

VOIP with NATs and Firewalls

Robert Sparks

Protocol Barbarian

Estacado Systems

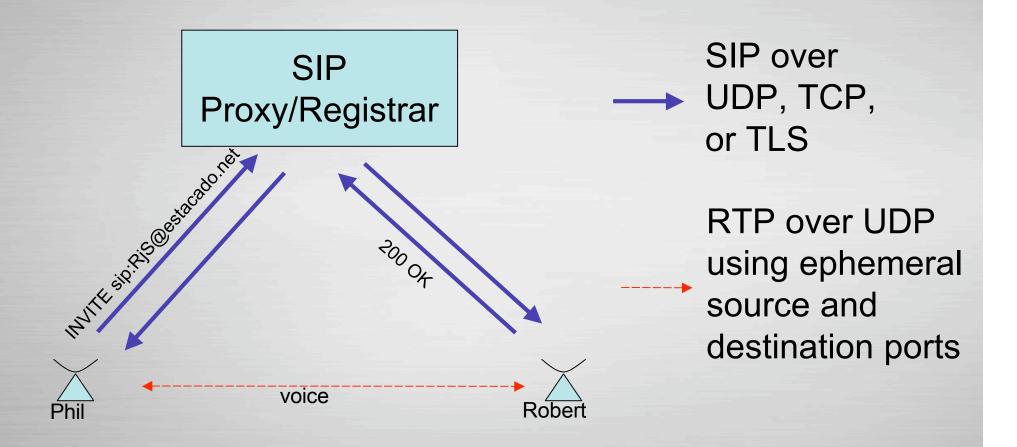
Voice (and more) over IP



- Signaling: Session Initiation Protocol (SIP)
 - Looks, but does not act, like HTTP
 - Defined over a variety of transports
 - Used for rendezvous (helping endpoints find each other
 - Used to negotiate media (addresses, formats)
- Media: Real-time Transport Protocol (RTP)
 - Carries media (voice, video, other)
 - Represented via standardized codecs
 - Internally sequenced and timestamped

