

CS 524 A

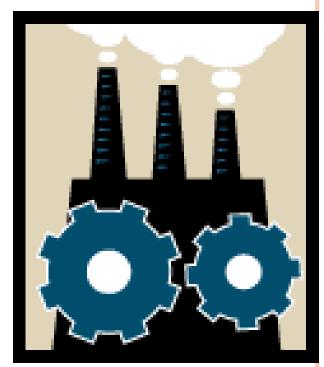
Introduction to Cloud Computing

Module 6: Data Networking and Distributed Computation (Part 2)

How the Cloud *pipes* are made

OUTLINE

- The business side: New Cloud service Use Cases and examples
- A bit more detail on the Internet network layer
 - Subnets (a different meaning) and masks
 - Autonomous Systems, Peering, and Border Gateway Protocol (BGP)
 - Private and Public IP address spaces
 - Network Address Translators (NATs)
 - IPv6
- QoS
 - Packet Scheduling Disciplines
 - Integrated Services
 - Differentiated Services
 - Multi-Protocol Label Switching (MPLS)
 - Generalized MPLS
 - Virtual Private Networks



CLOUD SERVICE EXAMPLES:

- Amazon Virtual Private Cloud (http://aws.amazon.com/vpc/)
- AWS Direct Connect (http://aws.amazon.com/directconnect/)
- Amazon S3 service transfer (http://aws.amazon.com/s3/#importexport)
- AT&T VPN Services https://www.business.att.com/products/vpn.html
- You will find other examples as part of the homework

Question: What does *Netflix* use to stream movies?

Answer: https://aws.amazon.com/solutions/casestudies/netflix/

REMEMBER THE IP ADDRESING SCHEME (IPV4)?

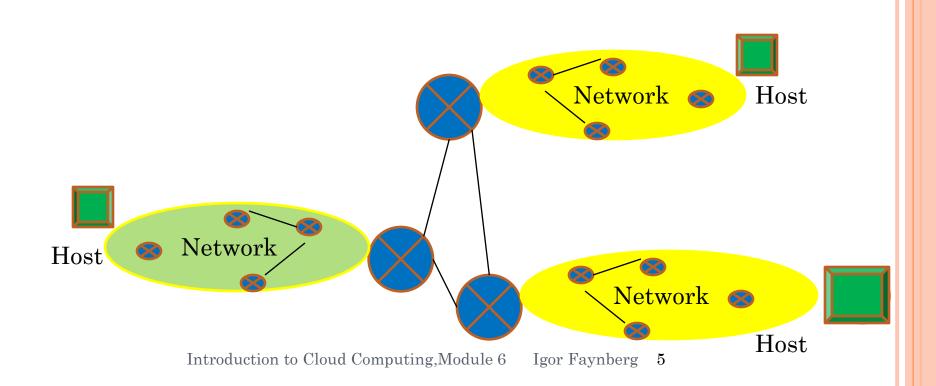
- IP addresses are assigned to all hosts and routers
- All IP addresses are 32-bit-long; they are normally written in decimal form byte-bybyte, separated by "."s (e.g., 123.100.86.35)
- Each address has a form of either
 - <class> <network> <host> (for classes A [0], B [10], and C [110]) or
 - 1110 <Multicast Address> (for class D [1110]) Or
 - 11110 <Reserved for future use>

THE (INITIAL) IP ADDRESS STRUCTURE
AND ROUTING

Pronounced

"cider"

- 1. Class (before CIDR)
- 2. Network address
- 3. Host address (within the network)



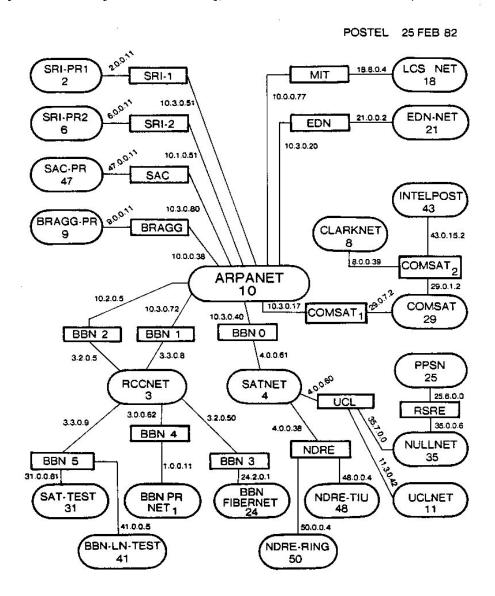
AS DEFINED IN RFC 791

- With Class A [0] (intended for huge networks)
 - Network address is 7-bit long
 - Host address is 24-bit long
- With Class B [10] (intended for average-size networks)
 - Network address is 14-bit long
 - Host address is 16-bit long
- With Class C [110] (intended for small networks)
 - Network address is 21-bit long
 - Host address is 8-bit long

The idea: The smaller the network, the more of them will be out there!

Jon Postel's map of the Internet in 1982

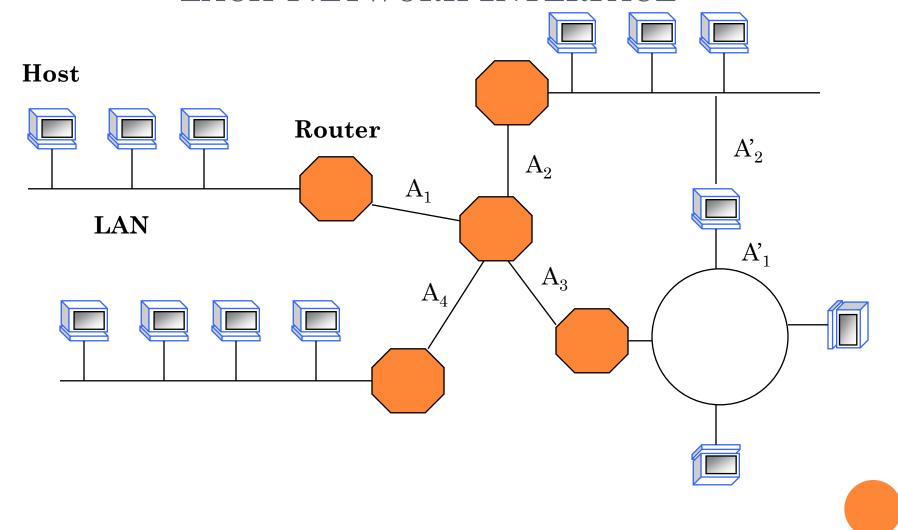
(Source: http://commons.wikimedia.org/wiki/File%3AInternet_map_in_February_82.png By Jon Postel [Public domain], via Wikimedia Commons)



