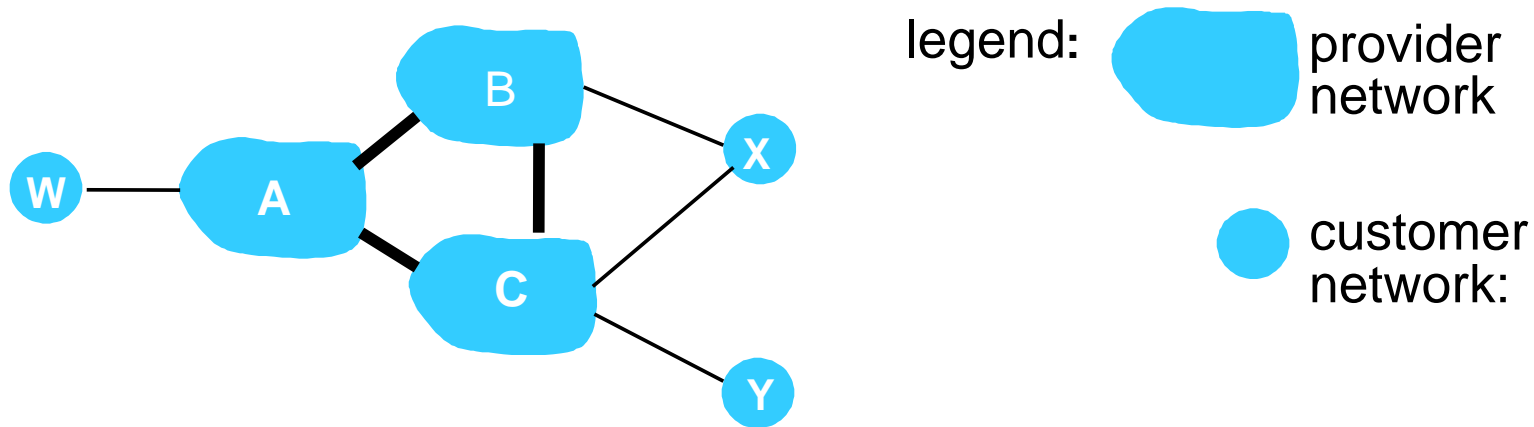
- ❖ BGP messages exchanged between peers over TCP connection
- ❖ BGP messages:
 - **OPEN:** opens TCP connection to peer and authenticates sender
 - **UPDATE:** advertises new path (or withdraws old)
 - **KEEPALIVE:** keeps connection alive in absence of UPDATES; also ACKs OPEN request
 - **NOTIFICATION:** reports errors in previous msg; also used to close connection

BGP routing policy



- ❖ A,B,C are *provider networks*
- ❖ X,W,Y are customer (of provider networks)
- ❖ X is *dual-homed*: attached to two networks
 - X does not want to route from B via X to C
 - .. so X will not advertise to B a route to C

BGP routing policy (2)



- ❖ A advertises path AW to B
- ❖ B advertises path BAW to X
- ❖ Should B advertise path BAW to C?
 - No way! B gets no “revenue” for routing $CBAW$ since neither W nor C are B’s customers
 - B wants to force C to route to w via A
 - B wants to route *only* to/from its customers!

Why different Intra-, Inter-AS routing ?

policy:

- ❖ inter-AS: admin wants control over how its traffic routed, who routes through its net.
- ❖ intra-AS: single admin, so no policy decisions needed

scale:

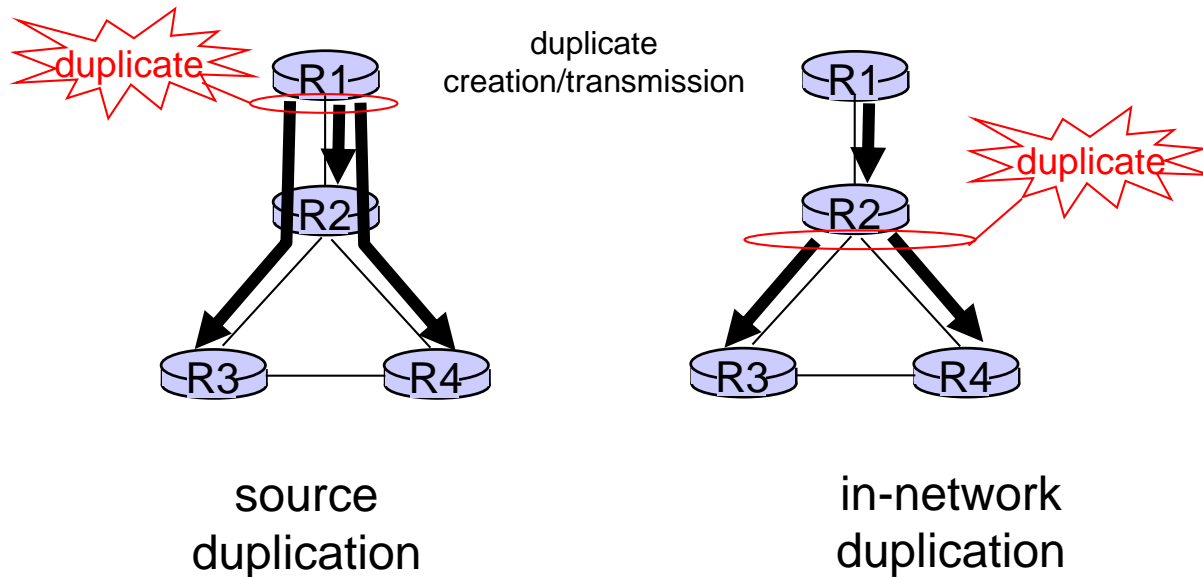
- ❖ hierarchical routing saves table size, reduced update traffic

performance:

- ❖ intra-AS: can focus on performance
- ❖ inter-AS: policy may dominate over performance

Broadcast routing

- ❖ deliver packets from source to all other nodes
- ❖ source duplication is inefficient:



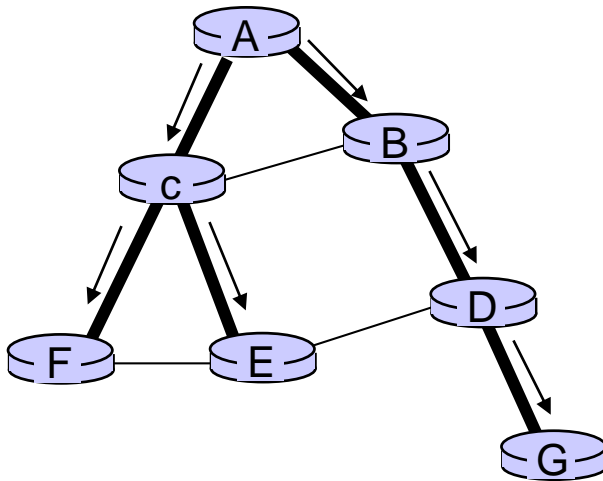
- ❖ source duplication: how does source determine recipient addresses?

In-network duplication

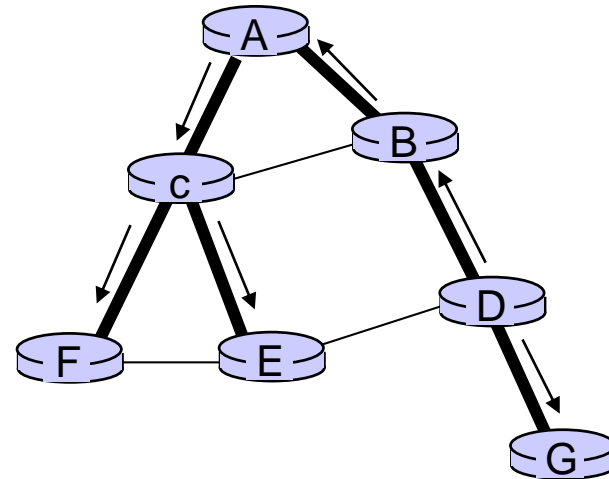
- ❖ *flooding*: when node receives broadcast packet, sends copy to all neighbors
 - problems: cycles & broadcast storm
- ❖ *controlled flooding*: node only broadcasts pkt if it hasn't broadcast same packet before
 - node keeps track of packet ids already broadcasted
 - or reverse path forwarding (RPF): only forward packet if it arrived on shortest path between node and source
- ❖ *spanning tree*:
 - no redundant packets received by any node

