Chair for Network Architectures and Services – Prof. Carle
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# Master Course Computer Networks IN2097

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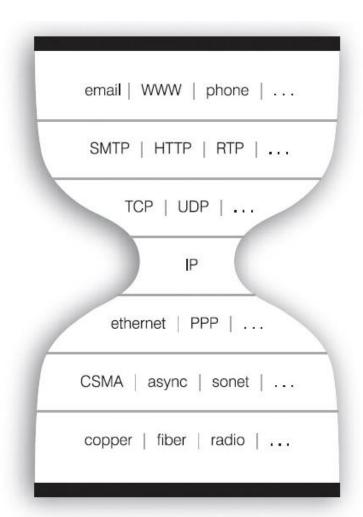




- Introduction to Network Address Translation
- Behavior of NAT
- □ The NAT Traversal problem
- Solutions to the problem
- Large Scale NATs



- More and more devices connect to the Internet
  - PCs
  - Cell phones
  - Internet radios
  - TVs
  - Home appliances
  - Future: sensors, cars...
- IP addresses need to be globally unique
  - IPv4 provides a 32bit field
  - Many addresses not usable because of classful allocation
- → We are running out of IP addresses





#### **Address Space**

- IP addresses are assigned by the Internet Assigned Numbers Authority (IANA)
- □ RFC 1918 (published in 1996) directs IANA to reserve the following IPv4 address ranges for private networks
  - **1**0.0.0.0 10.255.255.255
  - 172.16.0.0 172.31.255.255
  - 192.168.0.0 192.168.255.255
- The addresses may be used and reused by everyone
  - Not routed in the public Internet
  - Therefore a mechanism for translating addresses is needed

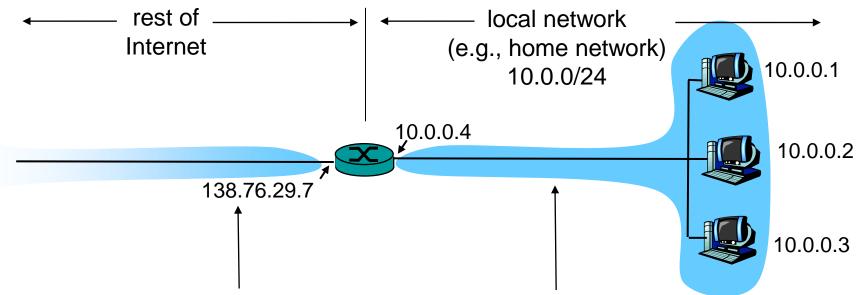


#### First approach - Network Address Translation

- Idea: only hosts communicating with the public Internet need a public address
  - Once a host connects to the Internet we need to allocate one
  - Communication inside the local network is not affected
- A small number of public addresses may be enough for a large number of private clients
- Only a subset of the private hosts can connect at the same time
  - not realistic anymore (always on)
  - we still need more than one public IP address



# **NAPT: Network Address and Port 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



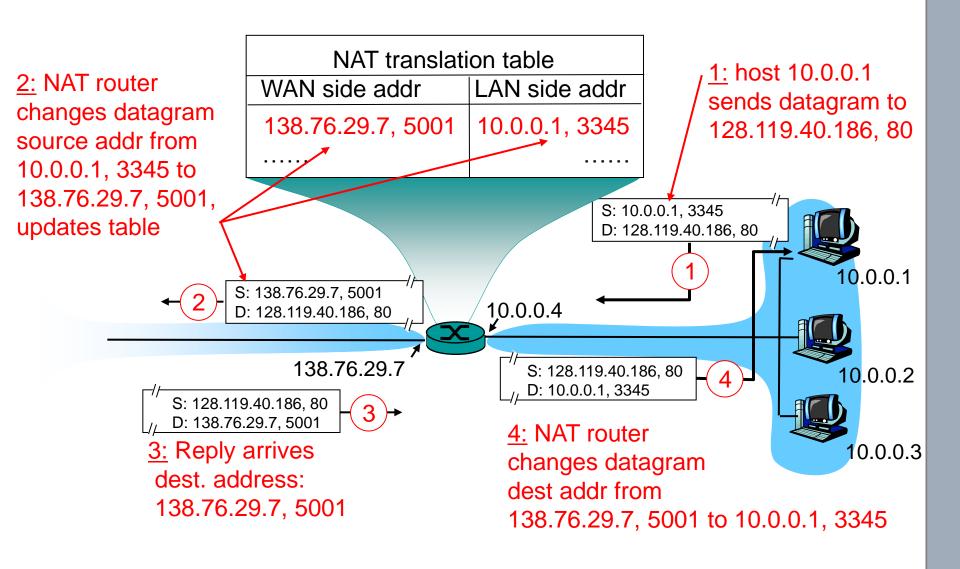
#### **NAT: Network Address Translation**