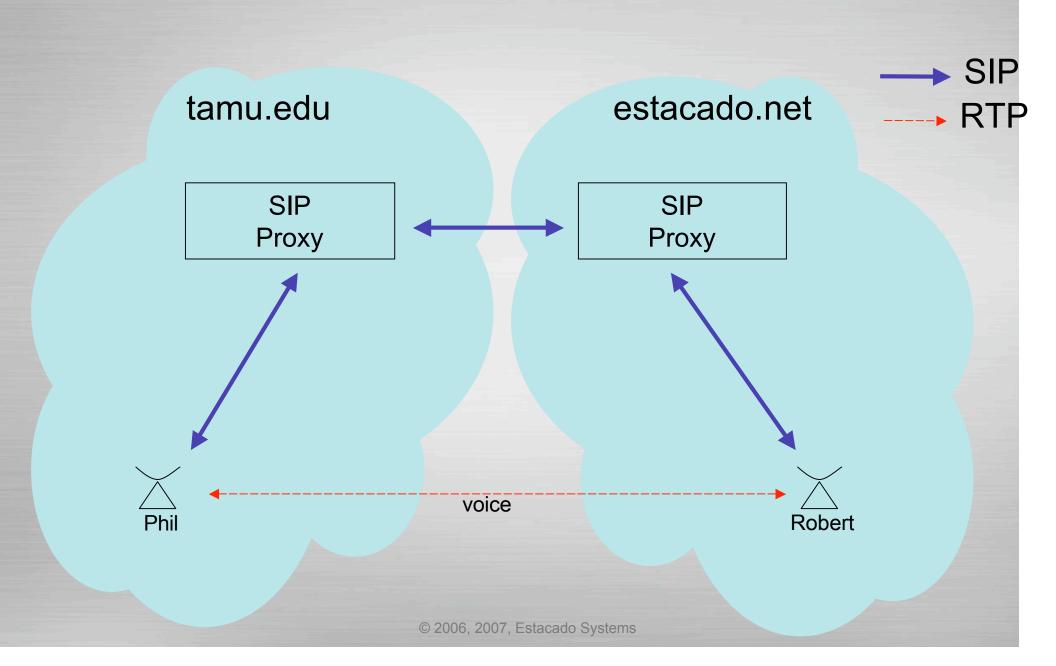
Basic Architectural Model





Trapezoid Model



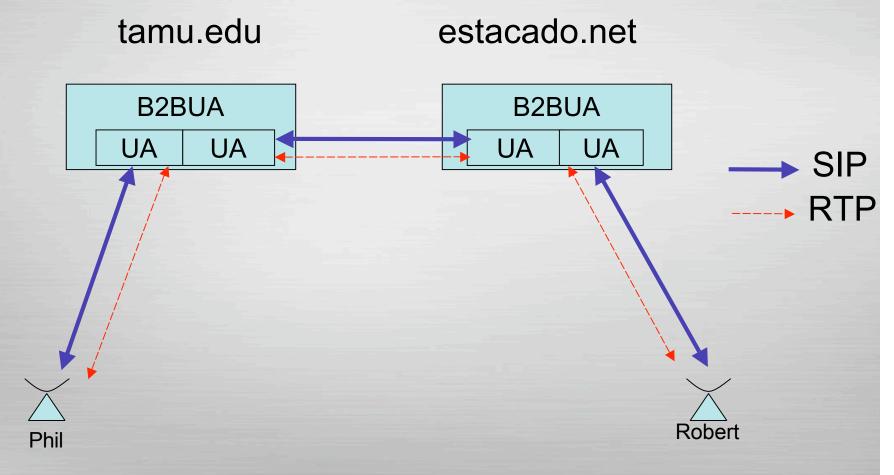


Bending the Architectures



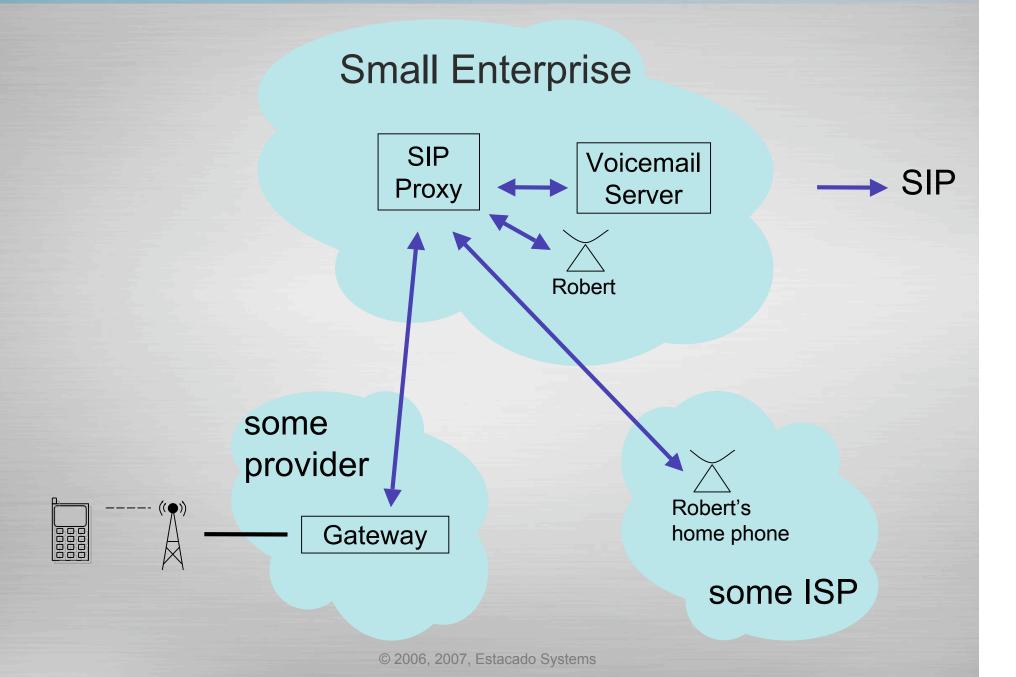
Back-to-Back User Agents

(also called Session Border Controllers or SBCs)



Example Deployment





The Problem



- SIP and RTP were originally designed for an end-to-end transparent network
- NATs, Firewalls, and other elements
 (sometimes even SBCs) violate that
 assumption to the point that SIP or media fails

The Problem



NATs

- Change the apparent source address, and sometimes port, of packets
 - SIP puts these addresses in the IP packet payload, where NATs can't "fix" them
 - Via, Contact, SDP c= lines
- Prevent incoming TCP connections
- Prevent incoming UDP unless you've sent traffic establishing a binding
 - Many different types of binding behavior

The Problem



Firewalls

- Tend to prevent all incoming traffic
- Sometimes allow "pinholes"
 - no standard way (yet) to manage them
 - tend to close them without warning or notice

SBCs

- Have to be explicitly aware of any new places a protocol might need to be "fixed"
 - Tend not to forward any bits they haven't been explicitly told to forward
 - Hinders deployment of new features

The Tools



- rport, received, symmetric rtp
 - Change protocol behaviors to avoid NAT damage
- STUN
 - Allows a client to discover what it looks like on the outside
- TURN
 - Reflects packets at globally reachable location
- outbound
 - "Nails up" a connection to something others can reach
- ICE
 - Allows endpoints to discover which of several alternative network traversal strategies work for each call

Within SIP



- Any SIP element receiving a request remembers (by marking the message) the IP address the request appeared to come from
 - UDP: responses will go back to that address
 - TCP: responses go back over the connection the request arrived on
 - But if the connection is gone, the UAS may attempt to open a new connection to that address (will almost never work if there's a NAT)

Within SIP (rport extension)



- The rport extension provides for ports what received provides for addresses
 - Requires support from both elements at each hop
 - Receiver remembers the port a request appeared to come from
 - Over UPD, response goes back to that port
 - Over TCP, response goes back to the connection the request arrived over

Media tools

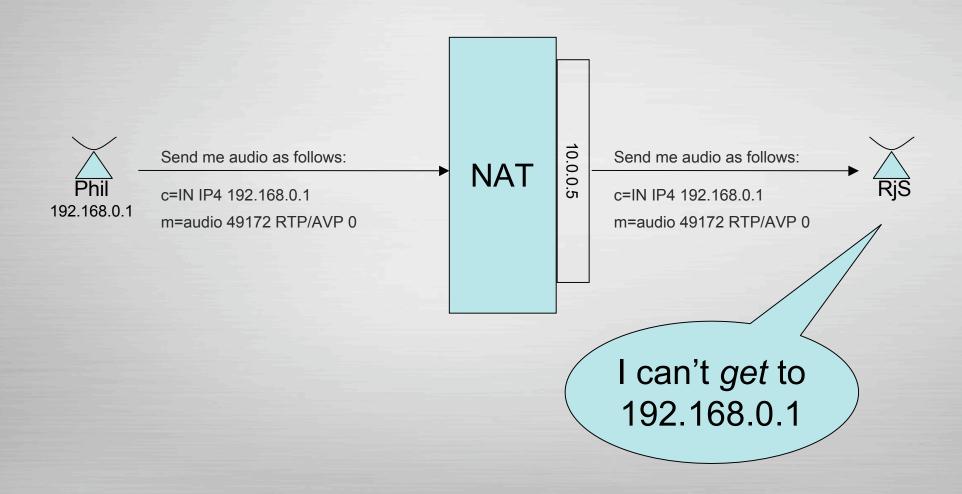


Symmetric RTP

- Sending media packets from the same port you have agreed to receive media improves the likelihood of traversing certain NATs
- This behavior is just done, not signaled

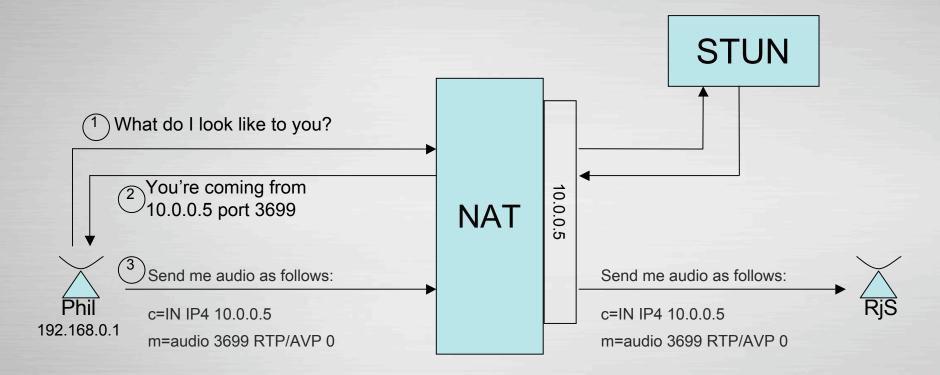


SDP offers contain an address for receiving media



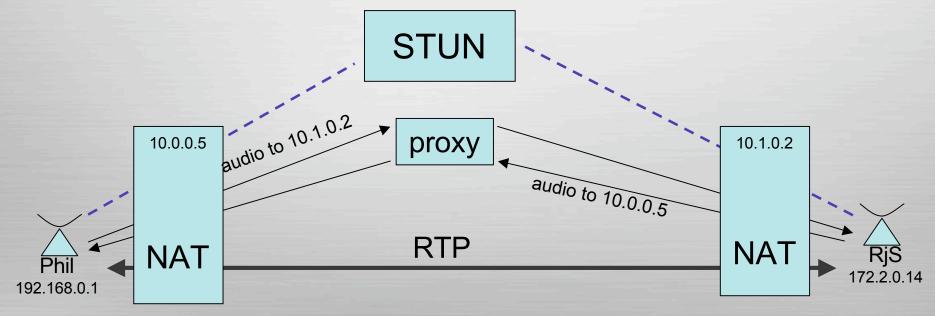


STUN lets Phil discover what his address looks like to RjS



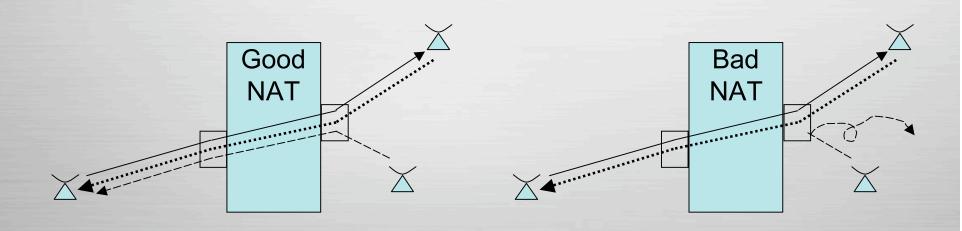


- Place the discovered address in
 - The SDP c= line
 - The Contact URI (if a domain name isn't appropriate)
- Allows traversal of a huge portion of NATed space
 - Even if both RjS and Phil are behind NATs





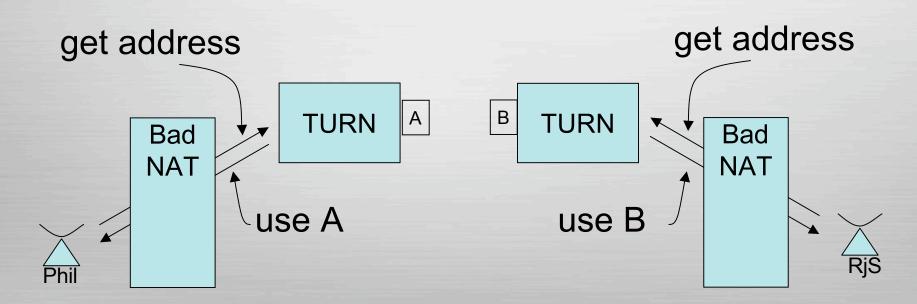
 Doesn't help with NATs that bind so that only the destination of the of the packet creating the binding can send packets back to the source



TURN



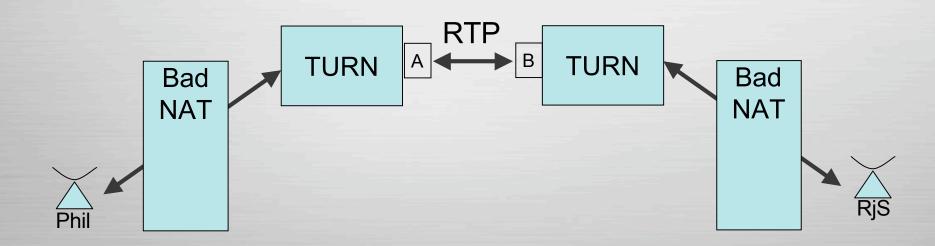
- Traversal using Relay NAT
 - Allows a client to request an address on a public interface and have media relayed to and from that address



TURN



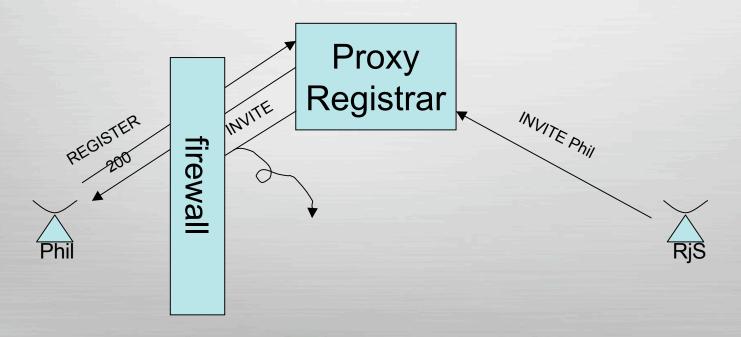
- Phil offers to receive media at A
- Robert answers asking to receive media at B



