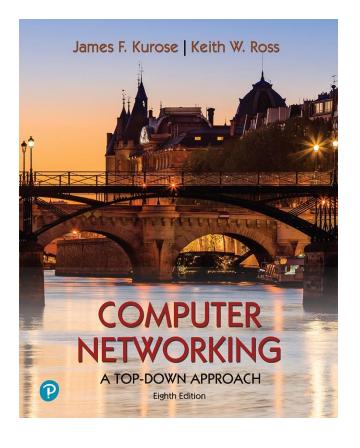


Computer Networks

Amir Mahdi Sadeghzadeh, Ph.D.

Chapter 4 Network Layer: Data Plane

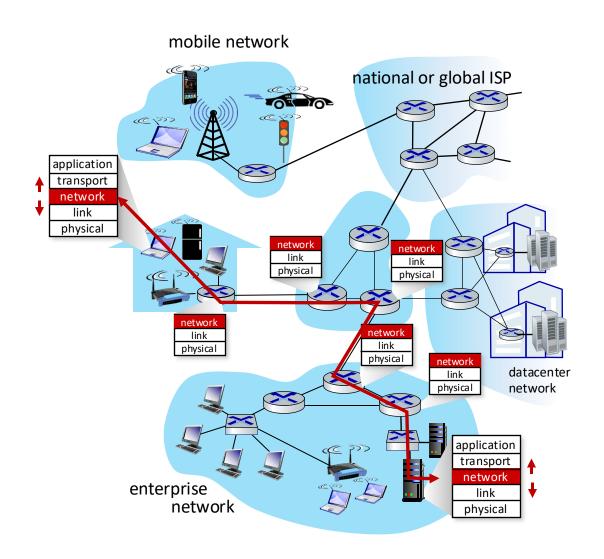


Computer Networking: A Top-Down Approach

8th edition Jim Kurose, Keith Ross Pearson, 2020

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



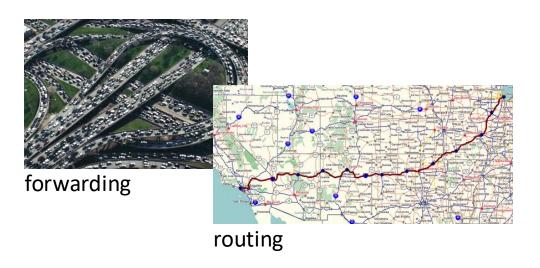
Two key network-layer functions

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" 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

IP Datagram format

layer overhead for

TCP+IP

32 bits IP protocol version number head. type of ver length header length(bytes) service len fragment "type" of service: 16-bit identifier | flgs offset diffserv (0:5) time to upper header • ECN (6:7) live layer checksum TTL: remaining max hops source IP address (decremented at each router) destination IP address upper layer protocol (e.g., TCP or UDP) options (if any) overhead 20 bytes of TCP payload data 20 bytes of IP (variable length, = 40 bytes + app typically a TCP

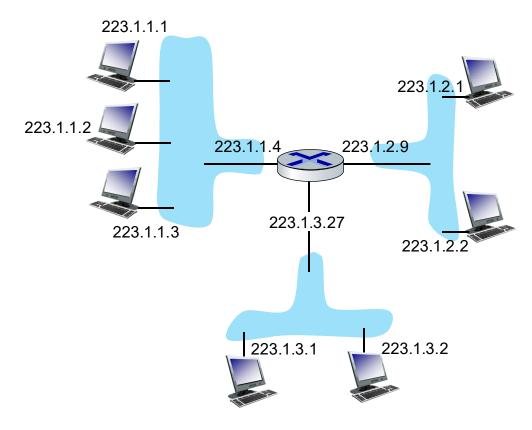
or UDP segment)

total datagram length (bytes) fragmentation/ reassembly header checksum 32-bit source IP address Maximum length: 64K bytes Typically: 1500 bytes or less e.g., timestamp, record route taken

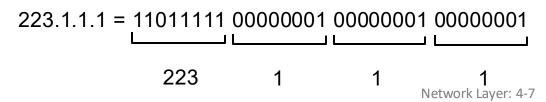
Network Layer: 4-6

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: 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

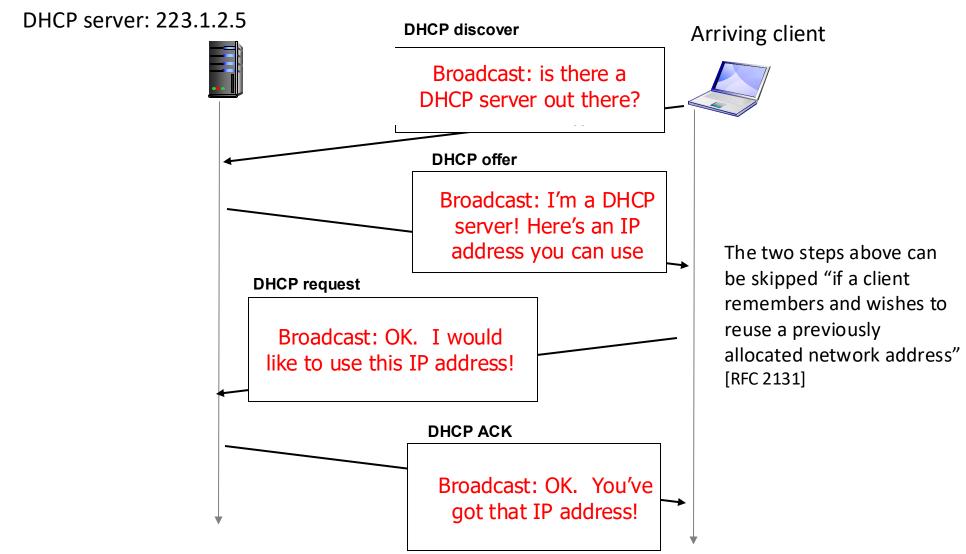
goal: host dynamically obtains IP address from network server when it "joins" network

- 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: 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 provider ISP's address space

ISP's block 11001000 00010111 00010000 00000000 200.23.16.0/20

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 00010000 00000000
        200.23.16.0/23

        Organization 1
        11001000 00010111 00010010 00000000
        200.23.16.0/23

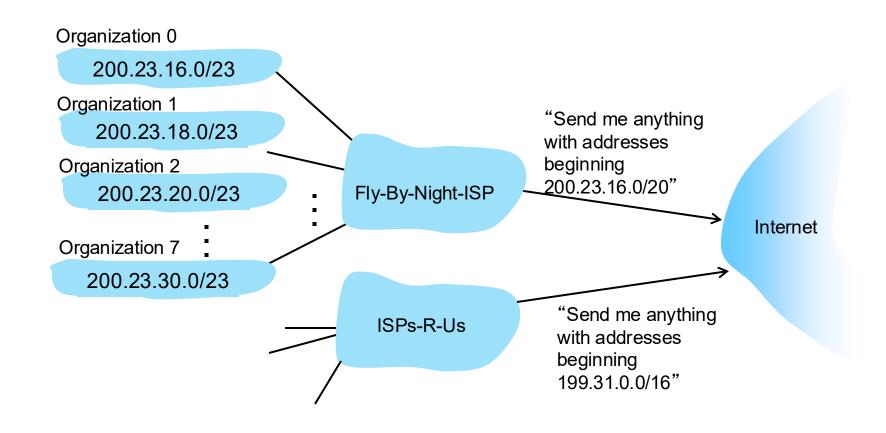
        Organization 2
        11001000 00010111 00010100 00000000
        200.23.20.0/23

        ...
        ...
        ...

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

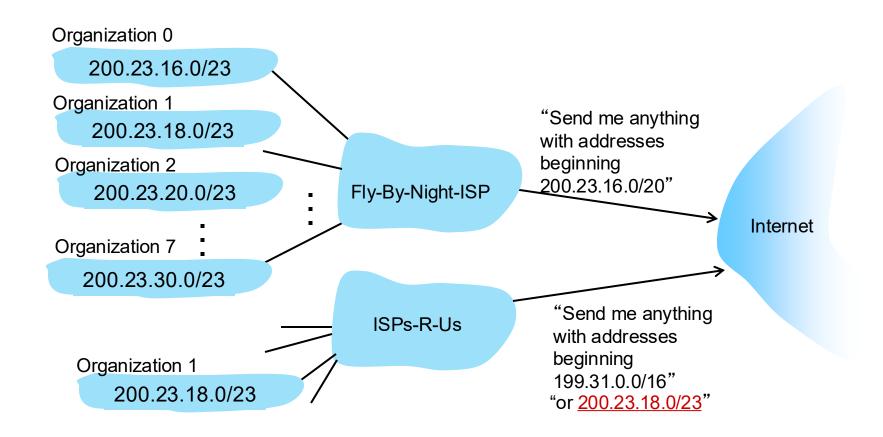
Hierarchical addressing: route aggregation

hierarchical addressing allows efficient advertisement of routing information:



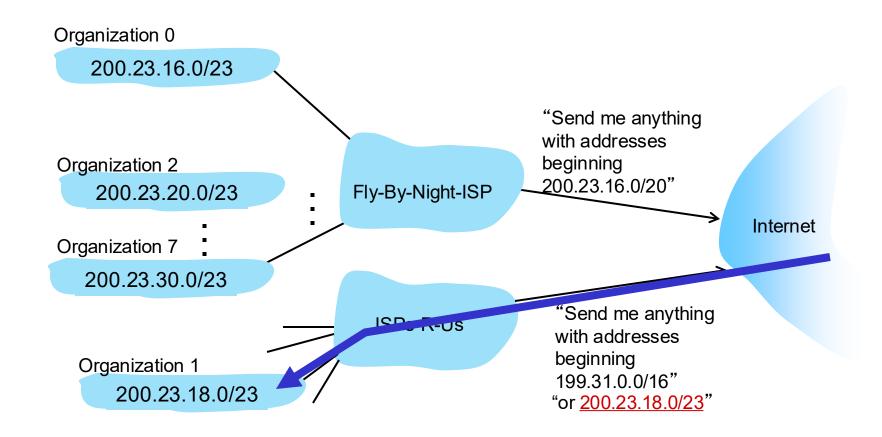
Hierarchical addressing: more specific routes

- 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)

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)

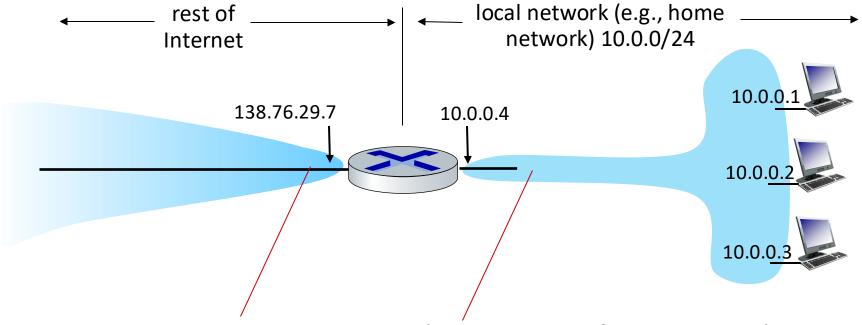
Network layer: "data plane" roadmap

- 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



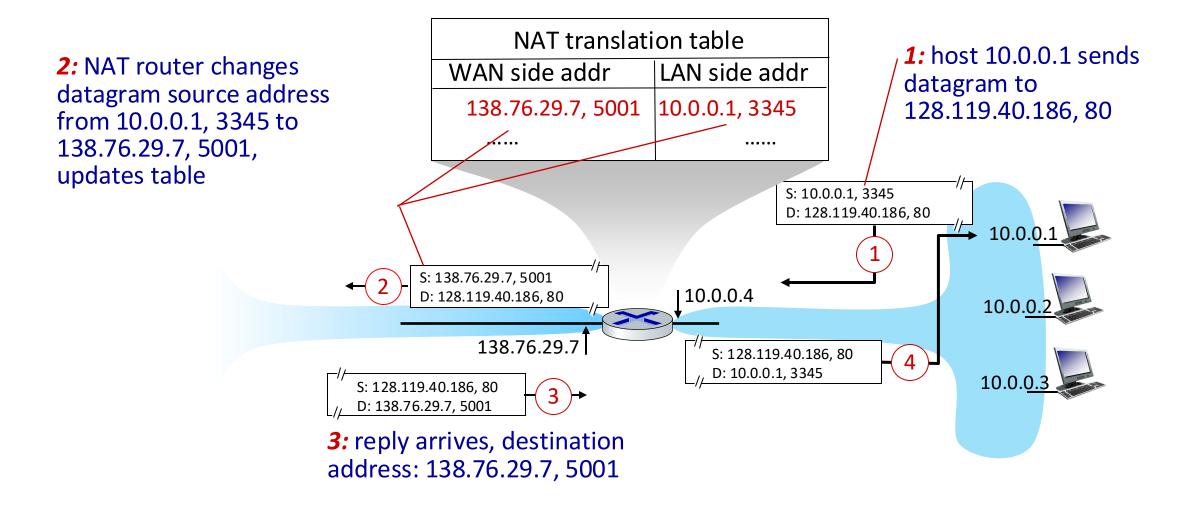
all datagrams leaving local network have same source NAT IP address: 138.76.29.7, but different source port numbers

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):

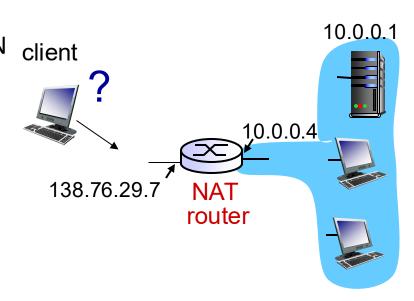
- outgoing datagrams: replace (source IP address, port #) of every outgoing datagram to (NAT IP address, new port #)
 - remote clients/servers will respond using (NAT IP address, new port
 #) as destination address
- remember (in NAT translation table) every (source IP address, port #)
 to (NAT IP address,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

NAT traversal problem

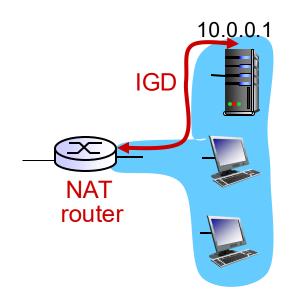
- client wants to connect to server with address 10.0.0.1
 - server address 10.0.0.1 local to LAN (client can't use it as destination addr)
 - only one externally visible NATed address: 138.76.29.7
- solution1: statically configure NAT to forward incoming connection requests at given port to server
 - e.g., (123.76.29.7, port 2500) always forwarded to 10.0.0.1 port 25000



NAT traversal problem

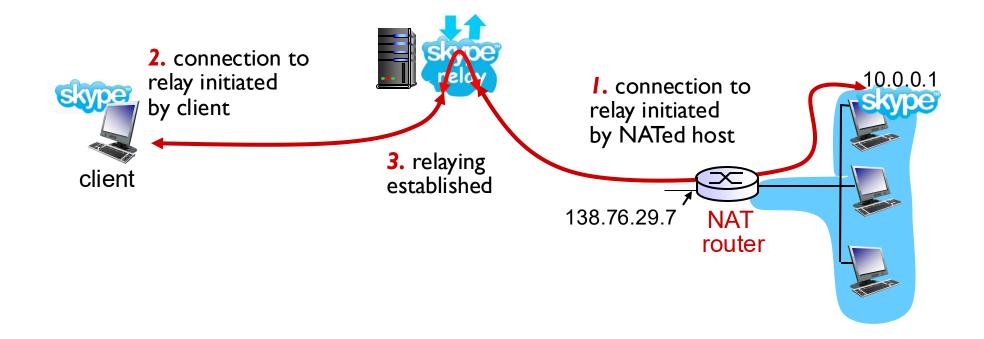
- solution 2: Universal Plug and Play (UPnP) Internet Gateway Device (IGD) Protocol. Allows NATed host to:
 - learn public IP address (138.76.29.7)
 - add/remove port mappings (with lease times)