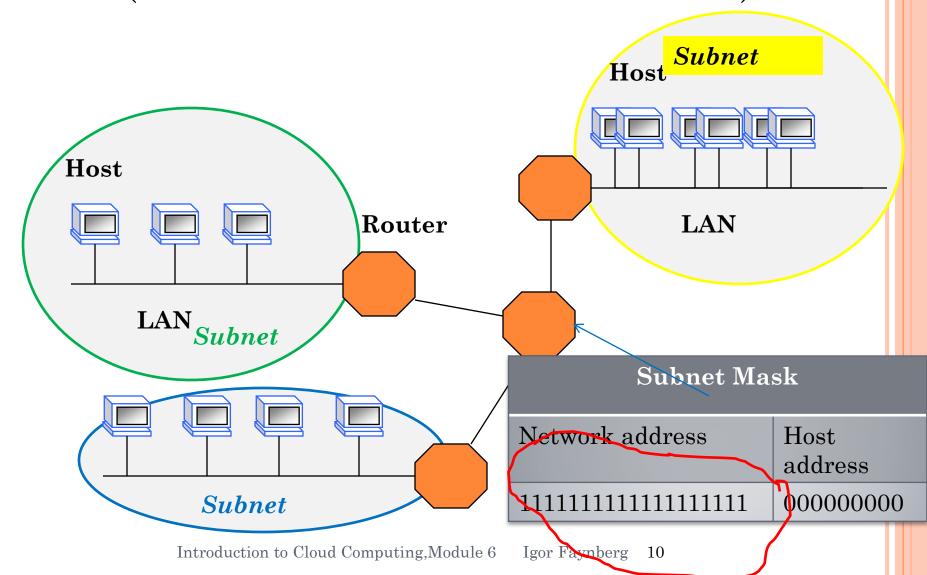
CLASS-BASED IP ADDRESS ASSIGNMENT PROBLEM

- Say, Company X has been allocated 2^{16} (65 536) host addresses for its Class B network. The network started with one LAN and only 200 hosts)
- As more departments are added, each gets its own LAN and more hosts, but only so many L2 bridges can be put in...
- And so the new LANs need to attach to routers and become *separate* networks
- Now, the company has 10,000 hosts, but it cannot do much with the remaining addresses because they are on the *same* network
- Getting new addresses is problematic (and expensive), and the old ones have not been even used!

THE RULE: A SEPARATE IP ADDRESS FOR EACH NETWORK INTERFACE



NEW (SUBNET IS DEFINED BY THE MASK)



Masks and routing tables

- Masks are shown with a "dash notation" indicating the number of bits in the network address: 124.32.17.21/16
- Now Company X can subdivide its address space into subnets
 - /17 (half of all addresses)
 - /18 (a quarter of all addresses)
 - /19 (an eighth part of addresses)
- With that, the edge router of Company X does not need to keep all hosts addresses; it keeps only prefixes for the subnet:

$Prefix = destination \ address \ AND \ mask$

• Classless Inter-Domain Routing (CIDR) system is built on the same prefixing idea: The routers aggregate and propagate an IP packet in the direction of the longest prefix (supernet)

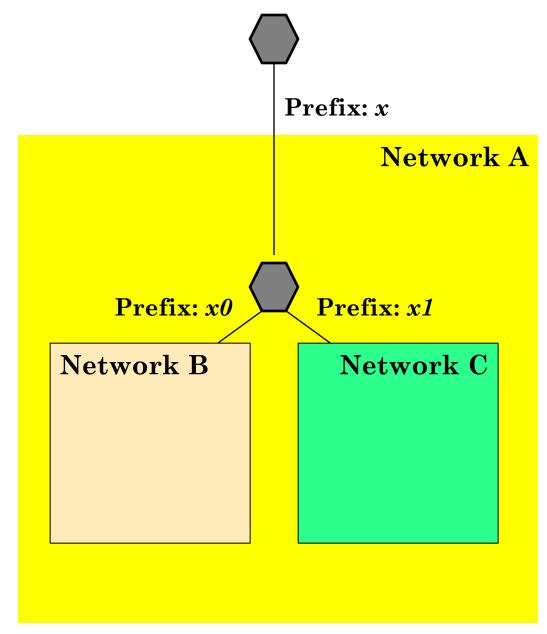
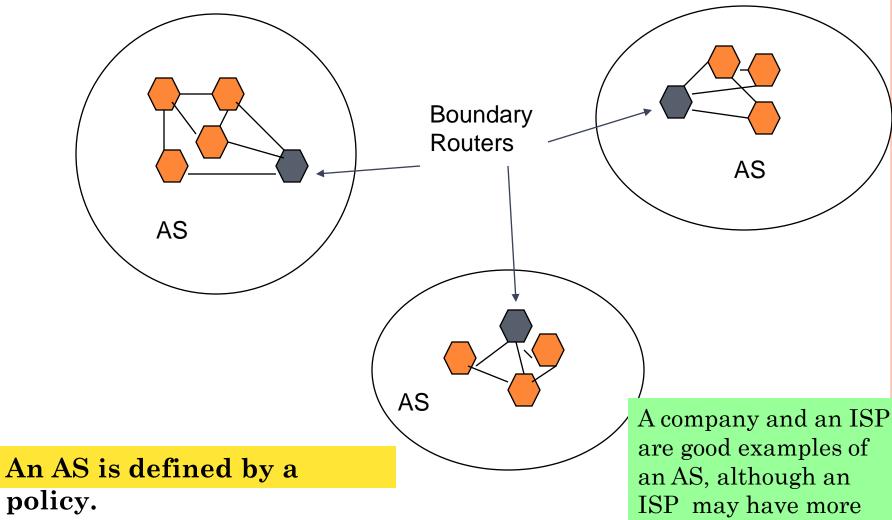