Spanning tree

- ❖ first construct a spanning tree
- ❖ nodes then forward/make copies only along spanning tree



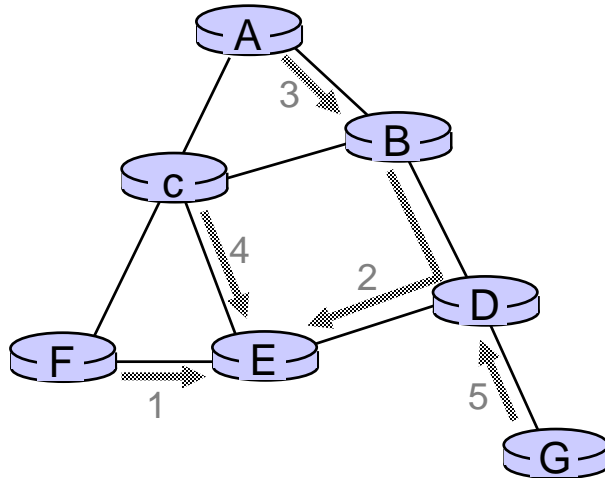
(a) broadcast initiated at A



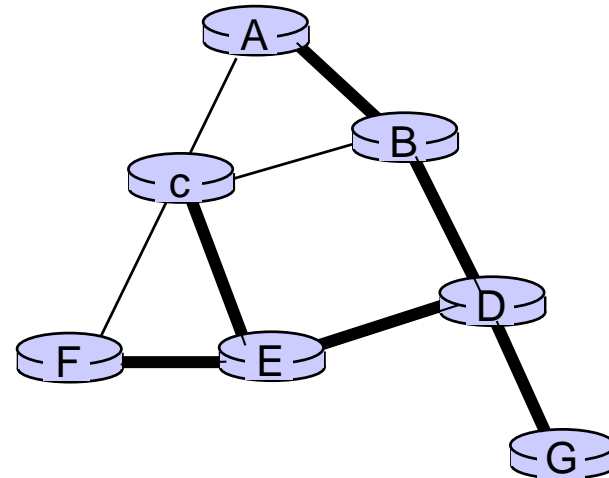
(b) broadcast initiated at D

Spanning tree: creation

- ❖ center node
- ❖ each node sends unicast join message to center node
 - message forwarded until it arrives at a node already belonging to spanning tree



(a) stepwise construction of spanning tree (center: E)



(b) constructed spanning tree

Multicast routing: problem statement

goal: find a tree (or trees) connecting routers having local mcast group members

- ❖ *tree:* not all paths between routers used
- ❖ *shared-tree:* same tree used by all group members
- ❖ *source-based:* different tree from each sender to rcvrs

