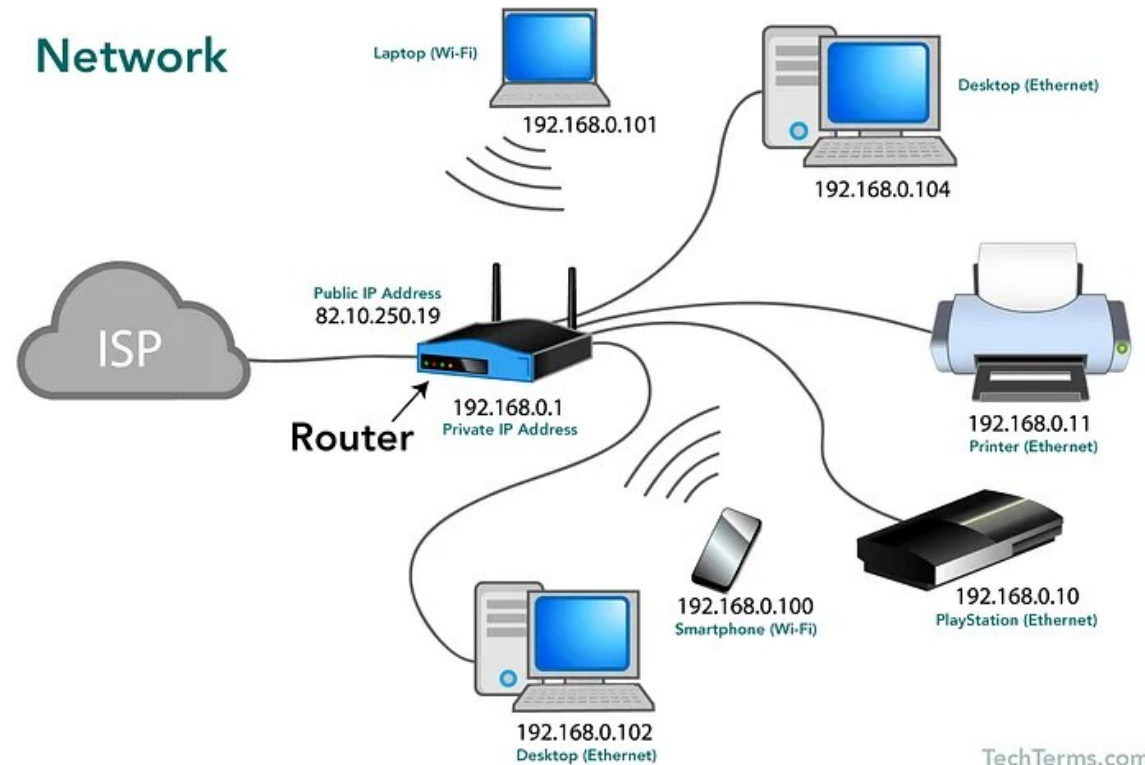


NAT, UDP hole punching,
WebRTC

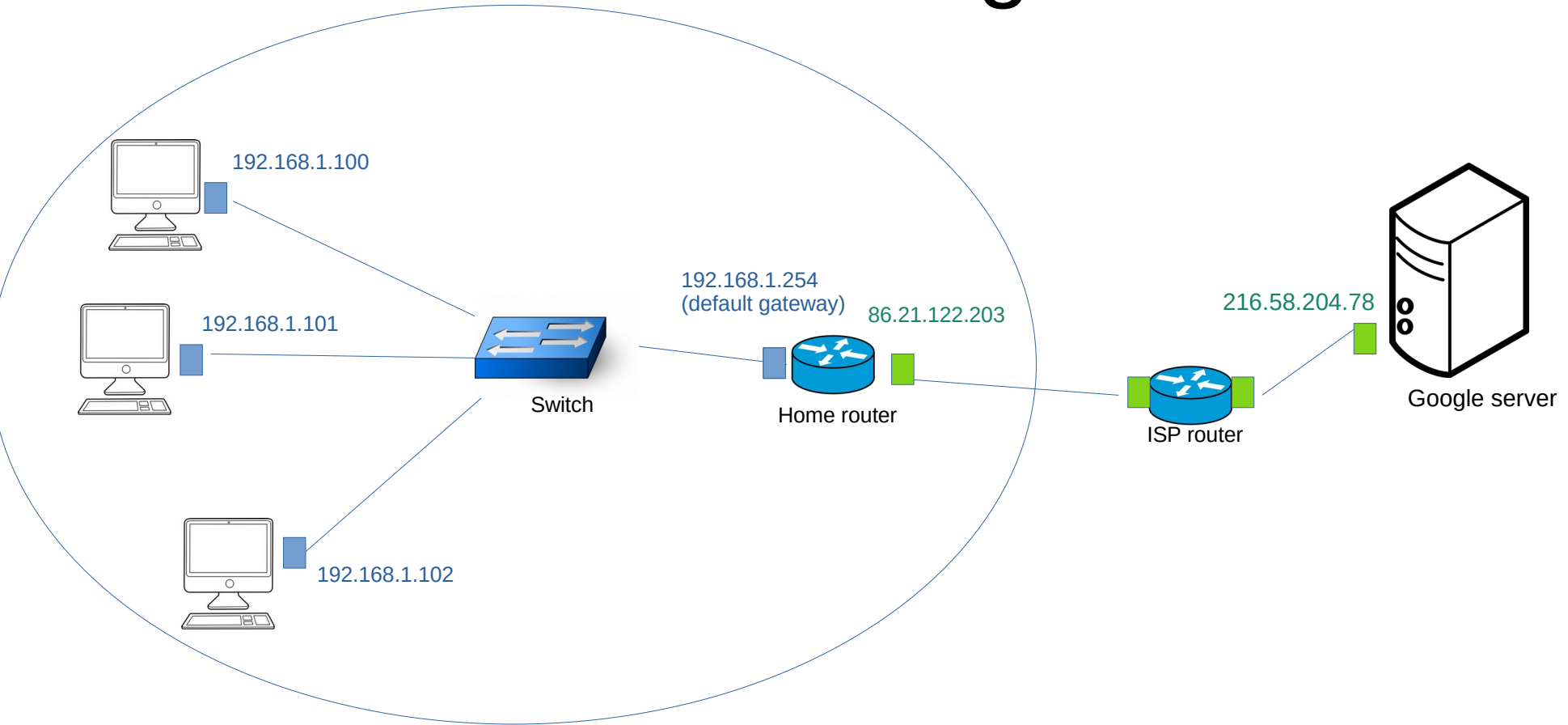
Network Address Translation (NAT)

- Only ~4.3 billion IPv4 address
- IPv6 is dumb or something idk why people don't use it
- The solution:
 - All devices on a local network have the same public IP address
 - Router must forward traffic to devices on the network