Figure 4.11: CIDR Aggregation

NEXT TOPIC: AUTONOMOUS SYSTEMS



Introduction to Cloud Computing, Module 6

Igor Faynbers

then one AS

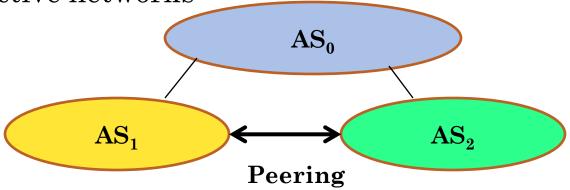
THE BORDER GATEWAY ROUTING PROTOCOL (BGP)

(Some aspects of this you will encounter in the homework, as you will study the *Amazon VPC* service)

- The current version (4) of BGP is defined in RFC 4271 (http://tools.ietf.org/html/rfc4271) and other BGP RFCs that you may need to take a look at
- BGP runs on top of TCP (why?)
- Routing constraints are expressed by *policies*. Examples:
 - Do not traverse network A when traffic originates and terminates in network B
 - When it is possible to reach a destination through either networks C or network D, always choose D
 - Don't use network X in country Y

ROUTING POLICIES ARE DEFINED ACCORDING TO BUSINESS NEEDS

- **Business**: who pays *whom* and *what* for *transit* service
- AS₁ and AS₂ may start by paying AS₀ for transit
- When AS_1 and AS_2 find that they need to exchange a lot of data, they may avoid services of AS_0 , connect their networks, and use the *peering policy* to reduce their bills
- To implement *peering*, AS₁ and AS₂ advertise to each other (via BGP) the addresses that reside in their respective networks



Multihoming and Autonomous System NUMBERS (ASNS)

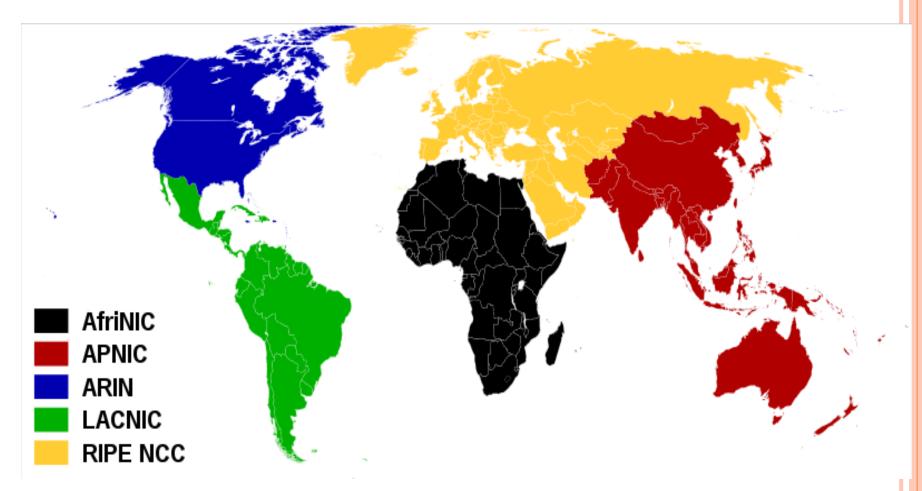
- When a network's router is connected to more than one network provider, this is called multihoming
- RFC 1930 (http://tools.ietf.org/html/rfc1930) defines multihoming as a major case for having a private autonomous system number (ASN)
- The ASN uniquely identifies each network on the Internet.
- These numbers (started with 16-bit sizes, which grew to 32) bits) are assigned in blocks by the Internet Assigned Numbers Authority (IANA) to Regional Internet Registries (RIRs)
- A unique ASN is assigned to each AS by an RIR for use in BGP routing
- An ASN fully identifies each AS on the Internet.

Present Regional Internet REGISTRIES (RIRS)

- African Network Information Centre (AfriNIC) for Africa
- American Registry for Internet Numbers (ARIN) for the United States, Canada, several parts of the Caribbean region, and Antarctica.
- Asia-Pacific Network Information Centre (APNIC) for Asia, Australia, New Zealand, and neighboring countries
- Latin America and Caribbean Network Information Centre(LACNIC) for Latin America and parts of the Caribbean region
- Réseaux IP Européens Network Coordination Centre (RIPE NCC)[for Europe, the Middle East, and Central Asia

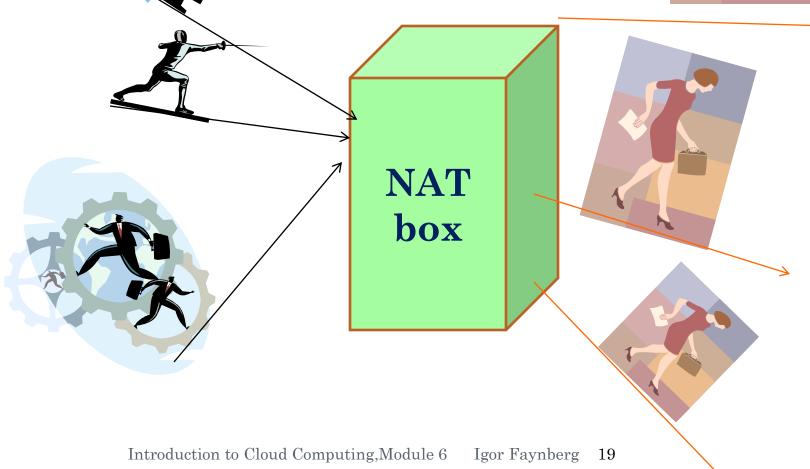
GEOGRAPHIC DISTRIBUTION OF RIRS

(SOURCE: WIKIMEDIA)