Outbound



- Clients behind some NATs and most firewalls can't accept a TCP connection or receive a UDP packet from a host they haven't sent to
 - Nobody can call Phil



Outbound

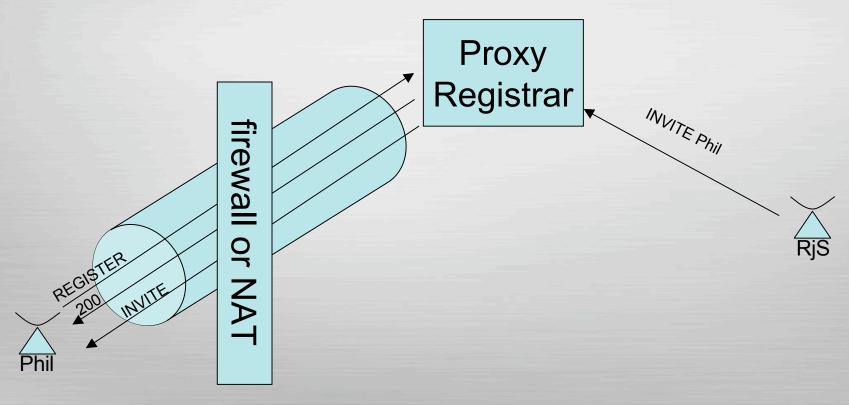


- The outbound extension "nails up" a connection, or flow, between Phil and his proxy-registrar
 - Can be UDP or TCP (or anything else that carries SIP)
 - The proxy agrees to send all traffic for Phil down the outbound connection

Outbound



 Phil keeps his connection alive by periodically exchanging traffic with the proxy (STUN for UDP, CRLF for TCP)





- Phil may have many addresses to use as alternatives for media
 - Native interface address
 - VPN address
 - STUN discovered address
 - addresses acquired using TURN
- One might work with Ben, but only a different one with Robert



- The Interactive Connectivity Establishment
 Framework (ICE) allows Phil to
 - Offer all the addresses he has to Ben and Robert
 - Test the addresses they give back to see which one works the best



- Phil places all his addresses as alternatives in an SDP offer, ordered by preference
 - A direct connection is much better than one using a TURN relay
- Ben and Robert return all their alternatives in their answers
- Everyone starts testing the alternatives (ordered by preference) by trying STUN requests



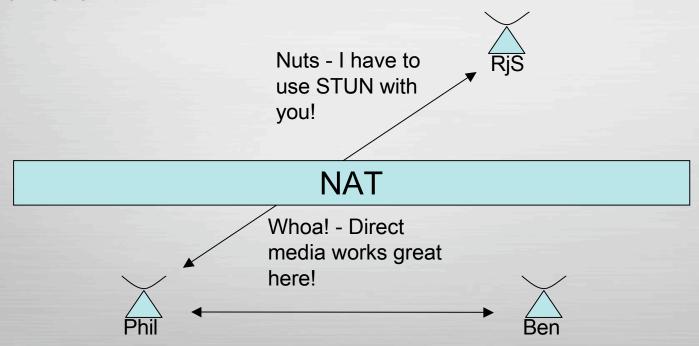
- Each alternative starts as a candidate, prioritized by the requested preference
- A successful STUN transaction between a local and remote candidate makes the pair "valid"
 - This transaction may expose a new address that should be considered as a candidate
- ICE can stop as soon as there is a valid pair for each media stream
 - It's legal to keep trying to find a better pair



- There may be multiple valid pairs for a stream
- The ICE "controlling" endpoint indicates which pair to use for each stream by sending a STUN request with a nominating flag
 - The "controlling" endpoint is almost always whoever sent the offer
- ICE stops when all streams have a nominated pair
- ICE can be restarted for any stream at any time by issuing a new offer (changing certain media stream level attributes)



- Phil can place a single request
 - that forks to Robert and Ben
 - either of which (or both) can answer and have media work, even though their address requirements are wildly different





Robert Sparks

mailto:RjS@estacado.net

mailto:RjS@nostrum.com

sip:RjS@estacado.net