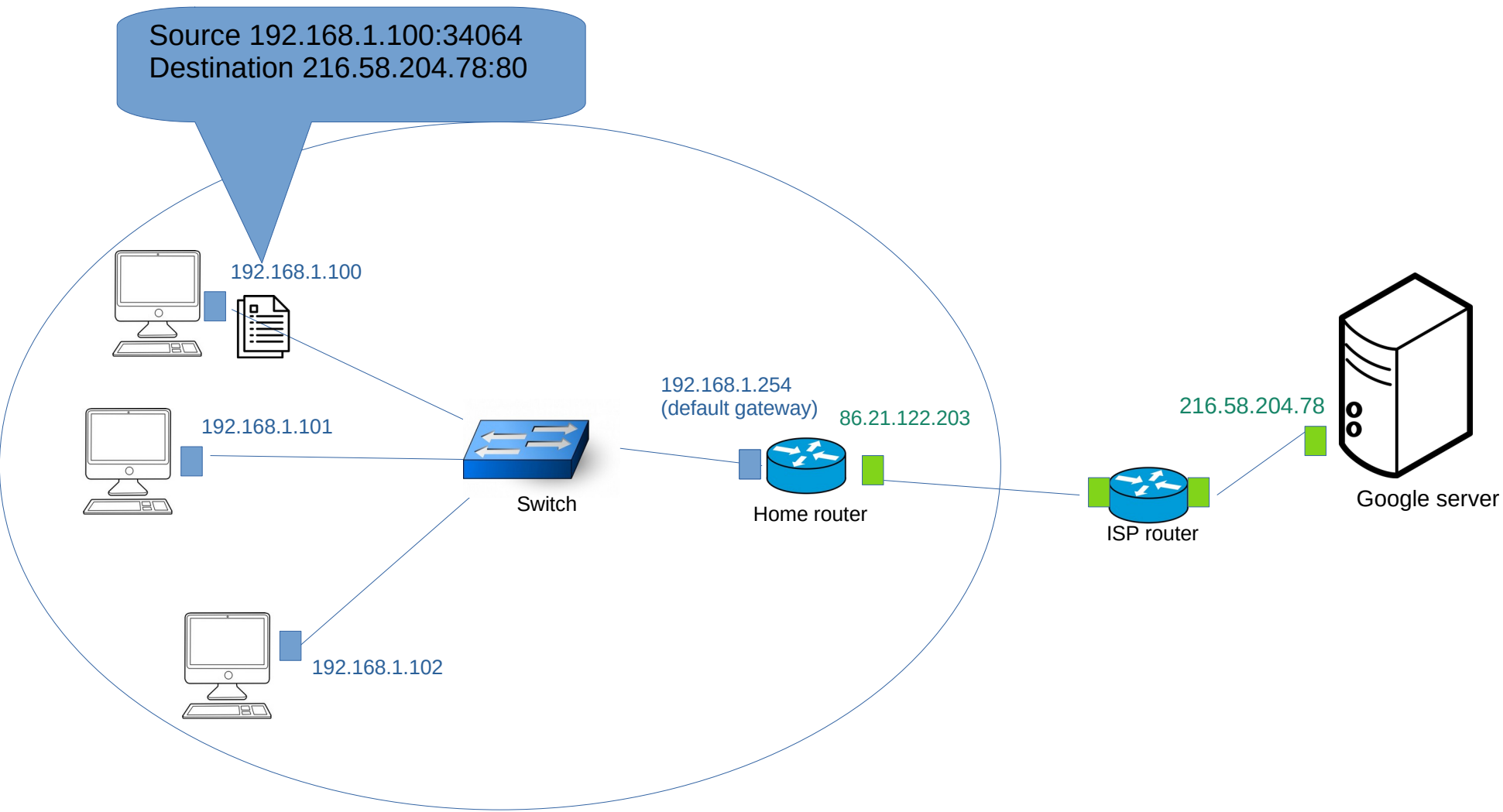
TechTerms.com

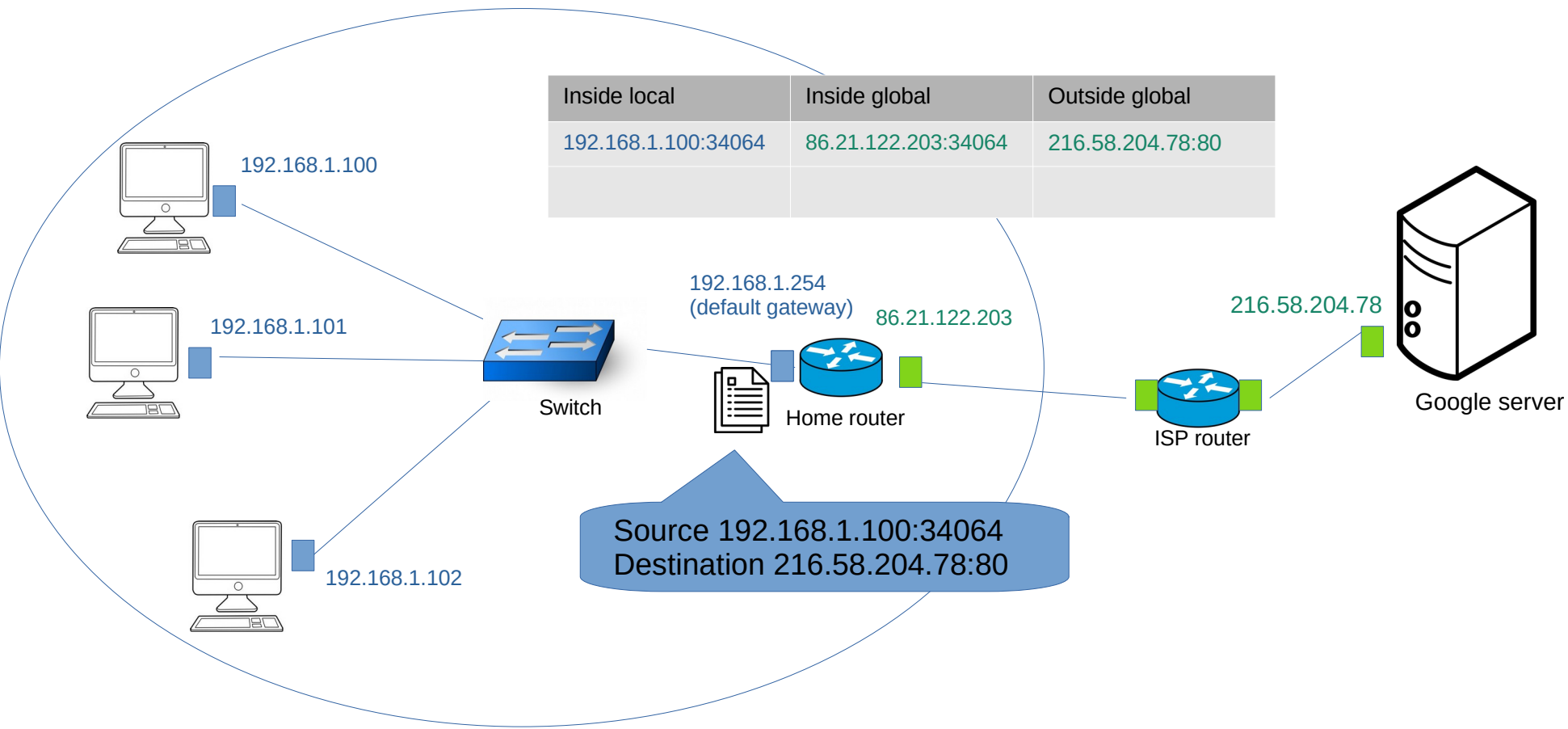
Network configuration

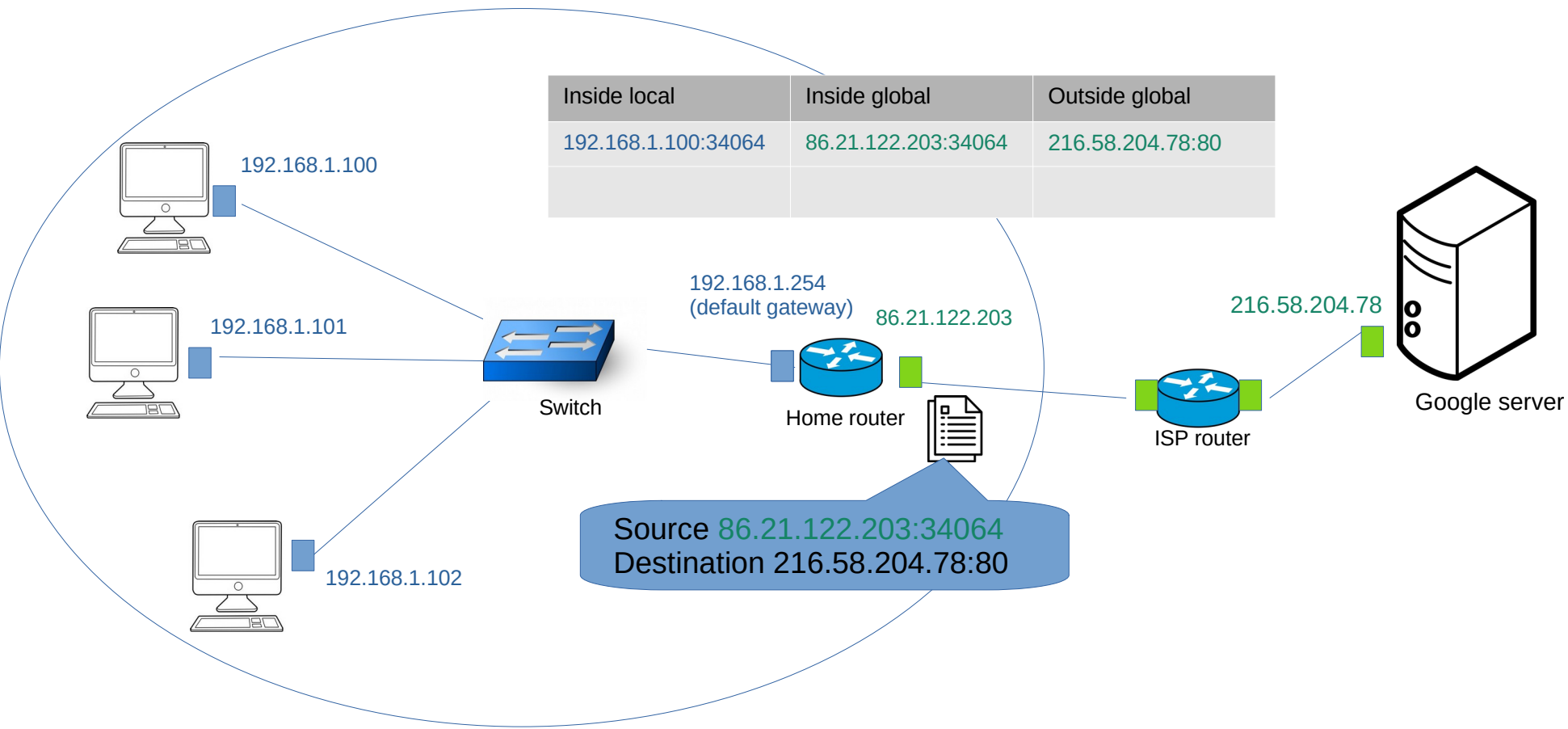