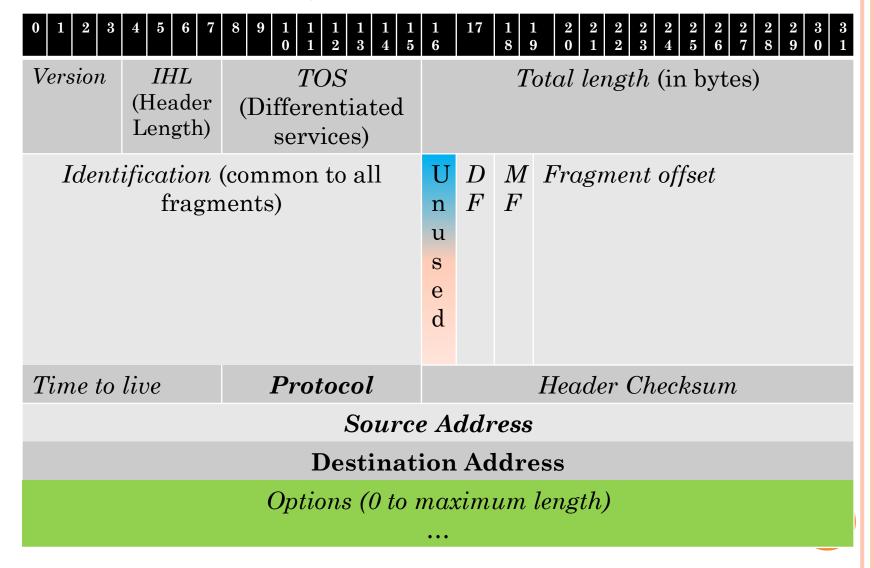


NEXT TOPIC: NETWORK ADDRESS TRANSLATORS (NATS)





IPv4 Header

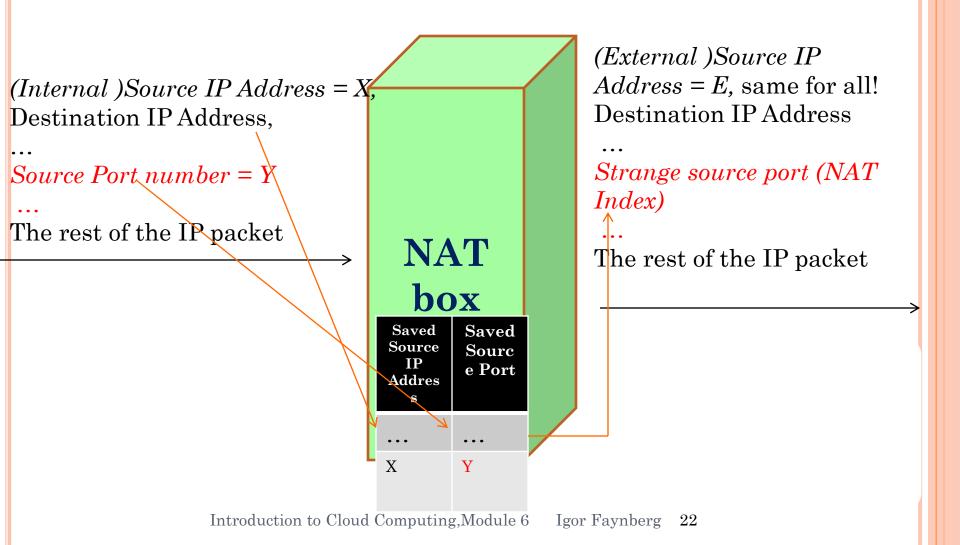


WITH A 32-BIT IP ADDRESS FIELD...

There are only 4 294 967 296 IP addresses

- Not enough for every person's computer, phone, washing machine, tooth pick, etc...
- Hence Network Address Translation (NAT) boxes which
 - are (typically) combined with firewalls
 - serve two *purposes*: 1) effective management of scarce IP addresses and 2) obfuscation of the internal network architecture

How NAT works (outgoing traffic)

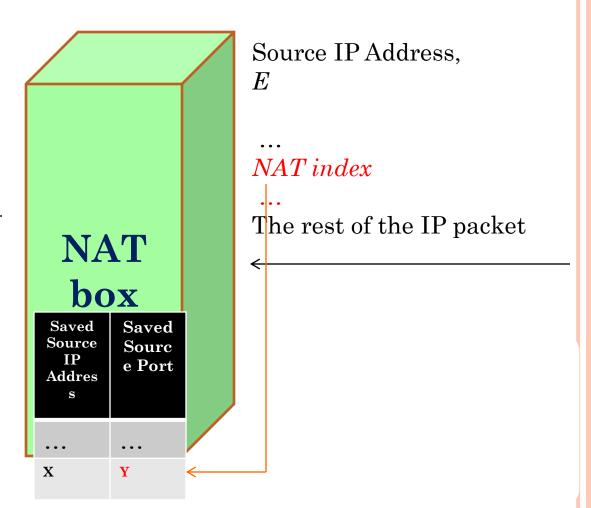


How NAT works (Incoming Traffic)

