

IPv6 header format:

version 4	traffic class 8	flow label 20		
payload length 16		next header 8	hop limit 8	
source address 128				
destination address 128				

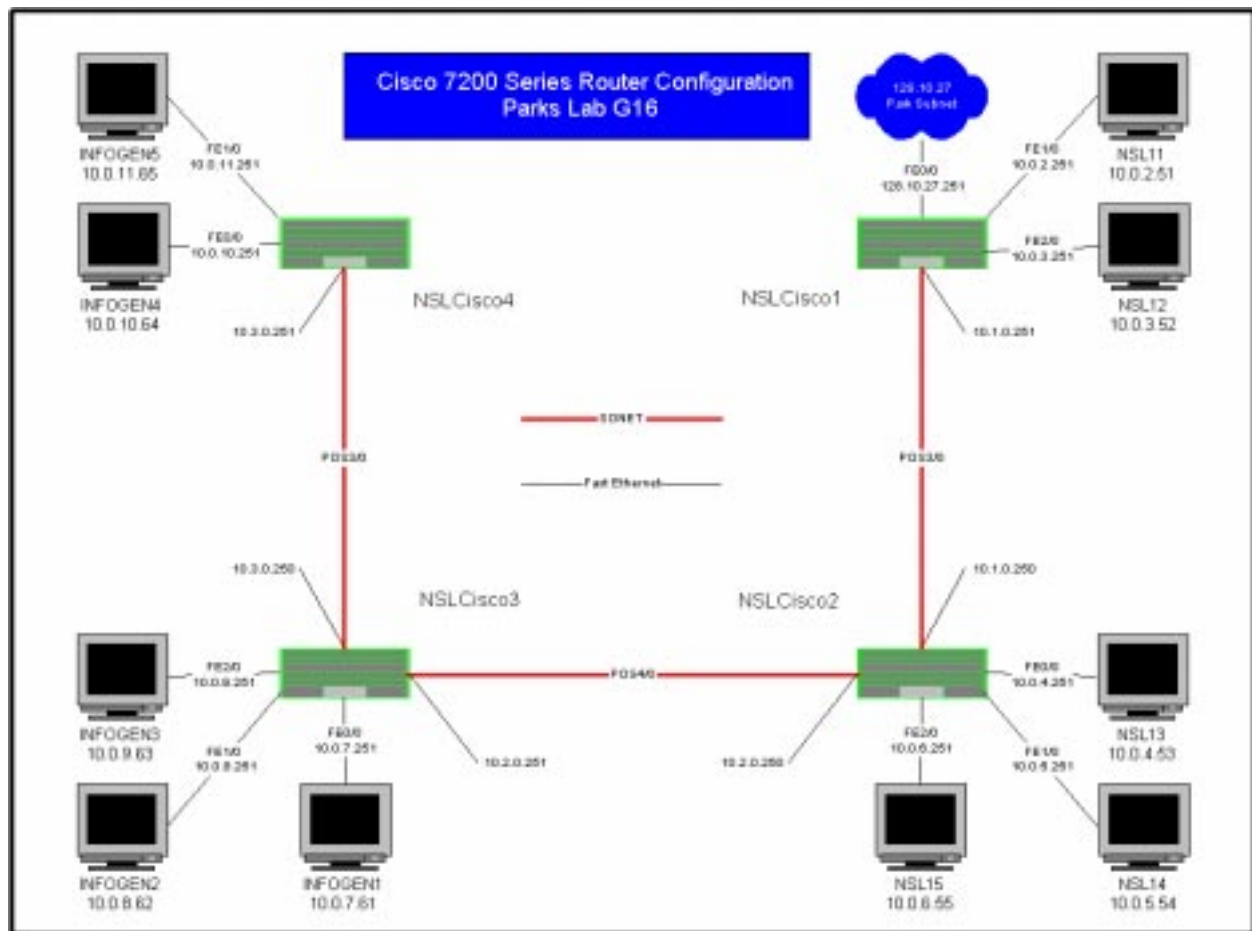
- traffic class: similar role as TOS field in IPv4
- flow label: flow label + source address
  - per-flow traffic management
  - significant extra bits
- next header: similar to IPv4 protocol field
  - plus double duty for option headers
- note missing fields

- Dynamically assigned IP addresses
  - share an IP address pool
  - reusable
  - e.g., DHCP (dynamic host configuration protocol)
  - UDP-based client/server protocol (ports 67/68)
  - note: process/daemon based service
  - used by access ISPs, enterprises, etc.
  - where are the IP address savings?

Old days of non-permanent dial-up modems . . .

- Network address translation (NAT)
  - dynamically assigned + address translation
  - private vs. public IP address
  - private: Internet routers discard them
  - e.g., 192.168.0.0 is private
  - 10.x.x.x are also private
  - useful for home networks, small businesses
  - also industry and university research labs

Example: network testbed intranet (HAAS G50)



- intranet NICs have 10.0.0.0/24 addresses
  - each interface: a separate subnet
- only one of the routers connected to Internet

- NAPT (NAT + port)

→ variant of NAT: borrow src port field as address bits

Ex.: 192.168.10.10 and 192.168.10.11 both map to 128.10.27.10

but

→ 192.168.10.10 maps to 128.10.26.10:6001

→ 192.168.10.11 maps to 128.10.26.10:6002

What about port numbers of 192.168.10.10 and 192.168.10.11?

→ e.g., client process bound to 192.168.10.10:22222

→ e.g., client process bound to 192.168.10.11:33333

Doesn't matter: NAPT translation table entries

→ 192.168.10.10:22222 maps to 128.10.26.10:6001

→ 192.168.10.11:33333 maps to 128.10.26.10:6002

For example:

if 192.168.10.10:22222 is a web browser (say Firefox) downloading web page from www.purdue.edu:80

→ web server knows client as 128.10.27.10:6001

→ no ambiguity or confusion

→ similarly for 192.168.10.11:33333

NAPT yields huge increase in effective IP address space

→ IP address bits are increased to 48 ( $= 32 + 16$ )

→ biggest factor preventing IP address depletion

Technical problems with NAPT?

Difficult to run servers behind DHCP intranet:

- how to discover server's dynamic IP address?
- how to discover server's dynamic port number?
- NAT traversal problem

Old solution: pay more to ISP to get fixed public IP address and port number

- not a good customer solution
- lots of P2P apps, VoIP, gaming, etc.

Two methods:

1. Proxies/relays

- e.g., Skype: clients contact well-known server—server knows their dynamic addresses
- server informs client its peer's dynamic IP address and port number
- peers can talk to each directly
- also called UDP/TCP hole punching

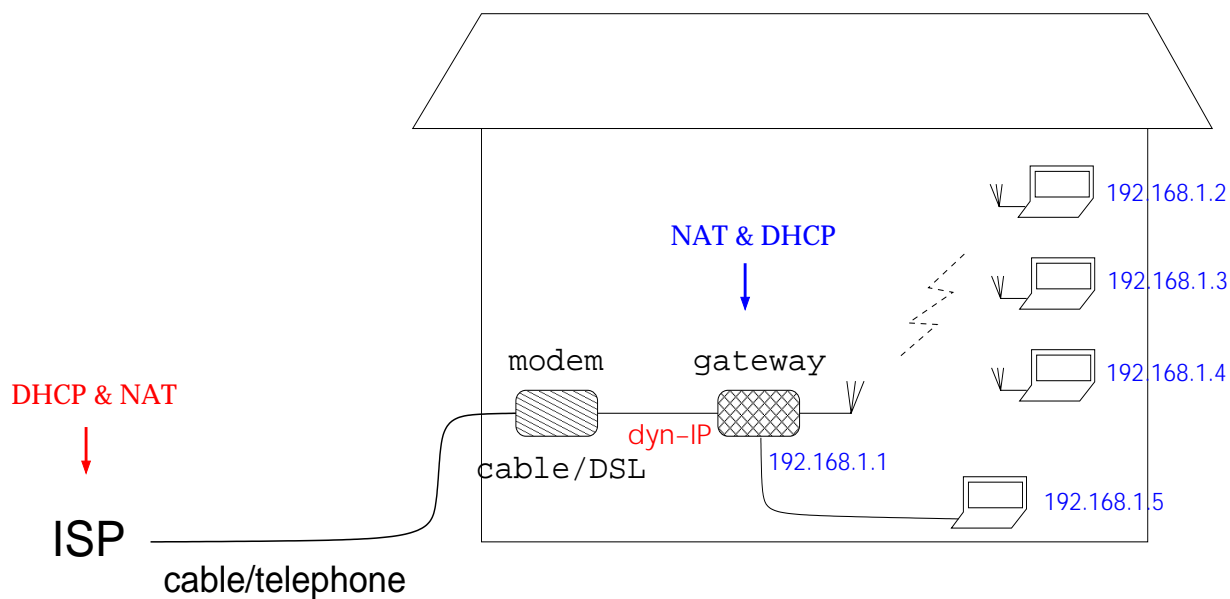
2. Enhanced gateway capabilities

- e.g., IGD (Internet Gateway Device) in UPnP
- IGD compliant router allows user to specify desired port number
- not much help with dynamic IP address
- user communicates desired port number via UPnP protocol



Ex.: SOHO (small office/home office)

→ now: home networking



- dynamic IP address provided by ISP is shared through NAT
- recall: private IP addresses

→ 10.0.0.0/8, 172.16.0.0–172.31.255.255, 192.168.0.0/16

## DHCP: 2-phase protocol

### 1. Discovery

- client sends broadcast discovery message (UDP, client port 68, server port 67) on LAN
- one or more DHCP servers respond with dynamic IP address

### 2. Allocation

- client sends broadcast message requesting selected IP address
- DHCP server confirms assignment

DHCP does other network configuration chores:

- provides DNS server names
- first-hop router/gateway
- subnet mask

CIDR and dynamically assigned IP addresses with NAT

→ significant increase of Internet's effective address space

→ saved the day

But last address block allocated by IANA (suborganization of ICANN) to regional registries early 2011

→ RIRs: ARIN, RIPE, APNIC, LACNIC, AFRINIC

→ back to address space crunch?

→ another push for IPv6

→ ISPs and companies reluctant

→ a number of technical and performance issues

→ 40-byte header

→ not backward compatible with IPv4

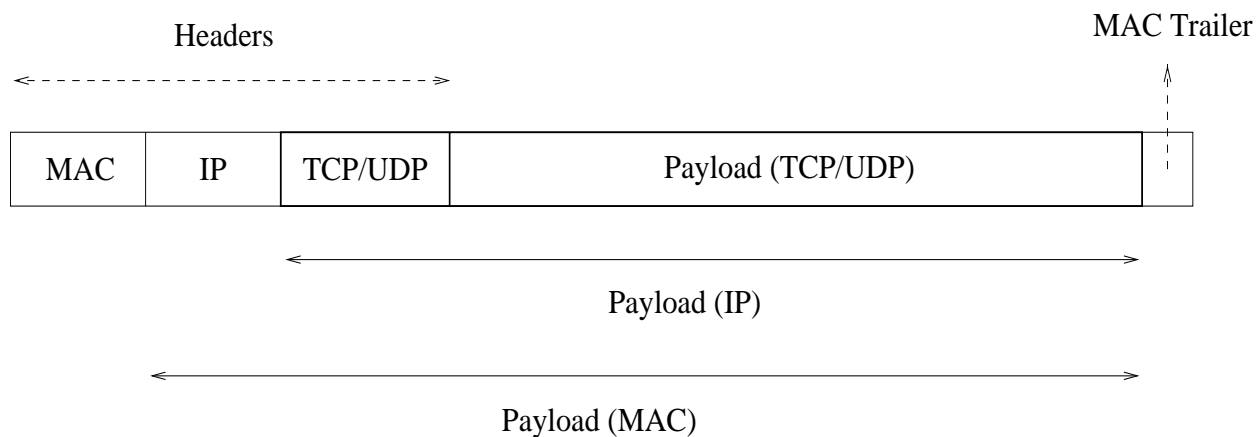
→ must use separate compatibility mechanisms (e.g., tunneling, hybrid sockets)

→ not-so-pleasant history/memories

## Transport Protocols: TCP and UDP

- end-to-end protocol
- runs on top of network layer protocols
- treat network layer & below as black box

Three-level encapsulation:



- meaning of protocol “stack”: push/pop headers
- common TCP payload: HTTP

Network layer (IP) assumptions:

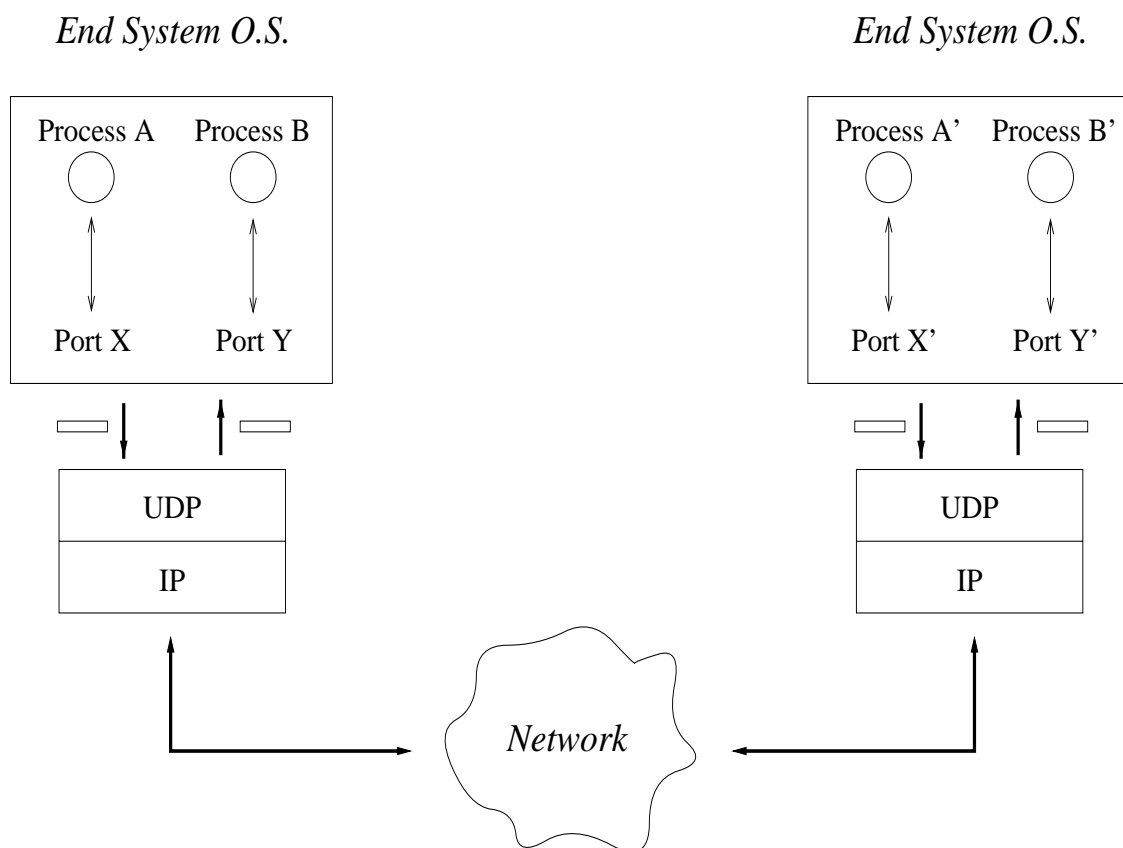
- unreliable
- out-of-order delivery
- absence of QoS guarantees (delay, throughput, etc.)
- insecure (IPv4)
  - IPsec (native in IPv6)

Additional performance properties:

- works “ok”
- can break down under high load conditions
  - flash crowds
  - DoS and worm attack
- wide behavioral range
  - sometimes good, so so, or bad
  - multitude of causes (e.g., end systems)

Goal of UDP (User Datagram Protocol):

- process identification
- port number as demux key
- minimal support beyond IP



UDP packet format:

2		2	
Source Port		Destination Port	
Length		Checksum	
Payload			

Checksum calculation (pseudo header):

4		
Source Address		
Destination Address		
00 ... 0	Protocol	UDP Length

→ pseudo header, UDP header and payload

UDP usage:

- multimedia streamining
  - lean and nimble
  - at minimum requires process identification
  - congestion control carried out above UDP
- stateless client/server applications
  - persistent state can be a hinderance
  - lightweight



Goals of TCP (Transmission Control Protocol):

- process identification
  - reliable communication: ARQ
  - speedy communication: congestion control
  - segmentation
- 
- connection-oriented, i.e., stateful
  - complex mixture of functionalities

Segmentation task: provide “stream” interface to higher level protocols

—→ exported semantics: contiguous byte stream

—→ recall ARQ

- segment stream of bytes into blocks of fixed size
- segment size determined by TCP MTU (Maximum Transmission Unit)
- actual unit of transmission in ARQ

TCP packet format:

2

2

Source Port				Destination Port				
Sequence Number								
Acknowledgement Number								
Header Length	<div><div></div><div></div><div></div><div></div><div></div><div></div></div>	URG	ACK	PSH	RST	SYN	FIN	Window Size
Checksum				Urgent Pointer				
Options (if any)								
DATA (if any)								

- Sequence Number: position of first byte of payload
- Acknowledgement: next byte of data expected (receiver)
- Header Length (4 bits): 4 B units
- URG: urgent pointer flag
- ACK: ACK packet flag
- PSH: override TCP buffering
- RST: reset connection
- SYN: establish connection
- FIN: close connection
- Window Size: receiver's advertised window size
- Checksum: prepend pseudo-header
- Urgent Pointer: byte offset in current payload where urgent data begins
- Options: MTU; take min of sender & receiver (default 556 B)

Checksum calculation (pseudo header):

4

Source Address		
Destination Address		
00 ... 0	Protocol	TCP Segment Length

→ pseudo header, TCP header and payload