i.e., automate static NAT port map configuration



NAT traversal problem

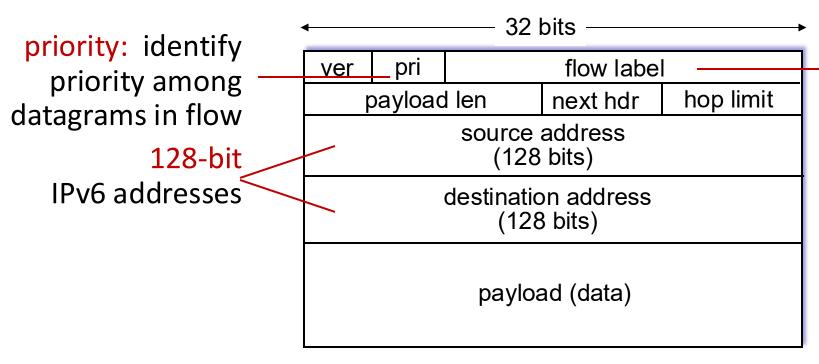
- solution 3: relaying (used in Skype)
 - NATed client establishes connection to relay
 - external client connects to relay
 - relay bridges packets between to connections



IPv6: motivation

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

IPv6 datagram format



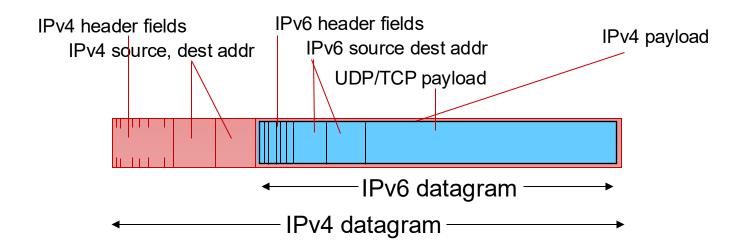
flow label: identify datagrams in same "flow." (concept of "flow" not well defined).

What's missing (compared with IPv4):

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

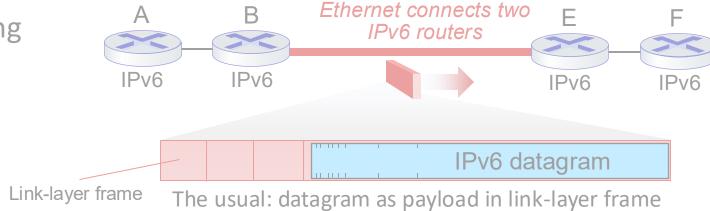
Transition from IPv4 to IPv6

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

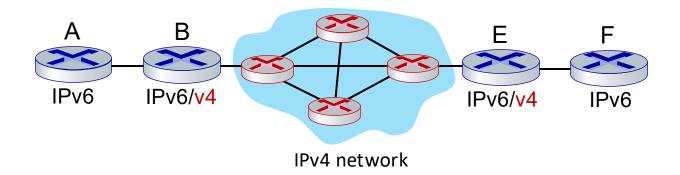


Tunneling and encapsulation

Ethernet connecting two IPv6 routers:

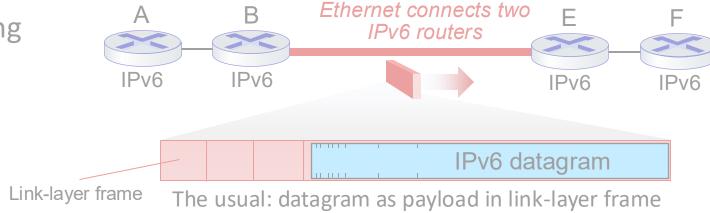


IPv4 network connecting two IPv6 routers

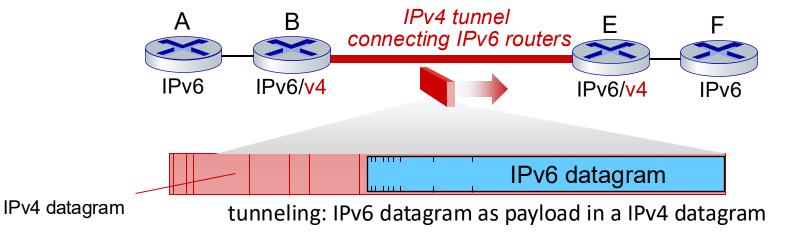


Tunneling and encapsulation

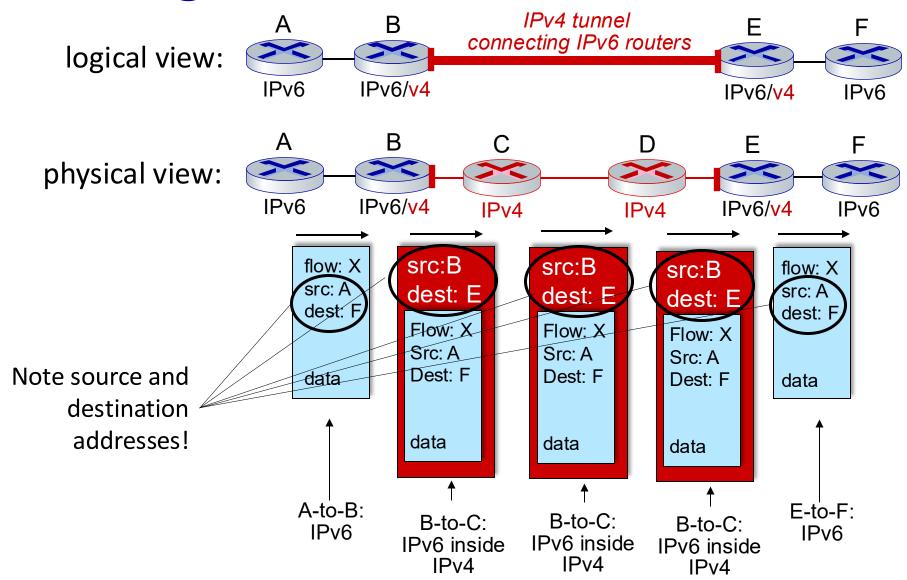
Ethernet connecting two IPv6 routers:



IPv4 tunnel connecting two IPv6 routers



Tunneling

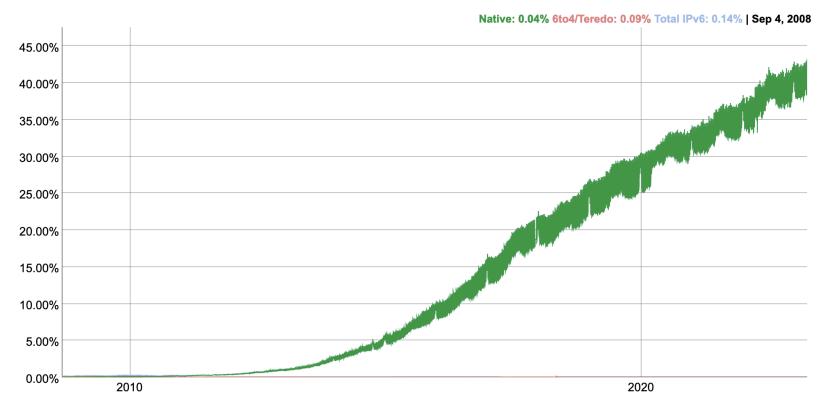


IPv6: adoption

- Google¹: ~ 40% of clients access services via IPv6 (2023)
- NIST: 1/3 of all US government domains are IPv6 capable

IPv6 Adoption

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



IPv6: adoption

- Google¹: ~ 40% of clients access services via IPv6 (2023)
- NIST: 1/3 of all US government domains are IPv6 capable
- Long (long!) time for deployment, use
 - 25 years and counting!
 - think of application-level changes in last 25 years: WWW, social media, streaming media, gaming, telepresence, ...
 - Why?

¹ https://www.google.com/intl/en/ipv6/statistics.html