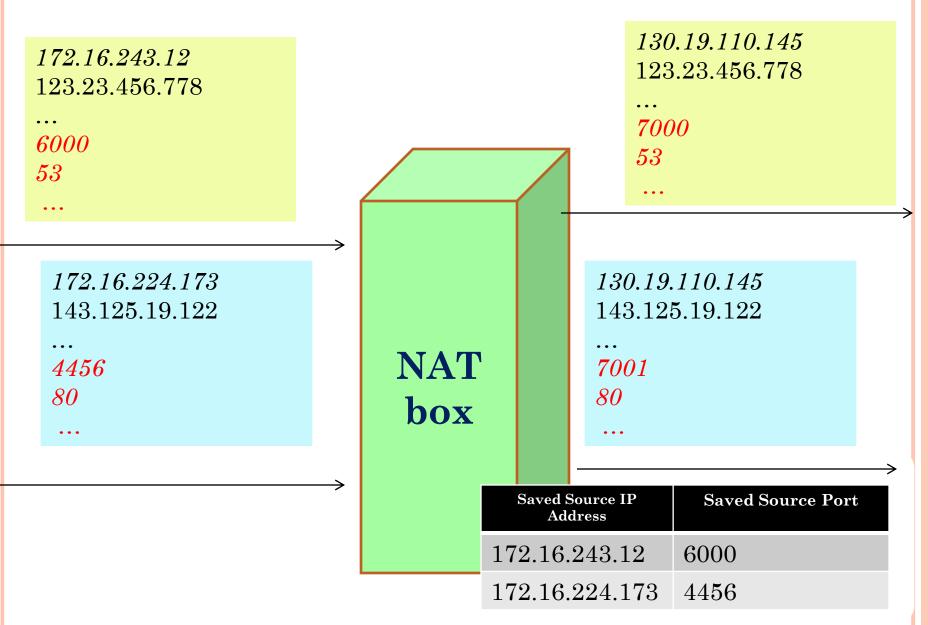
 $(Internal\) Source\ IP\ Address,$ Destination IP Address =X,