legend



group member



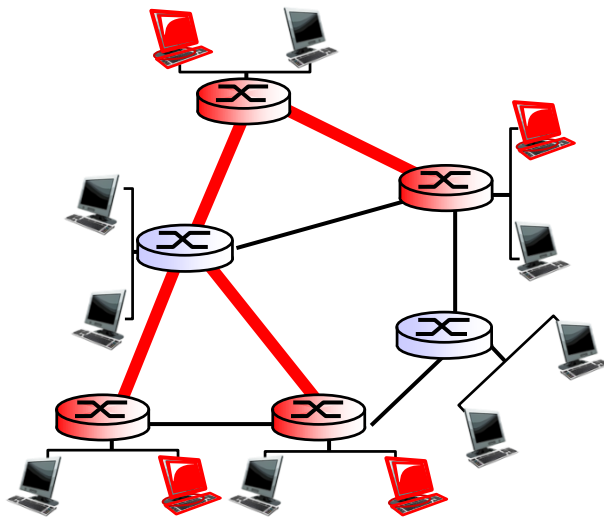
not group member



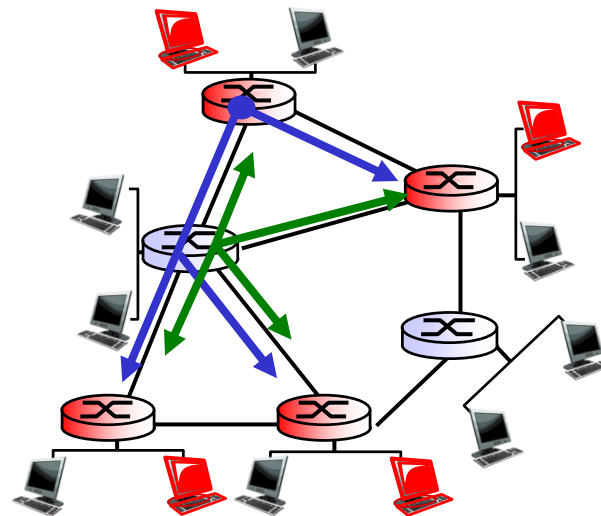
router with a group member



router without group member



shared tree



source-based trees

Approaches for building mcast trees

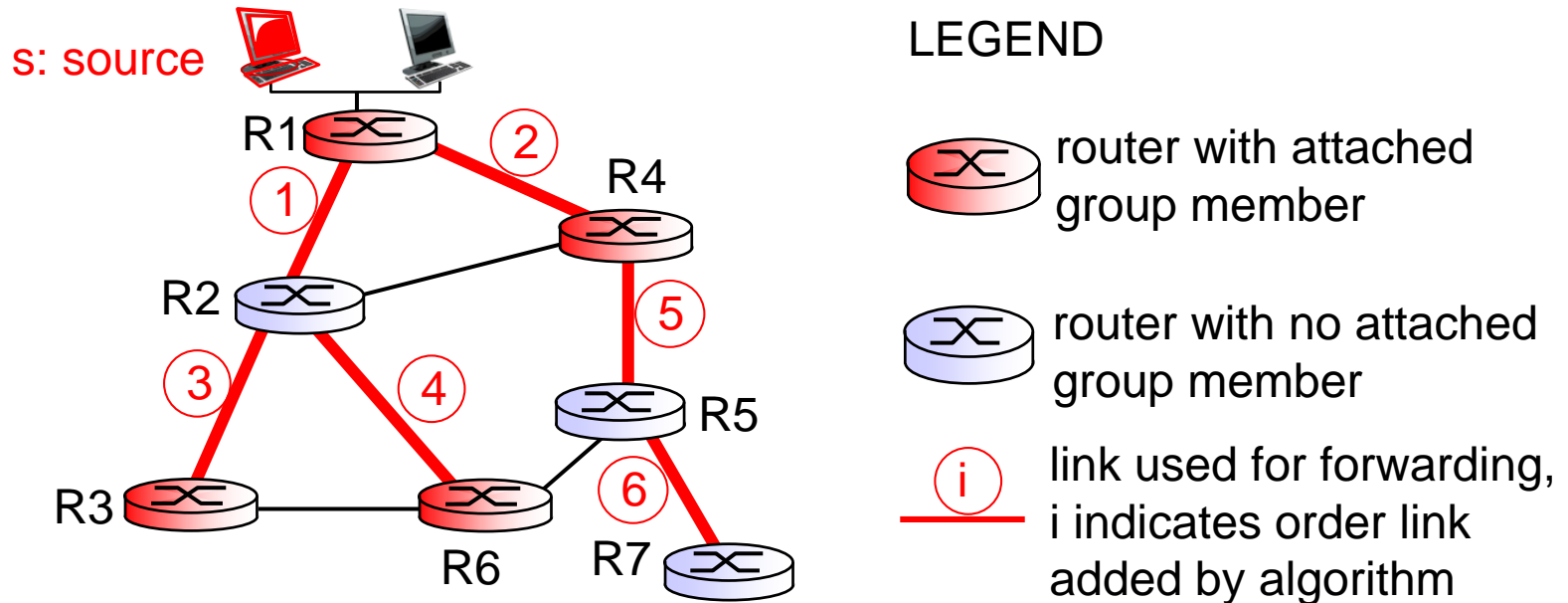
approaches:

- ❖ *source-based tree*: one tree per source
 - shortest path trees
 - reverse path forwarding
- ❖ *group-shared tree*: group uses one tree
 - minimal spanning (Steiner)
 - center-based trees

...we first look at basic approaches, then specific protocols adopting these approaches

Shortest path tree

- ❖ mcast forwarding tree: tree of shortest path routes from source to all receivers
 - Dijkstra's algorithm

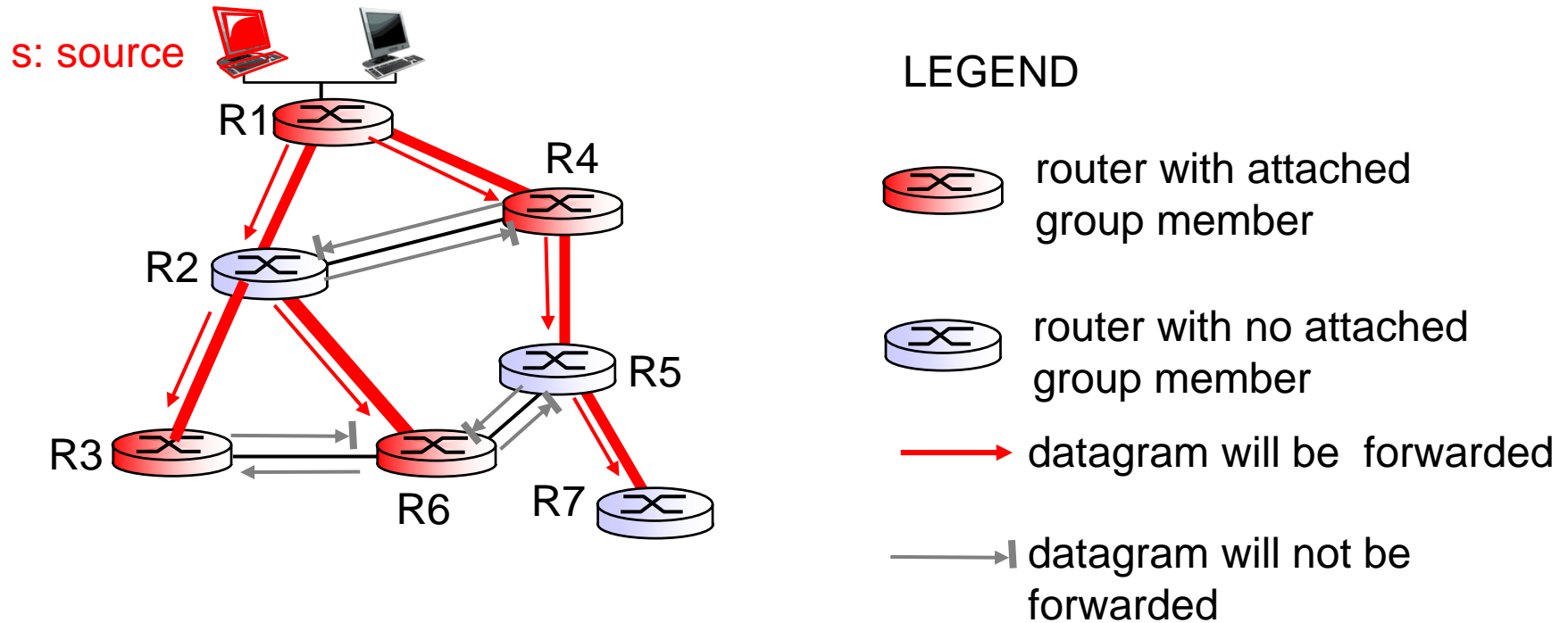


Reverse path forwarding

- ❖ rely on router's knowledge of unicast shortest path from it to sender
- ❖ each router has simple forwarding behavior:

if (mcast datagram received on incoming link on shortest path back to center)
then flood datagram onto all outgoing links
else ignore datagram

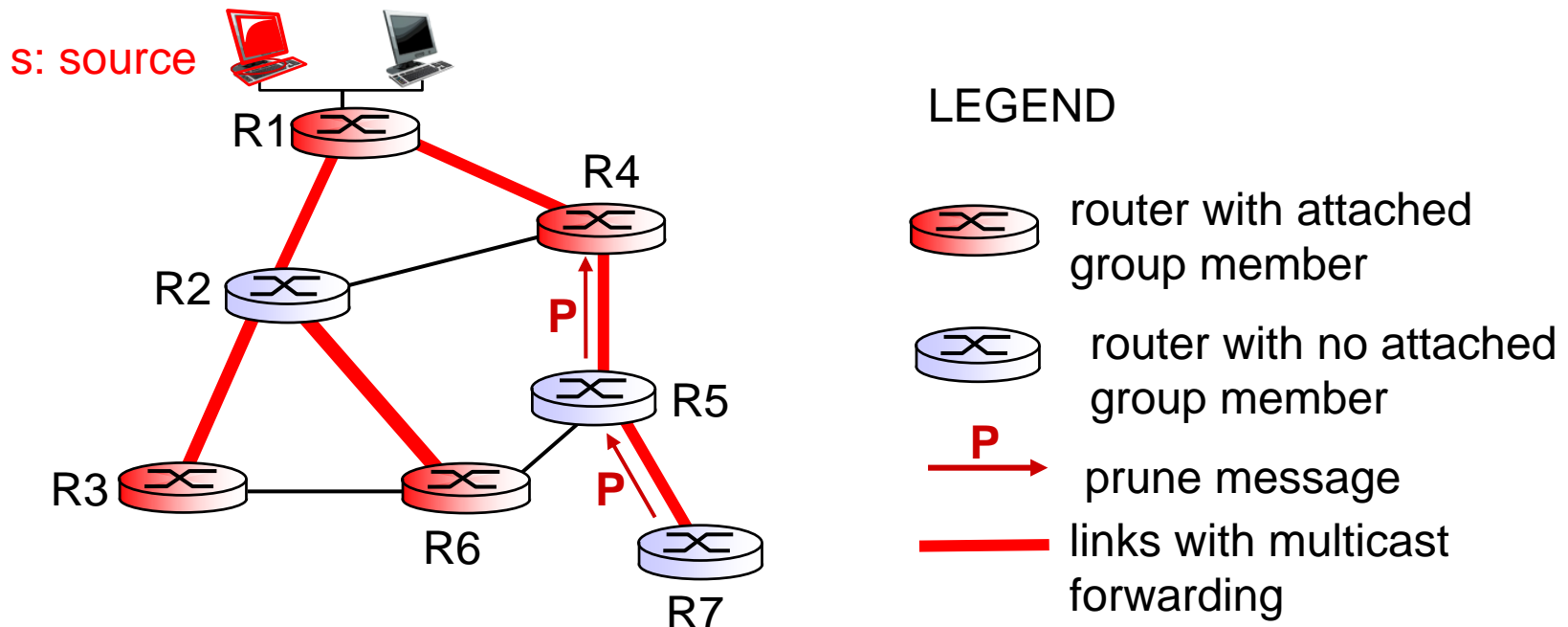
Reverse path forwarding: example



- ❖ result is a source-specific *reverse* SPT
 - may be a bad choice with asymmetric links

Reverse path forwarding: pruning

- ❖ forwarding tree contains subtrees with no mcast group members
 - no need to forward datagrams down subtree
 - “prune” msgs sent upstream by router with no downstream group members



Shared-tree: steiner tree

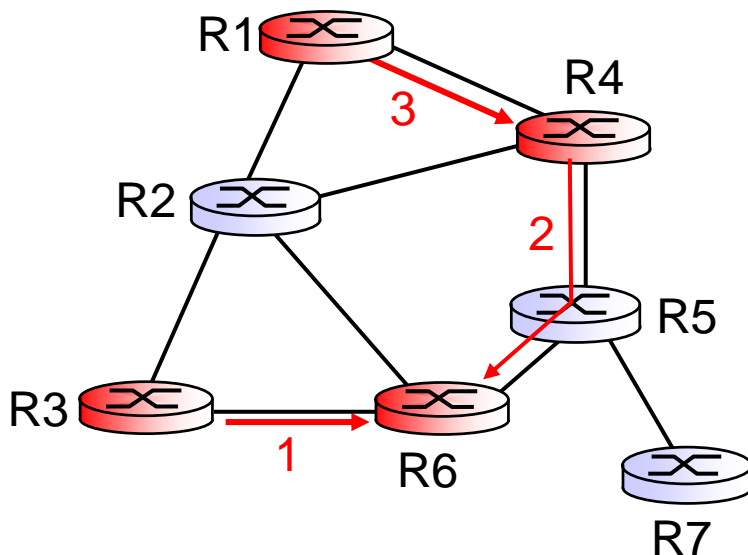
- ❖ ***steiner tree***: minimum cost tree connecting all routers with attached group members
- ❖ problem is NP-complete
- ❖ excellent heuristics exists
- ❖ not used in practice:
 - computational complexity
 - information about entire network needed
 - monolithic: rerun whenever a router needs to join/leave

Center-based trees

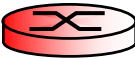


- ❖ single delivery tree shared by all
- ❖ one router identified as “*center*” of tree
- ❖ to join:
 - edge router sends unicast *join-msg* addressed to center router
 - *join-msg* “processed” by intermediate routers and forwarded towards center
 - *join-msg* either hits existing tree branch for this center, or arrives at center
 - path taken by *join-msg* becomes new branch of tree for this router

Center-based trees: example

suppose R6 chosen as center:



LEGEND

-  router with attached group member
-  router with no attached group member
-  path order in which join messages generated

Internet Multicasting Routing: DVMRP

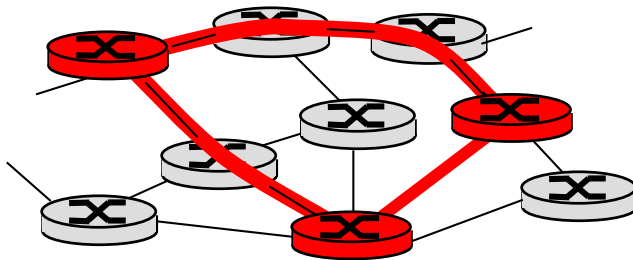
- ❖ **DVMRP**: distance vector multicast routing protocol, RFC1075
- ❖ ***flood and prune***: reverse path forwarding, source-based tree
 - RPF tree based on DVMRP's own routing tables constructed by communicating DVMRP routers
 - no assumptions about underlying unicast
 - initial datagram to mcast group flooded everywhere via RPF
 - routers not wanting group: send upstream prune msgs

DVMRP: continued...

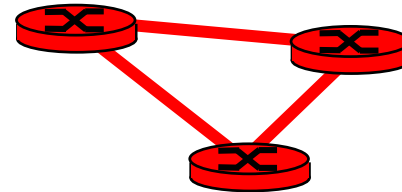
- ❖ ***soft state***: DVMRP router periodically (1 min.) “forgets” branches are pruned:
 - mcast data again flows down unpruned branch
 - downstream router: re prune or else continue to receive data
- ❖ routers can quickly regraft to tree
 - following IGMP join at leaf
- ❖ odds and ends
 - commonly implemented in commercial router