#### Implementation: NAT router must:

- On 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
  - -> we have to maintain a state in the NAT
- 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: Network Address Translation**

#### NAPT:

- ~65000 simultaneous connections with a single LAN-side address!
- helps against the IP shortage
- More advantages:
  - we can change addresses of devices in local network without notifying outside world
  - we can change ISP without changing local addresses
  - devices inside local net not explicitly addressable/visible by the outside world (a security plus)
- NAT is controversal:
  - routers should only process up to layer 3
  - violates end-to-end argument



### **NAT Behavior and Implementation**

- Implementation not standardized
  - thought as a temporary solution
- implementation differs from model to model
  - if an application works with one NAT does not imply that is always works in a NATed environment
- NAT behavior
  - Binding (which external mapping is allocated)
    - NAT binding
    - Port binding
  - Endpoint filtering (who is allowed to access the mapping)

# Binding

- When creating a new state, the NAT has to assign a new source port and IP address to the connection
- Port binding describes the strategy a NAT uses for the assignment of a new external source port
  - Port Preservation (if possible)
  - Some algorithm (e.g. +1)
  - Random

# NAT binding

- NAT binding describes the behavior of the NAT regarding the reuse of an existing binding
  - two consecutive connections from the same transport address (combination of IP address and port)
  - 2 different bindings?
  - If the binding is the same → Port prediction possible
- Endpoint Independent
  - the external port is only dependent on the source transport address
  - both connections have the same IP address and port
- Endpoint Dependent
  - a new port is assigned for every connection
  - strategy could be random, but also something more predictable
  - Port prediction is hard