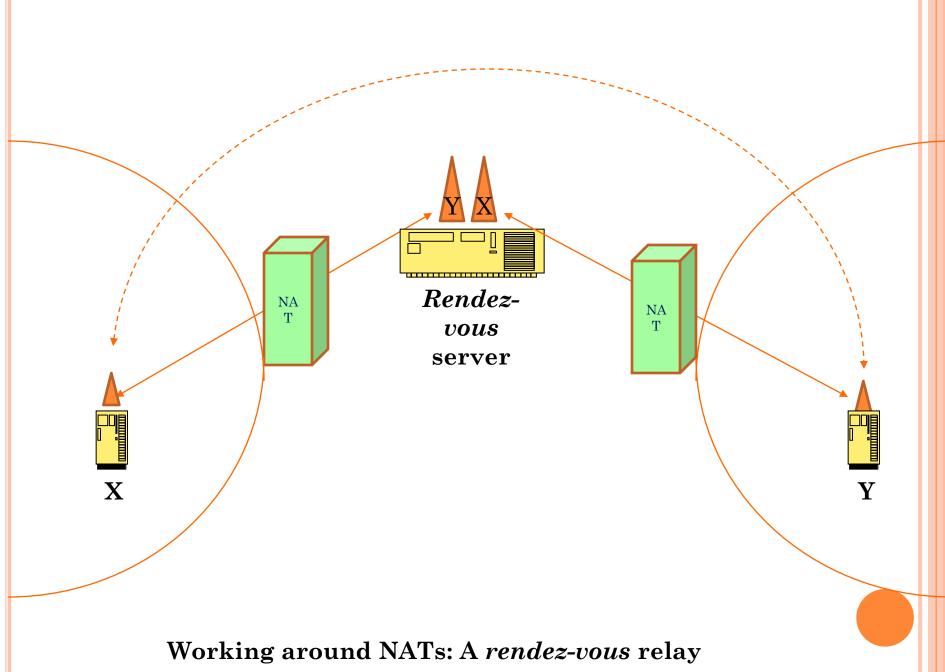
Source Port number = Y

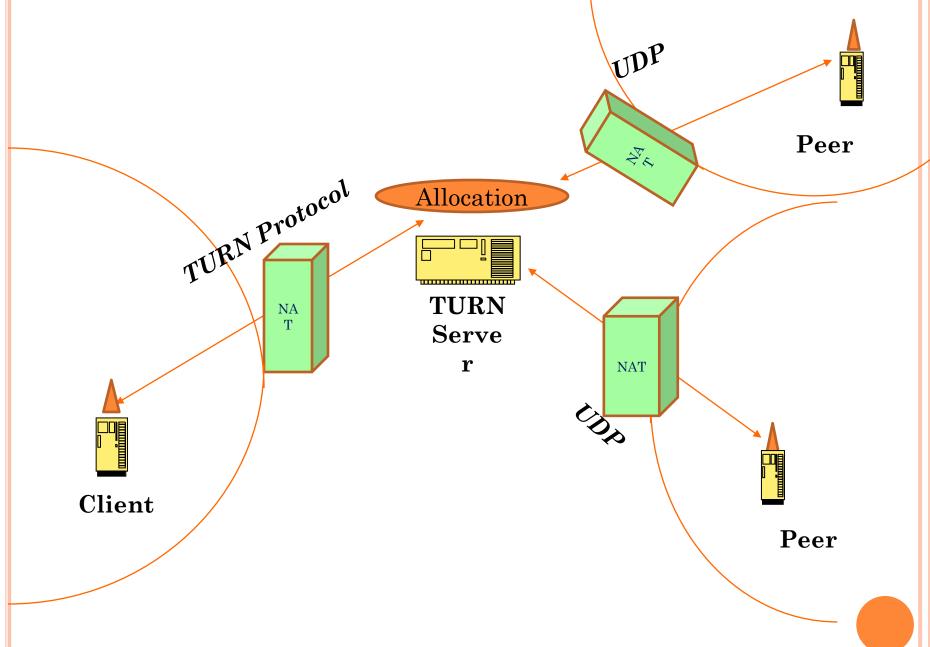
The rest of the IP packet



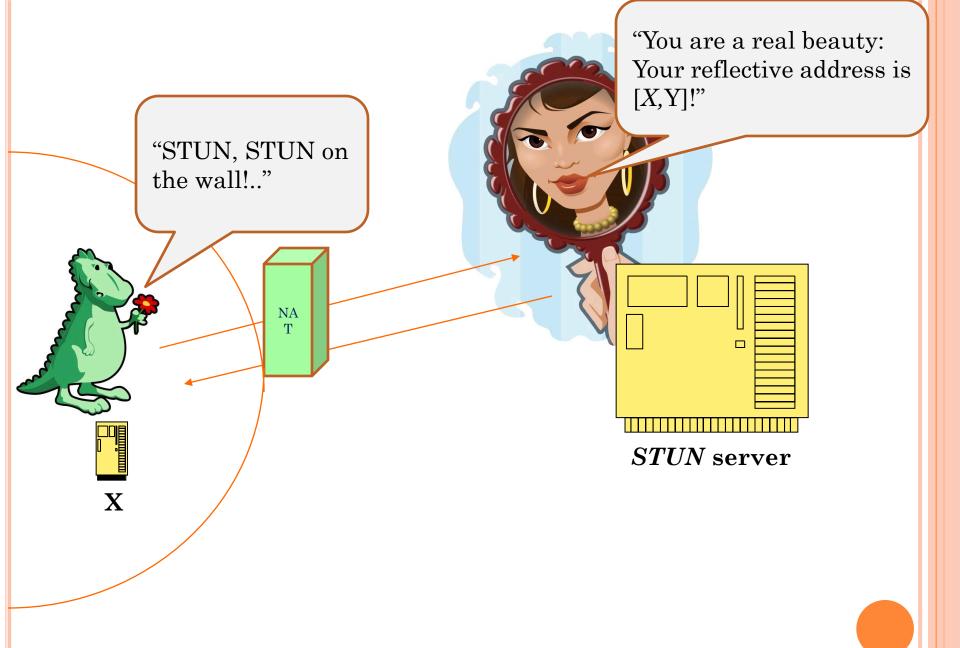
AN EXAMPLE (OUTGOING TRAFFIC)

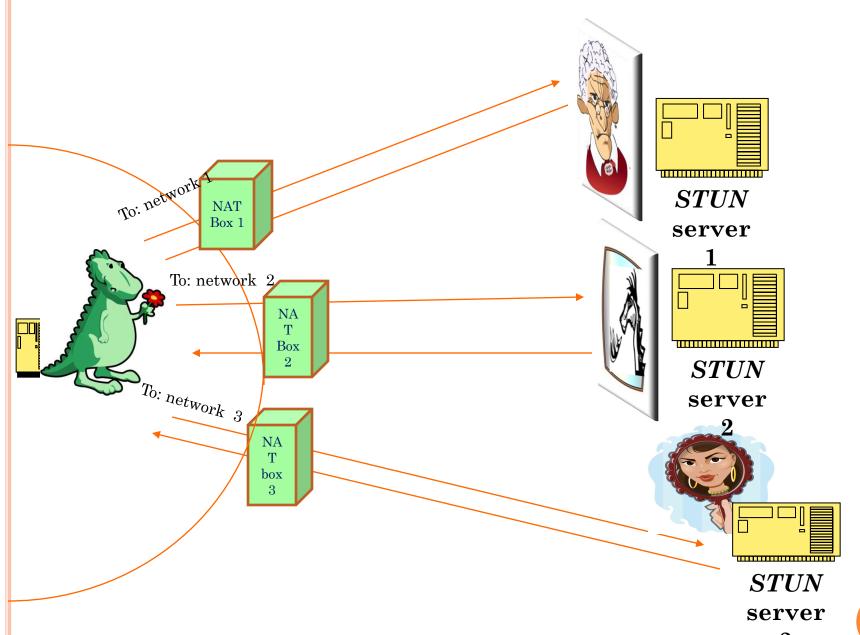






Working around NATS: Traversal Using Relays around NAT (TURN)



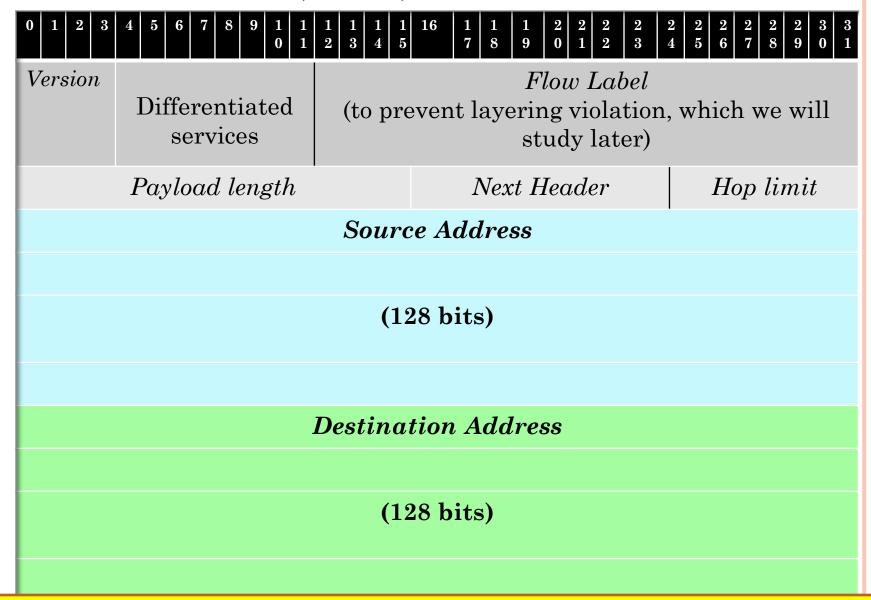


Working around NATs: Different NATs for different paths...