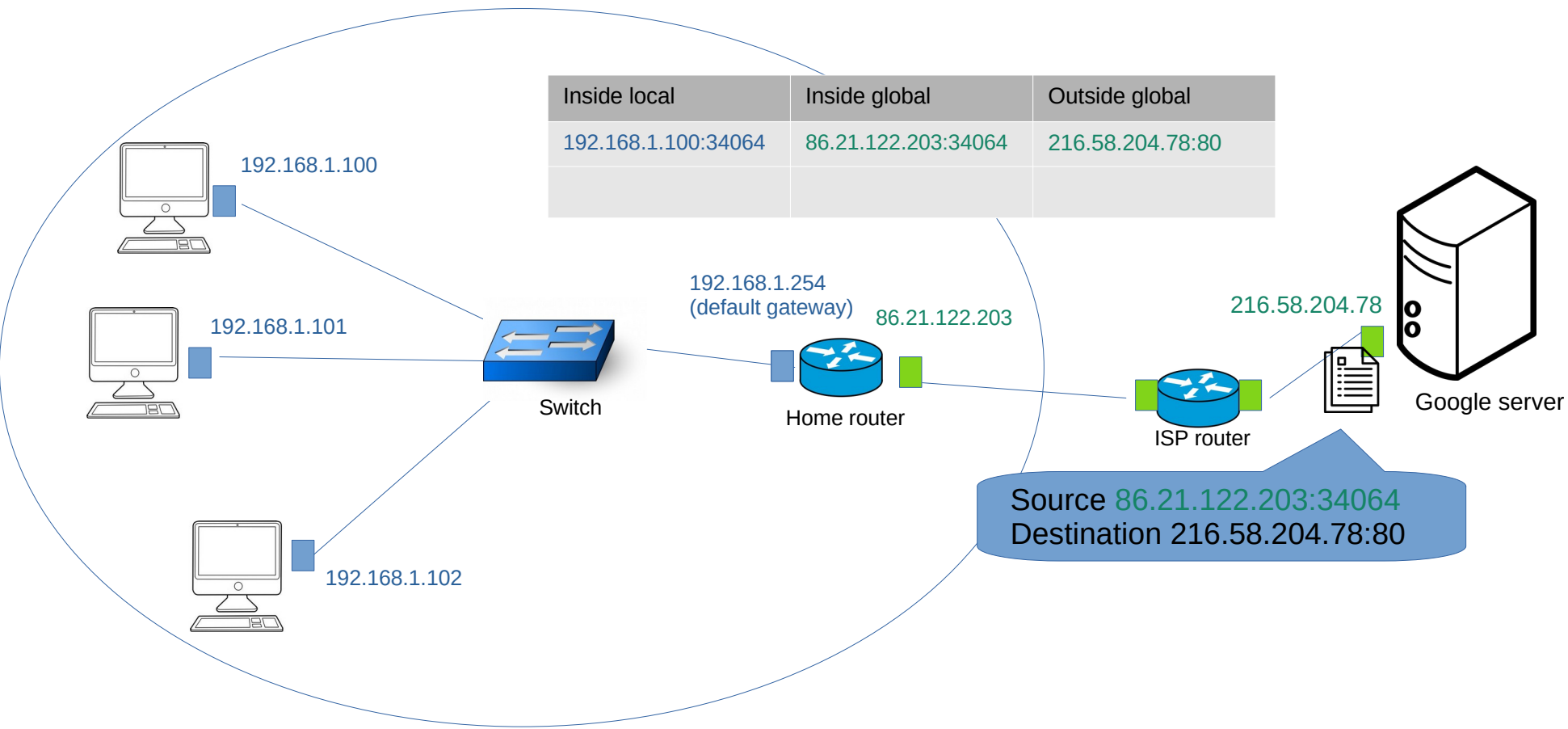
- Packets to 192.168.1.* go to the device directly
- Everything else is sent to the default gateway for forwarding

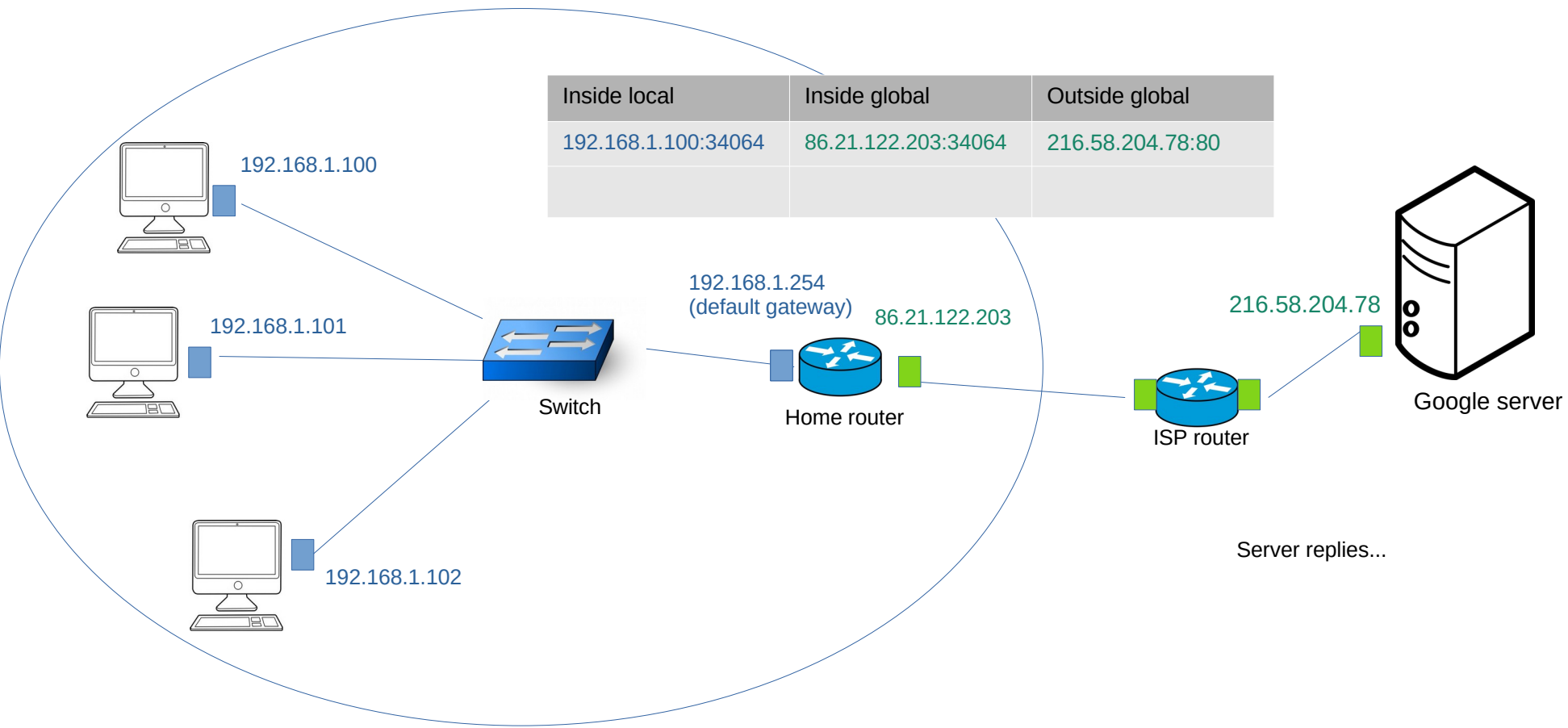
Source 192.168.1.100:34064
Destination 216.58.204.78:80

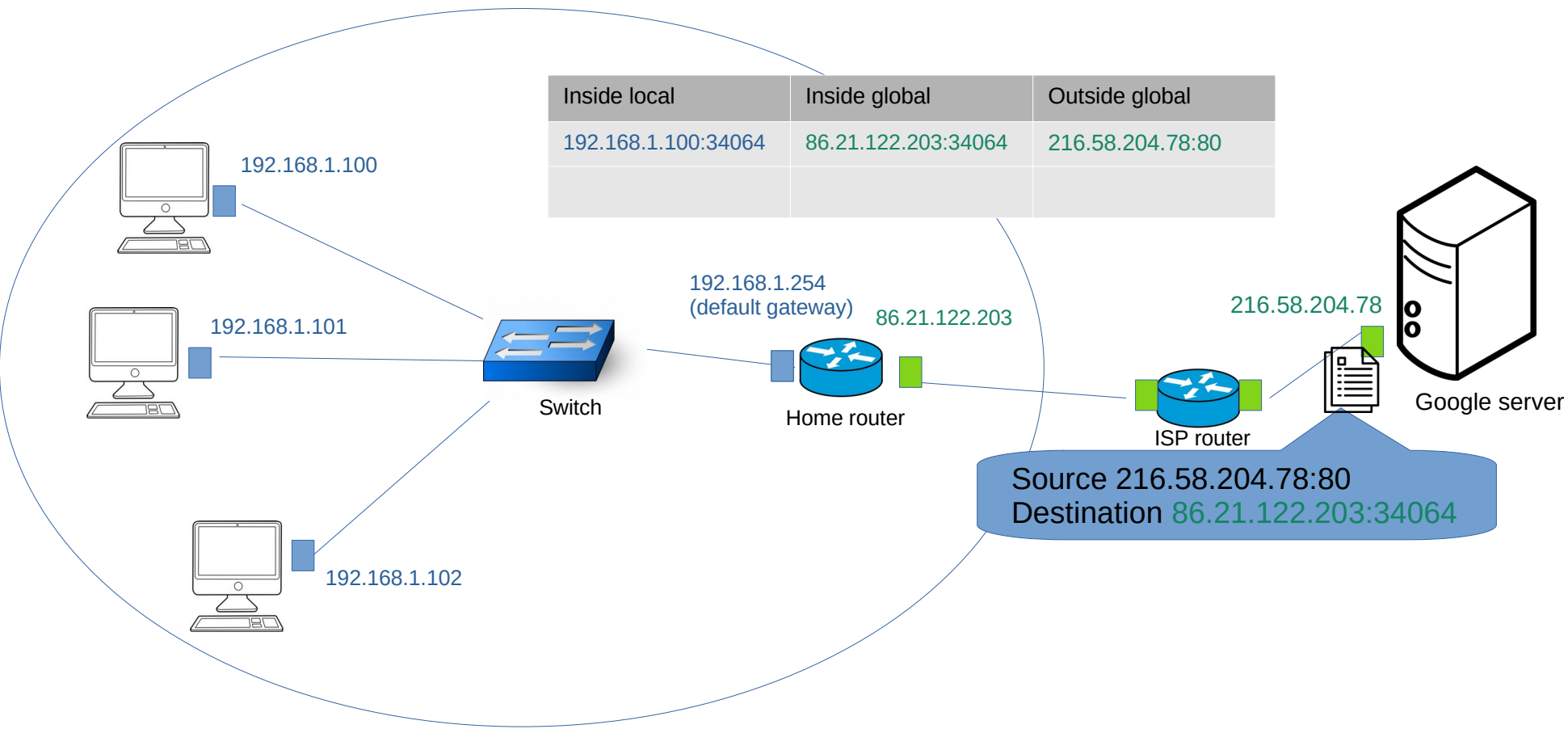


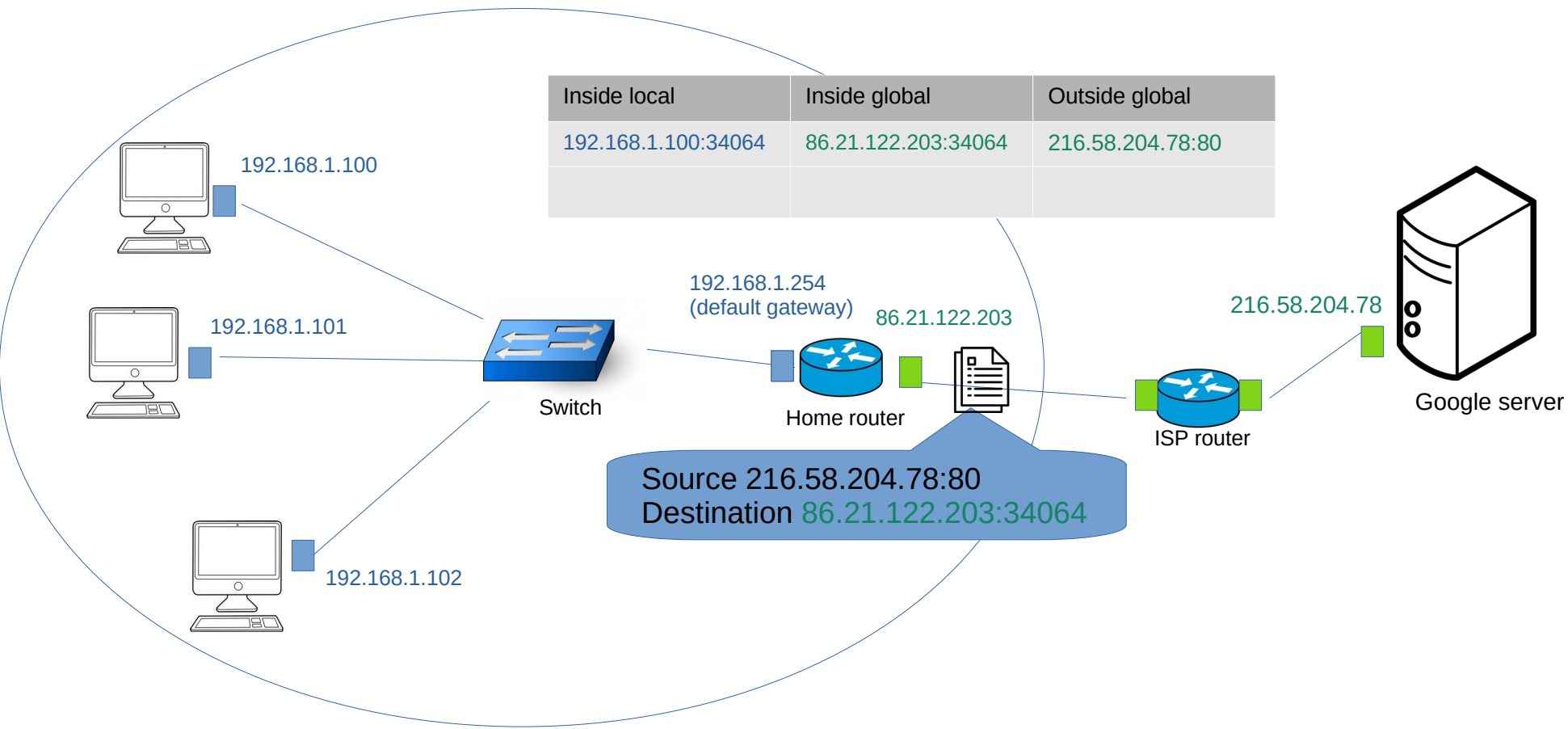


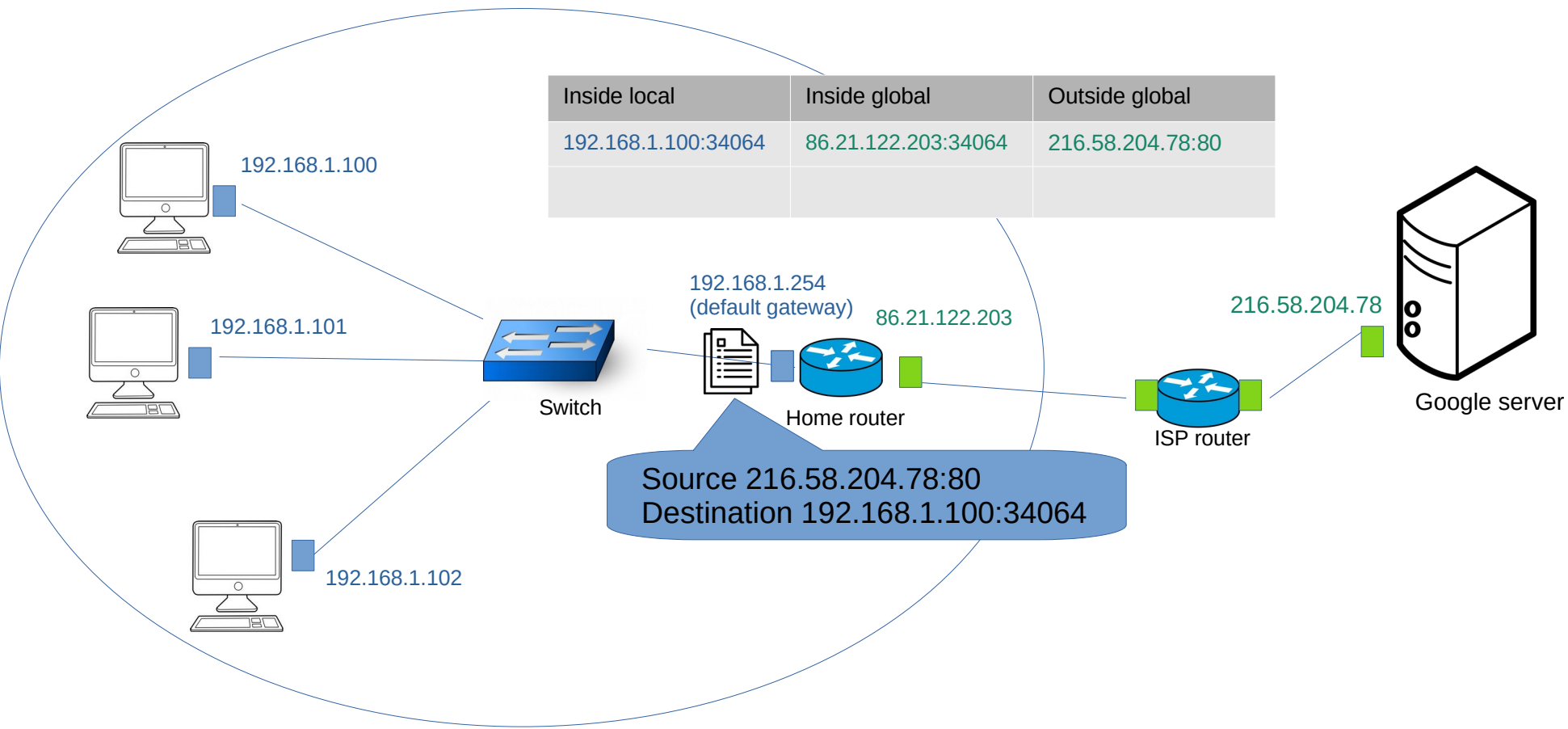






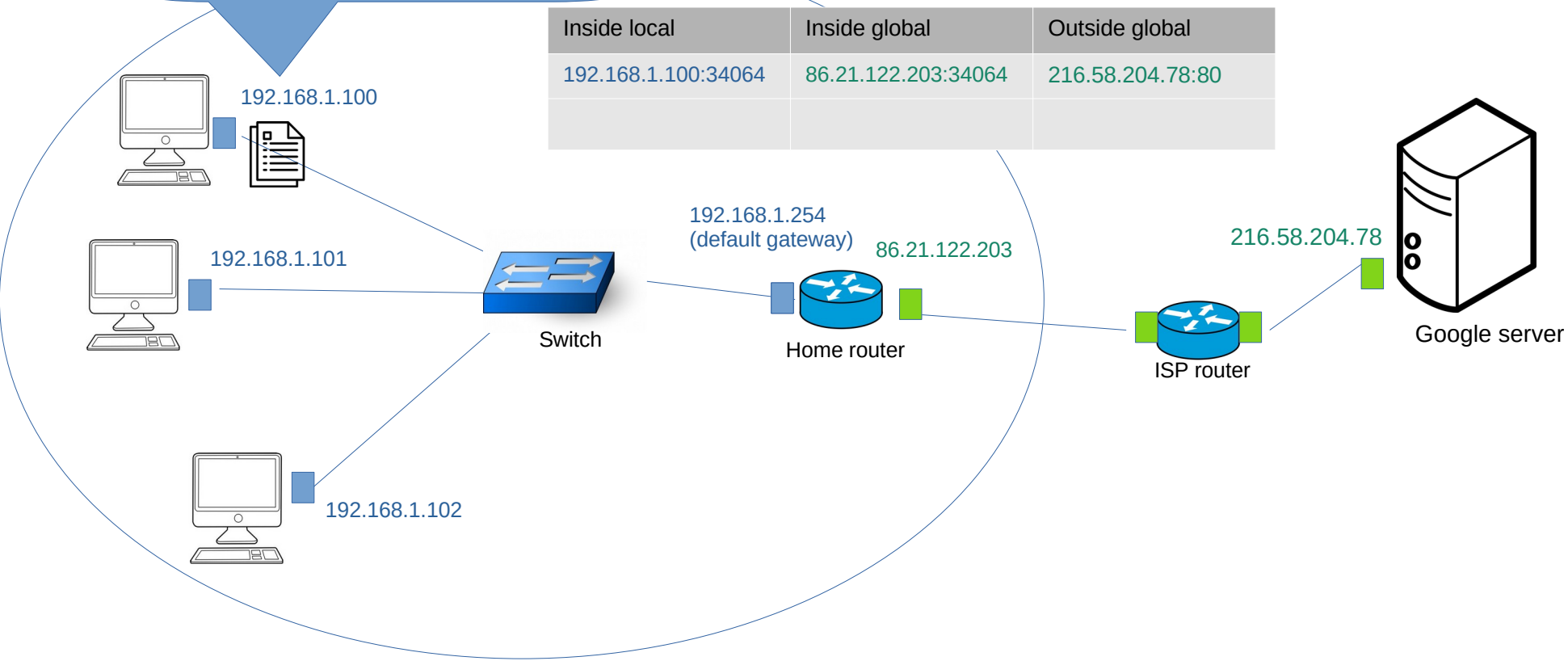






Source 216.58.204.78:80
Destination 192.168.1.100:34064

Inside local	Inside global	Outside global
192.168.1.100:34064	86.21.122.203:34064	216.58.204.78:80



Network Address Translation (NAT)

- If two computers are using the same inside local port number, they will need to be mapped to different inside global ports



- Some routers change ports by default
- In general completely up to the router how it wants to map ports

Session Traversal Utilities for NAT (STUN)

- A protocol to work out how the router is translating a source address
- Send a UDP packet to a STUN server (e.g. stun.ekiga.net)
- STUN server replies with what it saw for the source IP and port

Verbose mode wlan0 network interface Source port Test number

```
[~]$ stun -v -i wlan0 -p 11341 stun.ekiga.net 1
STUN client version 0.97
error was 11
running test number 1
Opened port 11341 with fd 3
Encoding stun message:
Encoding ChangeRequest: 0

About to send msg of len 28 to 216.93.246.18:3478
Got a response
Received stun message: 92 bytes
MappedAddress = [REDACTED]:11341
SourceAddress = 216.93.246.18:3478
ChangedAddress = 216.93.246.15:3479
Unknown attribute: 32800
ServerName = Vovida.org 0.98-CPC
  ok=1
  id=1:93:92:7:238:92:66:50:214:57:131:24:102:177:178:15
  mappedAddr=[REDACTED]:11341
  changedAddr=216.93.246.15:3479

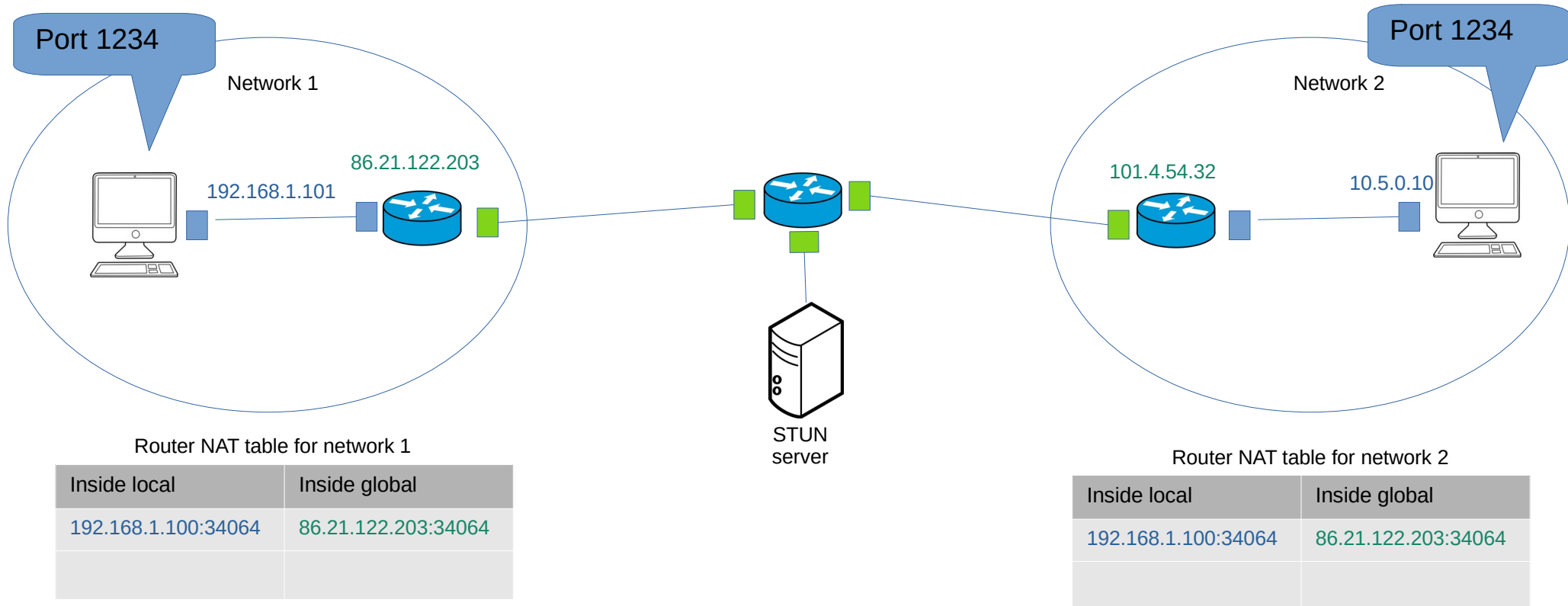
Return value is 0x000000
[~]$
```

What the IP and port were mapped to

UDP Hole Punching

- How can we create a peer-to-peer connection with all this NAT stuff????
- A solution: **UDP hole punching**

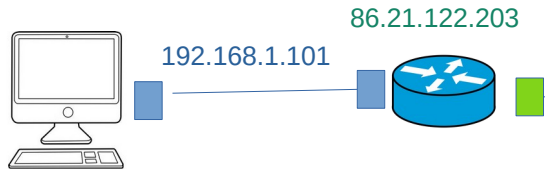
- 1) Make a request to a STUN server to find the mapped address
- 2) Send mapped address to peer by any method
- 3) Both peers send UDP packets from their source port to the other's mapped address



- 1) Make a request to a STUN server to find the mapped address
- 2) Send mapped address to peer
- 3) Both peers send UDP packets from their source port to the other's mapped address

The STUN server's
response said
86.21.122.203:4321

Network 1



Router NAT table for network 1

Inside local	Inside global	Outside global
192.168.1.101:1234	86.21.122.203:4321	216.93.246.18:3478
		86.21.122.203:4321

The STUN server's
response said
201.4.54.32:1234

Network 2

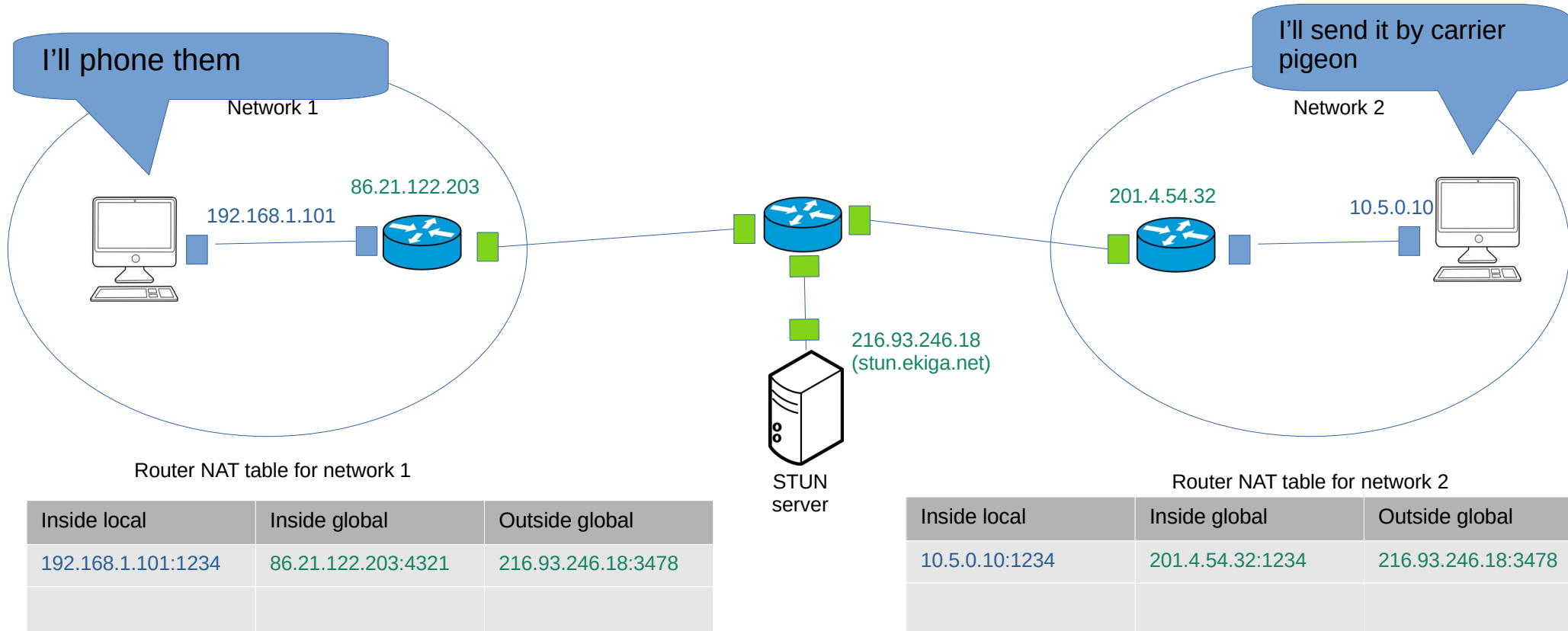


Router NAT table for network 2

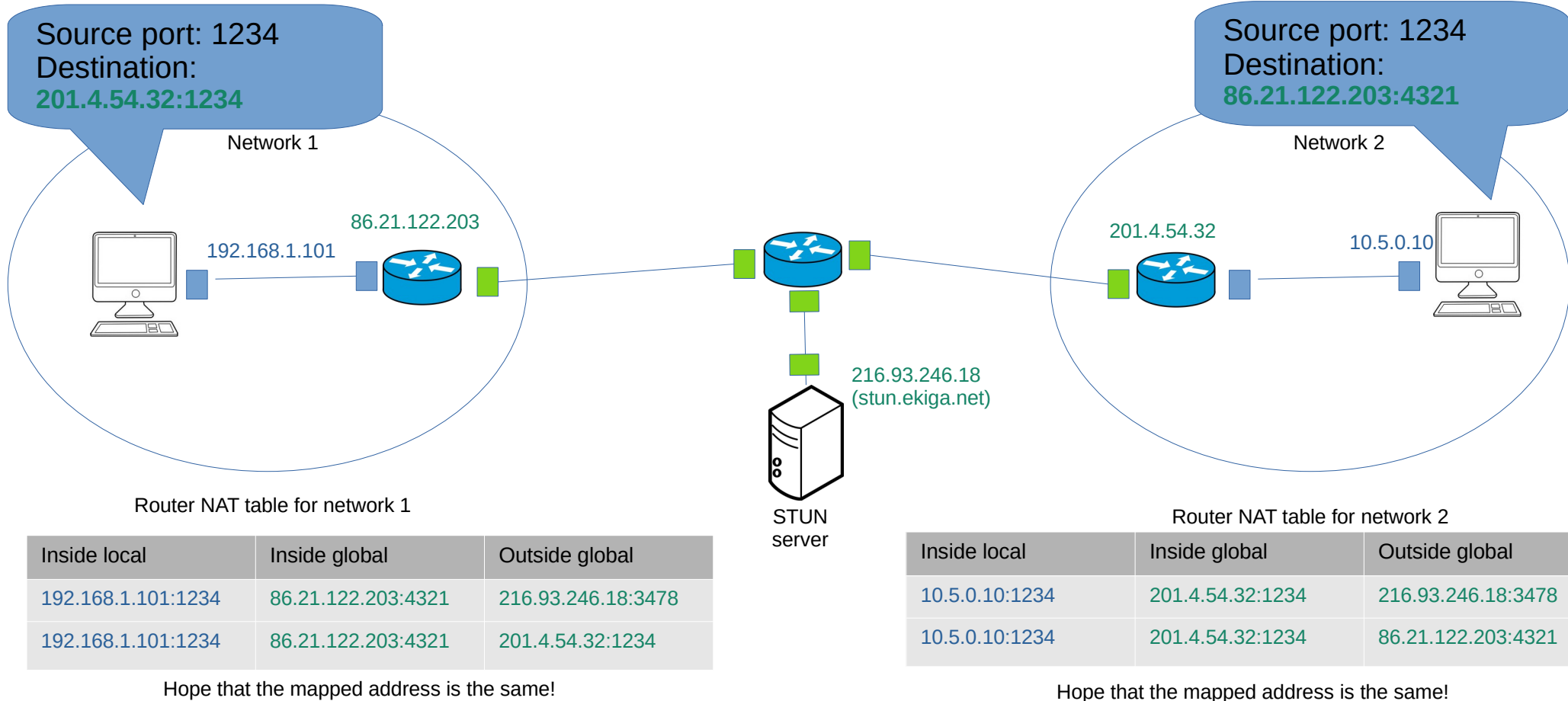
Inside local	Inside global	Outside global
10.5.0.10:1234	201.4.54.32:1234	216.93.246.18:3478



- 1) Make a request to a STUN server to find the mapped address
- 2) Send mapped address to peer**
- 3) Both peers send UDP packets from their source port to the other's mapped address



- 1) Make a request to a STUN server to find the mapped address
- 2) Send mapped address to peer
- 3) **Both peers send UDP packets from their source port to the other's mapped address**



Problems

- Packets must be sent periodically so that the NAT table entry does not surpass its TTL
- Have to coordinate creating the connection somehow
- Uses UDP – fast but unreliable
- Doesn't work with all NAT translation methods...

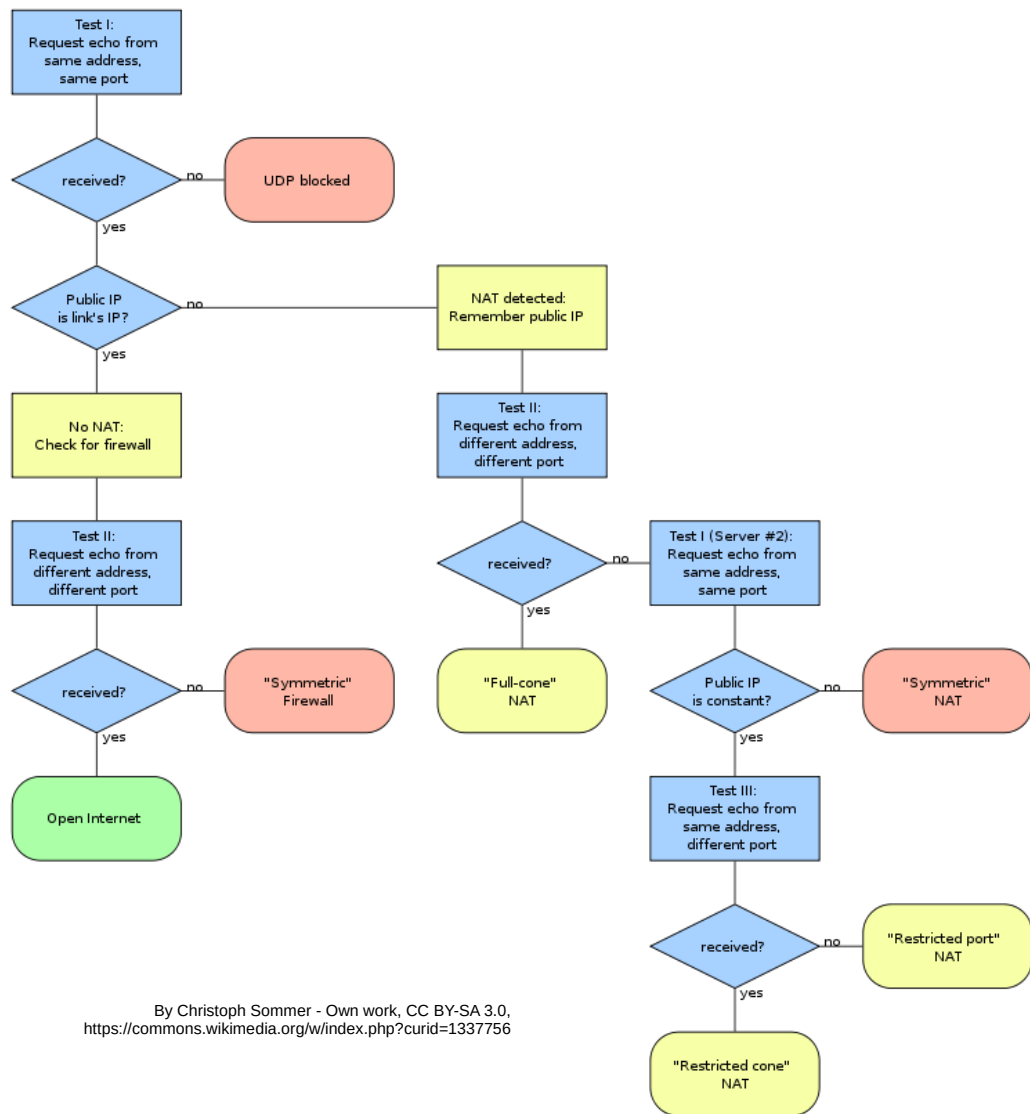
NAT translation methods

- Full cone
 - Addresses are mapped predictably
- (Address-)restricted cone
 - Addresses are mapped predictably
 - Only accept packets from IPs to which I have already sent a packet
- Port-restricted cone
 - Addresses are mapped predictably
 - Only accept packets from sockets (IP+port) to which I have already sent a packet
- Symmetric
 - Addresses are mapped **differently** depending on the destination
 - **Not possible** to do UDP hole punching on these networks.

NAT translation methods

- NAT translation method can be determined using this flowchart
- pystun3 (a pip package) can do this (<https://github.com/talkiq/pystun3>)

```
[~]$ pystun3
NAT Type: Restrict NAT
External IP: XXXXXXXXXX
External Port: 54320
Press any key to continue
[~]$
```



WebRTC – a protocol that basically does all of this

- “Web Real-Time Communication”
- Built into all modern browsers
- Makes requests to STUN servers
- Does UDP hole punching
- Can create connections through relay servers instead if a peer connection is not possible
- Can make ordered UDP connections
- Multiple uses (in addition to creating a standard peer-to-peer socket)
 - Audio and video conferencing, screen sharing, file exchange...



WebRTC – how it works

- Peer A wants to initiate a peer connection to peer B
- Peer A creates a **Session Description Profile (SDP)**
 - Contains information about codecs used, etc
- This SDP is the **offer**
- Peer A sends the offer to peer B (via a third party)
- Peer B creates an SDP **answer** and sends it back to peer A
- Peers A and B also exchange **Internet Connectivity Establishment (ICE) candidates**
 - Can refer to open ports on the local network, hole-punched UDP sockets, relay servers...
- The ICE candidates are tested in priority order until a connection is made
 - Local connections first, then UDP hole punching, then relays

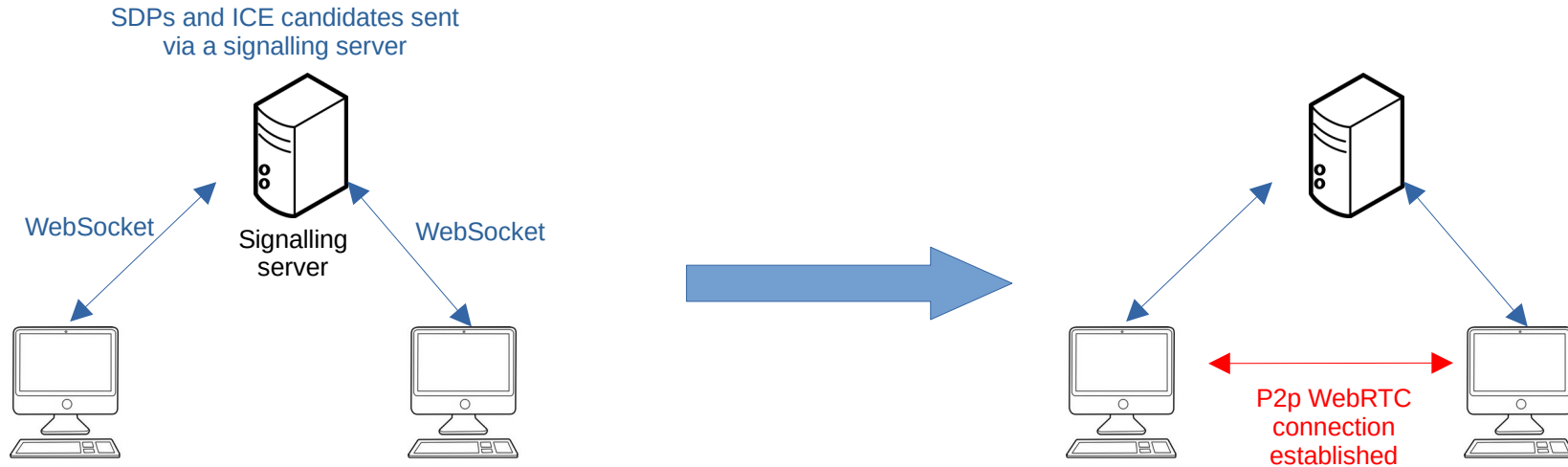
Sample SDP for a Multicast Flow

```
v=0
o=- 123456 123458 IN IP4 10.0.1.2
s=My sample flow
i=4 channels: c1, c2, c3, c4
t=0 0
a=recvonly
m=audio 5004 RTP/AVP 98
c=IN IP4 239.69.11.44/32
a=rtpmap:98 L24/48000/4
a=ptime:1
a=ts-refclk:ptp=IEEE1588-2008:00-11-22-FF-FE-33-44-55:0
a=mediaclock:direct=0
```

https://dev.audinate.com/GA/ddm/userguide/1.1/webhelp/content/appendix/sample_sdp_specification.htm

Signalling

- Before the peer connection is created the SDPs and ICE candidates must be exchanged
- This process is called **signalling**
- Typically done using an external server to forward the data using WebSockets



- WebRTC does not define any specific protocol for signalling