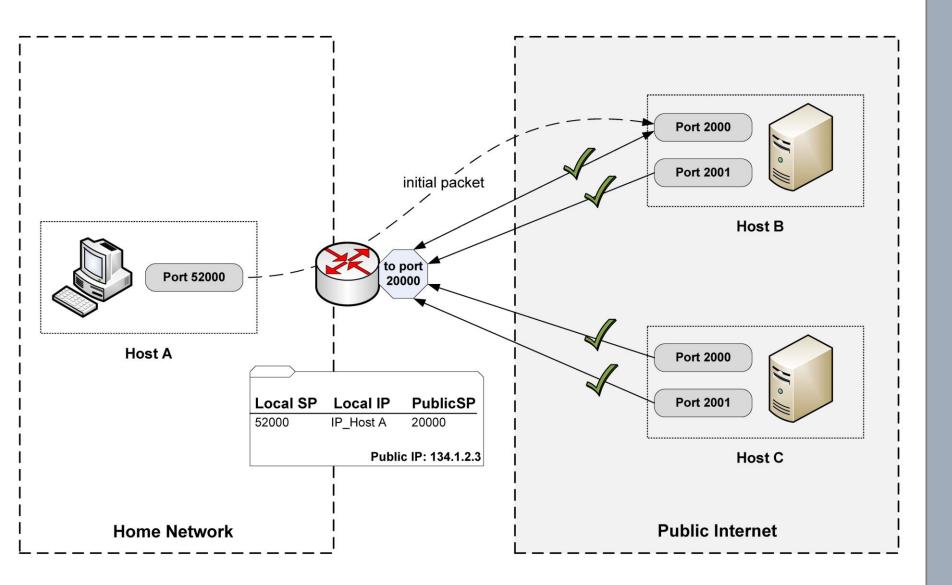


#### **Endpoint filtering**

- Filtering describes
  - how existing mappings can be used by external hosts
  - How a NAT handles incoming connections
- Independent-Filtering:
  - All inbound connections are allowed
  - Independent on source address
  - As long as a packet matches a state it is forwarded
  - No security
- Address Restricted Filtering:
  - packets coming from the same host (matching IP-Address) the initial packet was sent to are forwarded
- Address and Port Restricted Filtering:
  - IP address and port must match

- With Binding and Filtering 4 NAT types can be defined (RFC 3489)
- Full Cone NAT
  - Endpoint independent
  - Independent filtering
- Address Restricted NAT
  - Endpoint independent binding
  - Address restricted filtering
- Port Address Restricted NAT
  - Endpoint independent binding
  - Port address restricted filtering
- Symmetric NAT
  - Endpoint dependent binding
  - Port address restricted filtering

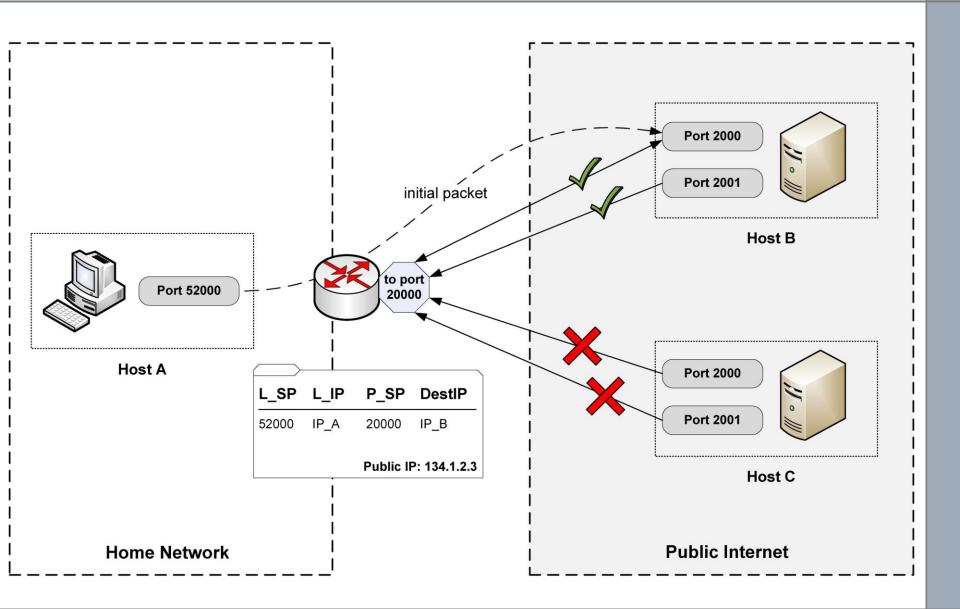
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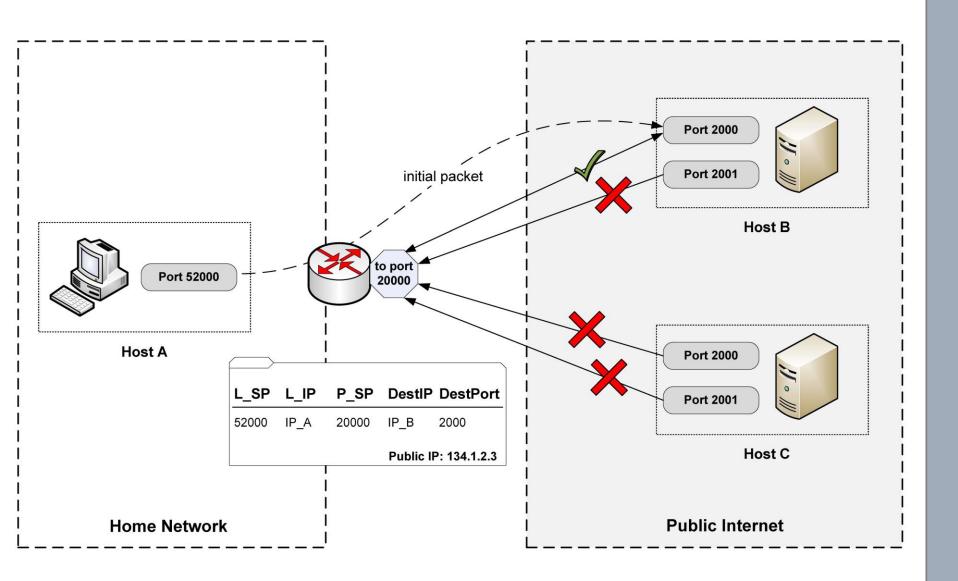
#### **Address Restricted Cone NAT**