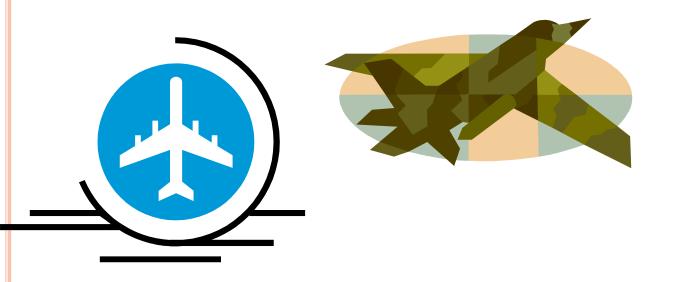
NATS

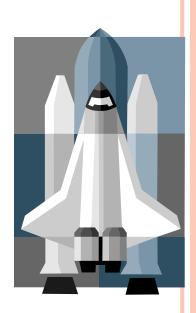
- Solve the IPv4 address shortage problem
- **Help** re-using addresses (http://tools.ietf.org/html/rfc1918)
- **Hide** internal network structure (an essential security measure)
- Violate modularity (by assuming the transport layer addressing structure and interfering with the transport header)
- Violate an Internet principle that requires a unique IP address for every host on the Internet
- Violate an Internet principle that no connection state be kept in the network
- Violate an Internet end-to-end connectivity principle: Host A behind NAT can *not* receive packets from Host B until after it had send a packet to node B
- **Interfere** with those applications protocols (such as SIP, SDP, FTP...) that carry IP addresses in payload—this gave birth to NAT traversal technology (ICE/STUN)
- Delay the IPv6 deployment

IPv6 (MAIN) HEADER



NEXT TOPIC: QUALITY OF SERVICE (QOS) IN THE NETWORK





The Internet Protocol Hourglass (after Steve Deering)

	TELNET, FTP, RPC,
Application	HTTP, SIP, RTP, RTCP, SMTP,
	Diameter, WebSocket
Transport	Diameter, WebSocket UDP, TCP
	SCTP
Network	IP
Link	PPP, HDLC, SDLC, LAN LLC, AALs
Physical	Optical, Ethernet, Wireless, twisted pair

AAL: ATM Adaptation Layer

HTTP: Hyper Text Transfer Protocol HDLC: High? Data Link Control

LLC: Logical Link Control
PPP: Point-to-Point Protocol
RPC: Remote Procedure Call

RTP: Real Time Protocol

RTCP: Real Time Control Protocol SDLC: High? Data Link Control

SMTP: Simple Mail Transfer Protocol

SCTP: Stream Control Transmission Protocol

SIP: Session Initiation Protocol

TCP: Transmission Control Protocol

UDP: User Datagram Protocol

Introduction to Cloud Computing, Module 6

QUALITY OF SERVICE (QOS) IN IP NETWORKS

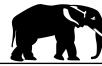
- Different applications define it differently
 - for telemedicine the accuracy of the delivery—no packets lost—is more important then *delay* or packet delay variation (*jitter*)
 - for IP telephony, jitter and delay must be minimized, but packet loss is tolerable
 - for streaming broadcast, initial delay is not a problem; jitter is controllable (how), but subsequent long delay is unacceptable
 - for e-mail, neither jitter, nor delay is problematic
 - For instant messaging, delay is a bit problematic; jitter is not
- Solutions provided by the IP routers deal with
 - recognizing the type of treatment a particular packet deserves
 - scheduling and forwarding the packet accordingly

SCHEDULING DISCIPLINES

- Best Effort: packets are sent on the first-come, firstserved basis; they are dropped when queues become too large
- Fair Queuing: packets are queued for each flow and transmitted round-robin to guarantee each flow an equal share of bandwidth; bigger things wait for smaller, faster things
- Weighted Fair Queuing: same as fair queuing, but bandwidth is allocated according to priority

SCHEDULING DISCIPLINES: AN ILLUSTRATION





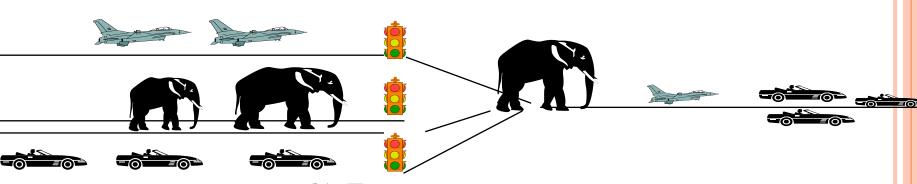




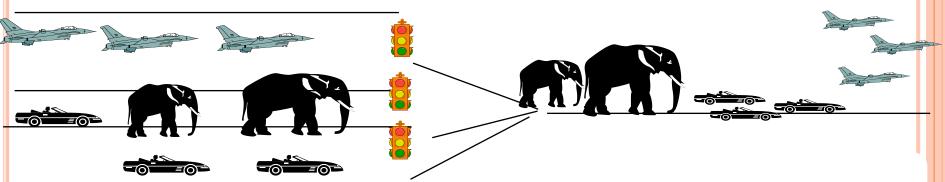




a) *First-come*, *first-served*



b) Fair queuing



c) Weighted fair queuing

THE INTEGRATED SERVICES MODEL

The Integrated Services model deals with two end-to-end services

Guaranteed service and

2. Controlled-load service on a per-flow

basis.