Tunneling

Q: how to connect “islands” of multicast routers in a “sea” of unicast routers?



physical topology



logical topology

- ❖ mcast datagram encapsulated inside “normal” (non-multicast-addressed) datagram
- ❖ normal IP datagram sent thru “tunnel” via regular IP unicast to receiving mcast router (recall IPv6 inside IPv4 tunneling)
- ❖ receiving mcast router unencapsulates to get mcast datagram

PIM: Protocol Independent Multicast

- ❖ not dependent on any specific underlying unicast routing algorithm (works with all)
- ❖ two different multicast distribution scenarios :

dense:

- ❖ group members densely packed, in “close” proximity.
- ❖ bandwidth more plentiful

sparse:

- ❖ # networks with group members small wrt # interconnected networks
- ❖ group members “widely dispersed”
- ❖ bandwidth not plentiful

Consequences of sparse-dense dichotomy:

dense

- ❖ group membership by routers *assumed* until routers explicitly prune
- ❖ *data-driven* construction on mcast tree (e.g., RPF)
- ❖ bandwidth and non-group-router processing *profligate*

sparse:

- ❖ no membership until routers explicitly join
- ❖ *receiver-driven* construction of mcast tree (e.g., center-based)
- ❖ bandwidth and non-group-router processing *conservative*

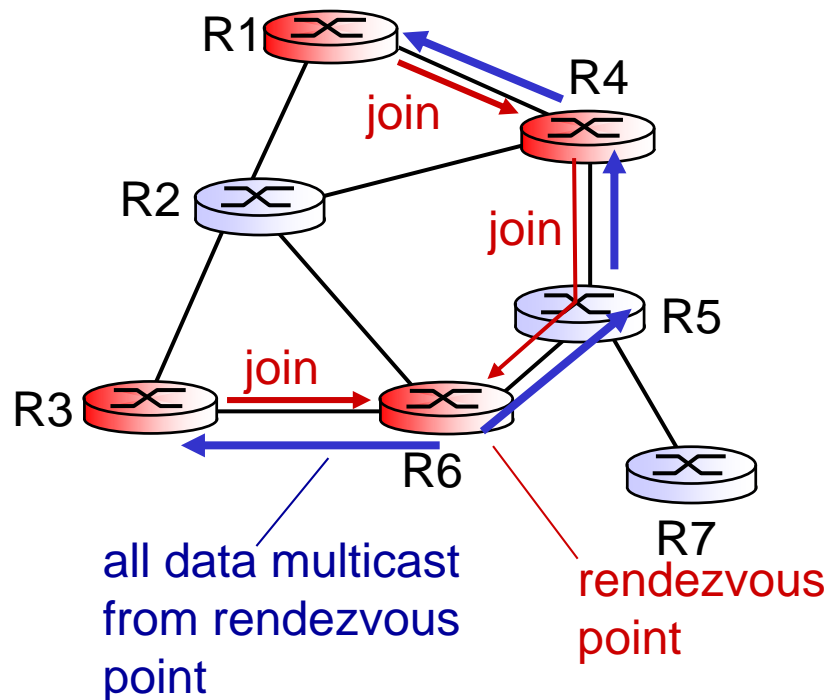
PIM- dense mode

flood-and-prune RPF: similar to DVMRP but...

- ❖ underlying unicast protocol provides RPF info for incoming datagram
- ❖ less complicated (less efficient) downstream flood than DVMRP reduces reliance on underlying routing algorithm
- ❖ has protocol mechanism for router to detect it is a leaf-node router

PIM - sparse mode

- ❖ center-based approach
- ❖ router sends *join* msg to rendezvous point (RP)
 - intermediate routers update state and forward *join*
- ❖ after joining via RP, router can switch to source-specific tree
 - increased performance: less concentration, shorter paths



PIM - sparse mode

sender(s):

- ❖ unicast data to RP, which distributes down RP-rooted tree
- ❖ RP can extend mcast tree upstream to source
- ❖ RP can send *stop* msg if no attached receivers
 - “no one is listening!”