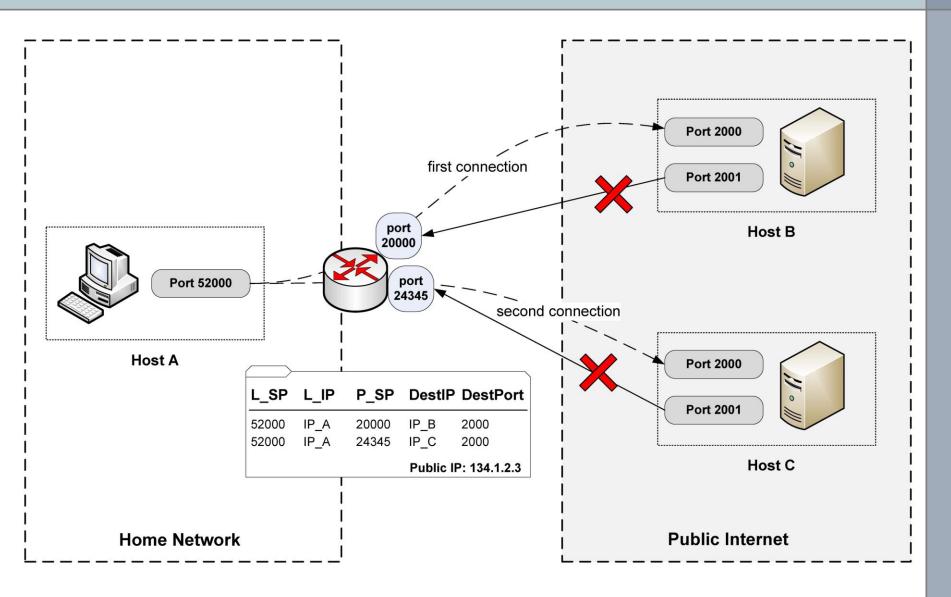
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# **Port Address Restricted Cone NAT**



- With Binding and Filtering 4 NAT types can be defined (RFC 3489)
- Full Cone NAT
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- Port Address Restricted NAT
  - Endpoint independent binding
  - Port address restricted filtering
- Symmetric NAT
  - Endpoint dependent binding
  - Port address restricted filtering





# And where is the problem?

- NAT was designed for the client-server paradigm
- Nowadays the internet consists of applications such as
  - P2P networks
  - Voice over IP
  - Multimedia Streams
- □ Protocols are getting more and more complex
  - Multiple layer 4 connections (data and control session)
  - Realm specific addresses in layer 7
- Connectivity requirements have changed
  - P2P is becoming more and more important
    - Especially for future home and services
  - Direct connections between hosts is necessary
- NATs break the end-to-end connectivity model of the internet
  - Inbound packets can only be forwarded if an appropriate mapping exists
  - Mappings are only created on outbound packets



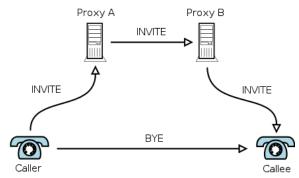
#### **NAT-Traversal Problem**

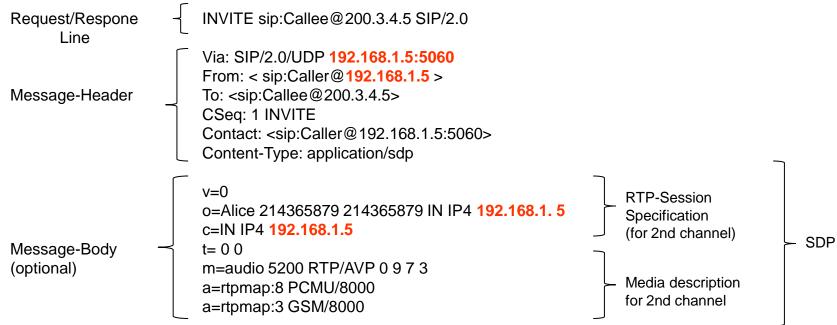
- □ Divided into four categories: (derived from IETF-RFC 3027)
  - Realm-Specific IP-Addresses in the Payload
    - Session Initiation Protocol (SIP)
  - Peer-to-Peer Applications
    - Any service behind a NAT
  - Bundled Session Applications (Inband Signaling)
    - FTP
    - Real time streaming protocol (RTSP)
    - SIP together with SDP (Session Description Protocol)
  - Unsupported Protocols
    - SCTP (Stream Control Transmission Protocol)
    - IPSec



#### **Example: Session Initiation Protocol (SIP)**

- Realm Specific IP addresses in the payload (SIP)
- Bundled Session Application (RTP)

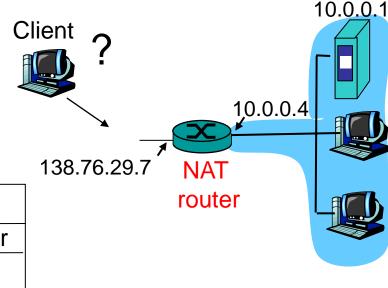






### **Example: P2P applications**

- 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 NATted address: 138.76.29.7
  - NAT does not have any idea where to forward packets to



NAT translation table	
WAN side addr	LAN side addr
138.76.29.7, 80	10.0.0.1, 80



### **Existing Solutions to the NAT-Traversal Problem**

- Individual solutions
  - Explicit support by the NAT
    - Static port forwarding, ALG, UPnP, NAT-PMP
  - NAT-behavior based approaches
    - dependent on knowledge about the NAT
    - Hole Punching using STUN (IETF RFC 3489)
  - External Data-Relay
    - TURN (IETF Draft)
- Frameworks integrating several techniques
  - framework selects a working technique
  - ICE as the most promising for VoIP (IETF Draft)



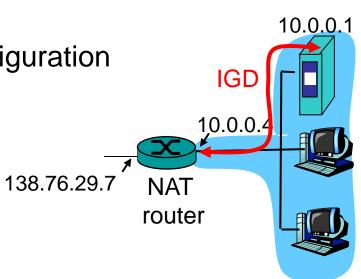
#### **Explicit support by the NAT (1)**

- Application Layer Gateway (ALG)
  - implemented on the NAT device and operates on layer 7
  - supports Layer 7 protocols that carry realm specific addresses in their payload
    - SIP, FTP
- Advantages
  - transparent for the application
  - no configuration necessary
- Drawbacks
  - protocol dependent (e.g. ALG for SIP, ALG for FTP...)
  - may or may not be available on the NAT device



#### **Explicit support by the NAT (2)**

- Universal Plug and Play (UPnP)
  - Automatic discovery of services (via Multicast)
  - Internet Gateway Device (IGD) for NAT-Traversal
- IGD allows NATed host to
  - automate static NAT port map configuration
  - learn public IP address (138.76.29.7)
  - add/remove port mappings (with lease times)



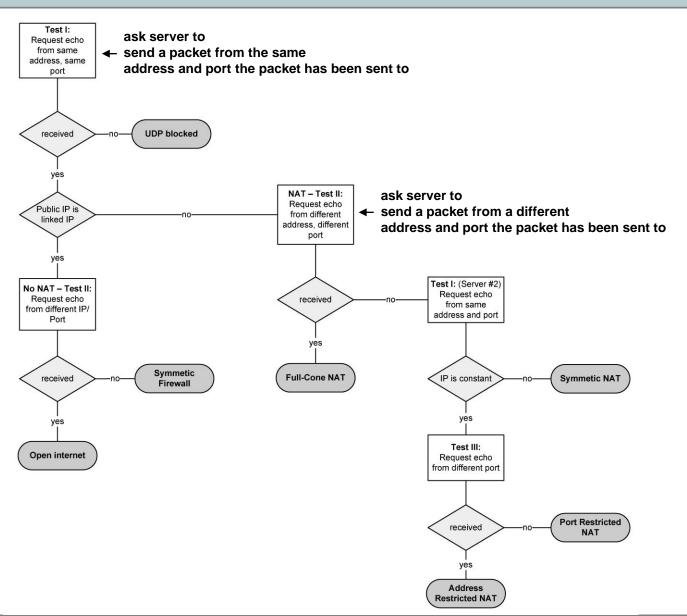
- Drawbacks
  - no security, evil applications can establish port forwarding entries
  - doesn't work with cascaded NATs



# Behavior based (1): STUN

- □ Simple traversal of UDP through NAT (old) (RFC 3489)
  - Session Traversal Utilities for NAT (new) (RFC 5389)
- Lightweight client-server protocol
  - queries and responses via UDP (optional TCP or TCP/TLS)
- Helps to determine the external transport address (IP address and port) of a client.
  - e.g. query from 192.168.1.1:5060 results in 131.1.2.3:20000
- Algorithm to discover NAT type
  - server needs 2 public IP addresses

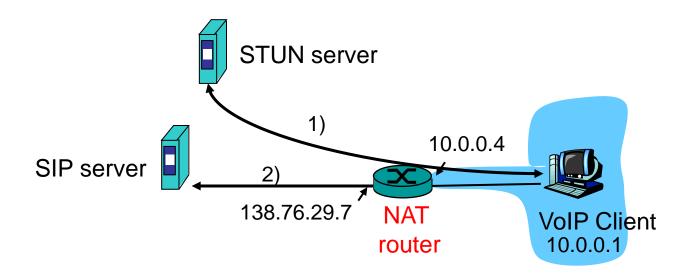






#### **Example: STUN and SIP**

- VoIP client queries STUN server
  - learns its public transport address
  - can be used in SIP packets



Request/Respone

Line

INVITE sip:Callee@200.3.4.5 SIP/2.0

Via: SIP/2.0/UDP **138.76.29.7:5060** 

From: < sip:Caller@138.76.29.7 >

Message-Header To: <sip:Callee@200.3.4.5>

CSeq: 1 INVITE

Contact: <sip:Caller@138.76.29.7:5060>

Content-Type: application/sdp



#### **Limitations of STUN**

- STUN only works if
  - the NAT assigns the external port (and IP address) only based on the source transport address
  - Endpoint independent NAT binding
    - Full Cone NAT
    - Address Restricted Cone NAT
    - Port Address restricted cone NAT
  - Not with symmetric NAT!
- □ Why?
  - Since we first query the STUN server (different IP and port) and then the actual server
  - The external endpoint must only be dependent on the source transport address

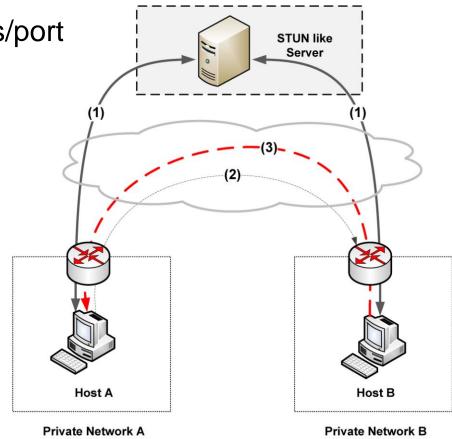


### **STUN and Hole Punching**

- STUN not only helps if we need IP addresses in the payload
  - also for establishing a direct connection between two peers

 determine external IP address/port and exchange it through Rendezvous Point

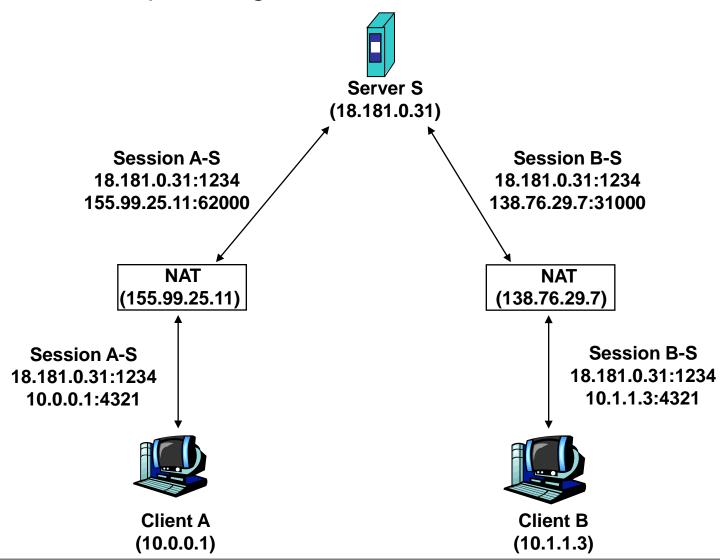
- 2) both hosts send packets towards the other host outgoing packet creates hole
- establish connection.
   hole is created by first packet





# **Hole Punching in detail**

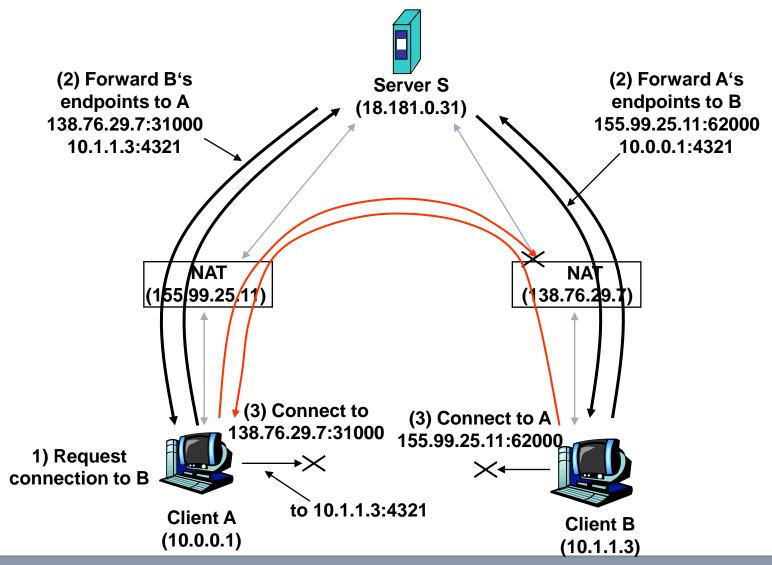
Before hole punching





### Hole Punching in detail

#### Hole punching





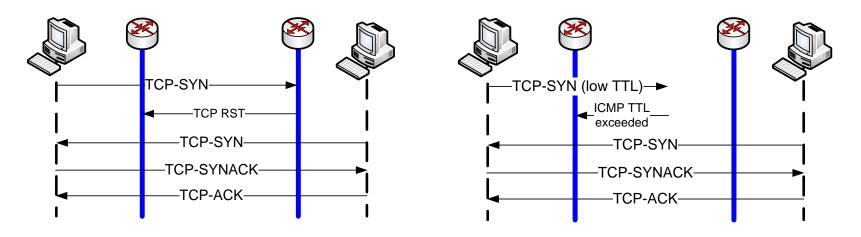
# **DIY Hole Punching: practical example**

- You need 2 hosts
  - One in the public internet (client)
  - One behind a NAT (server)
- Firstly start a UDP listener on UDP port 20000 on the "server" console behind the NAT/firewall
  - server/1# nc -u -l -p 20000
- □ An external computer "client" then attempts to contact it
  - client# echo "hello" | nc -p 5000 -u serverIP 20000
  - Note: 5000 is the source port of the connection
- as expected nothing is received because the NAT has no state
- Now on a second console, server/2, we punch a hole
  - Server/2# hping2 -c 1 -2 -s 20000 -p 5000 clientIP
- On the second attempt we connect to the created hole
  - client# echo "hello" | nc -p 5000 -u serverIP 20000



### **TCP Hole Punching**

- Hole Punching not straight forward due to stateful design of TCP
  - 3-way handshake
  - Sequence numbers
  - ICMP packets may trigger RST packets
- □ Low/high TTL(Layer 3) of Hole-Punching packet
  - As implemented in STUNT (Cornell University)

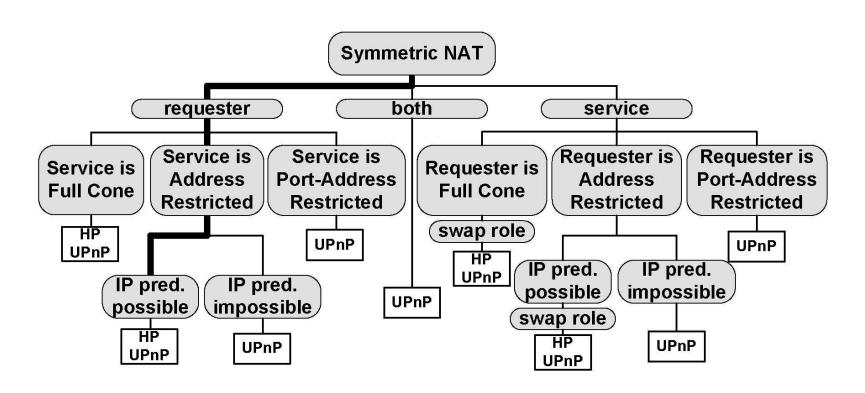


Bottom line: NAT is not standardized



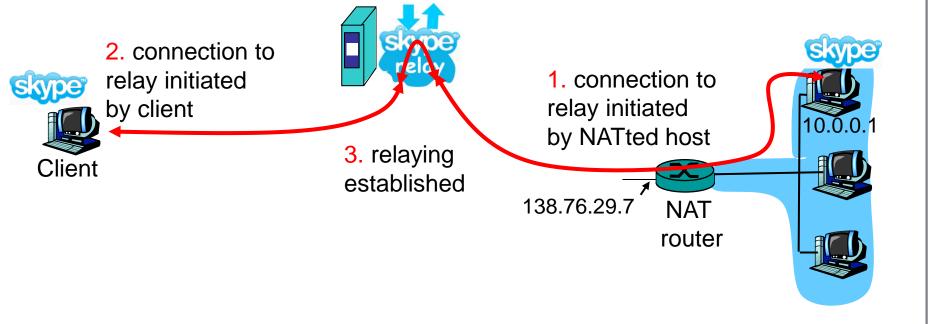
#### **Symmetric NATs**

- How can we traverse symmetric NATs
  - Endpoint dependent binding
    - hole punching in general only if port prediction is possible
  - Address and port restricted filtering





- □ relaying (used in Skype)
  - NATed client establishes connection to relay
  - External client connects to relay
  - relay bridges packets between to connections
  - Traversal using Relay NAT (TURN) as IETF draft





#### **Frameworks**

- Interactive Connectivity Establishment (ICE)
  - IETF draft
  - mainly developed for VoIP
  - signaling messages embedded in SIP/SDP
- All possible endpoints are collected and exchanged during call setup
  - local addresses
  - STUN determined
  - TURN determined
- All endpoints are "paired" and tested (via STUN)
  - best one is determined and used for VoIP session
- Advantages
  - high sucess rate
  - integrated in application
- Drawbacks
  - overhead
  - latency dependent on number of endpoints (pairing)