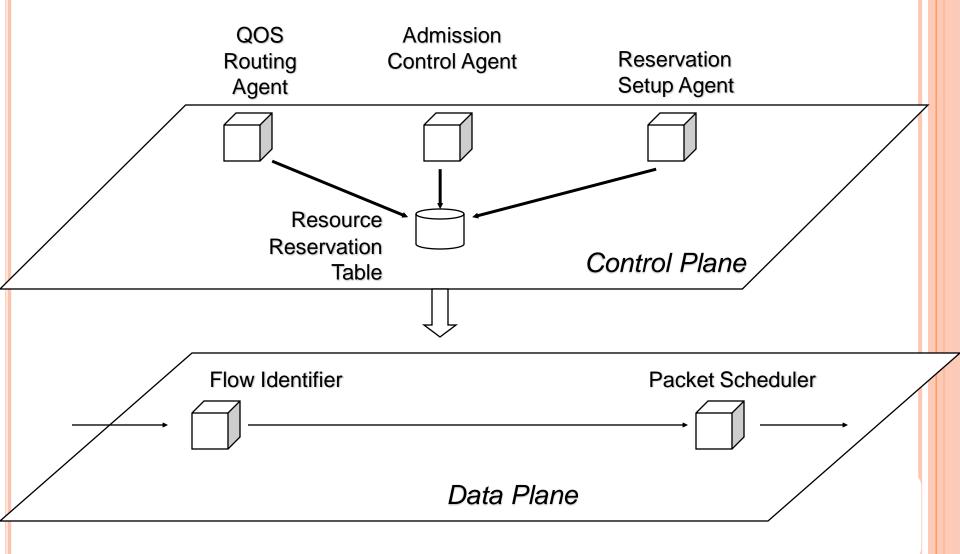


WHAT IS A FLOW?

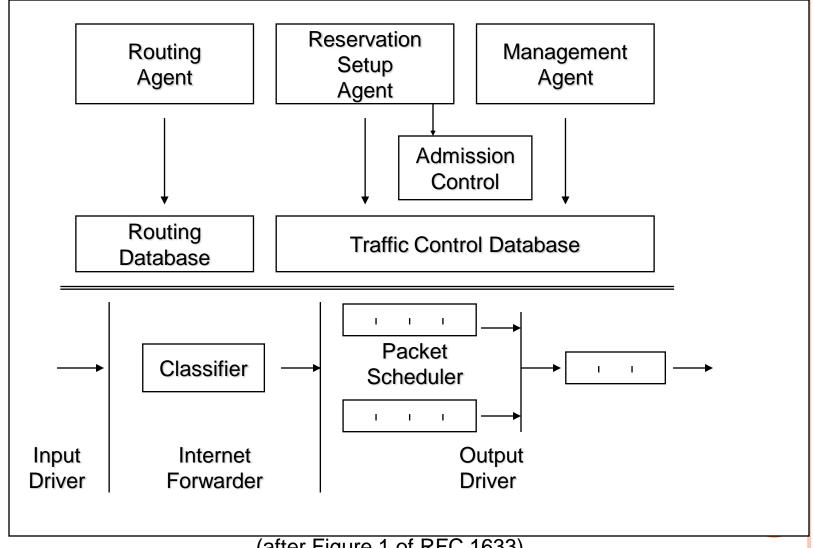
A flow is identified by a *quintuple* carried in both the IP- and transport-layer (TCP, UDP, SCTP) packet headers:

- Source IP Address
- Destination IP Address
- Protocol ID
- Source Port
- Destination Port

INTEGRATED SERVICES: CONTROL AND DATA PLANES



IMPLEMENTATION INTSERV MODEL FOR ROUTERS



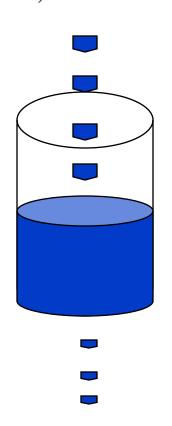
(after Figure 1 of RFC 1633)

FLOW SPECIFICATION (COMMON PARAMETERS)

- *Peak rate*: the highest rate at which a source may generate traffic (measured in bytes/sec)
- *Average rate:* the average transmission rate over a time interval
- Burst size: the maximum amount of data that can be injected into the network at peak rate
- *Minimum bandwidth:* the minimum amount of bandwidth required by an application flow over a specified period of time (measured in bytes/sec)
- *Delay:* either the average or worst-case end-to-end delay
- *Jitter:* the difference between the maximum delay and minimum delay
- Loss rate: the ratio of the number of lost packets to the total number of transmitted packets

Note: The full set of the parameters is defined in RFC 2215.

TRAFFIC SPECIFICATION MODEL (LEAKY BUCKET)

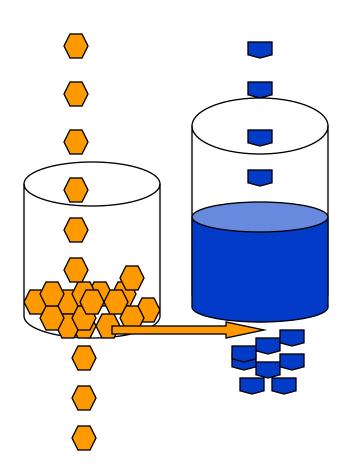


In this model, a host outputs packets into a networkcontrolled queue of size **b** bytes (bucket depth), which is processed at a rate *r* bytes/sec.

If the bucket is full, and packets arrive at a higher rate than r, the overflow packets get dropped.

Result: Bursts are eliminated

TRAFFIC SPECIFICATION MODEL (TOKEN BUCKET)



In this model, tokens are output into a token bucket of size **b** (token bucket depth), at a rate of **r**. Tokens stop arriving when the bucket is full.

A token allows output of a fixed number of bytes from the queue, and is then destroyed. No packet can be transmitted if there are no tokens in the token bucket, but the tokens can be saved up.

Result: Controlled bursts are allowed

TOKEN BUCKET PROPERTIES

• The total traffic volume V(t) the host can send over period t is bounded by a linear function:

$$V(t) \le rt + b$$

• Given the maximum output rate of *M* (bytes/sec), the maximum burst time T can be found (this is another homework assignment)

• The token arrival rate defines the long term *average rate* of the traffic

Note: The tokens can be saved, up to the full size **b** of the token bucket

MIX AND MATCH!

• We can shape the traffic never to exceed a given rate R while allowing controlled bursts by placing a token bucket after a leaky bucket that enforces rate R:

