Peer-to-Peer Systems and Applications



Lecture 10: Selected Topics II

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0. Lecture Overview



1. Kademlia

- 1. Routing Concept
- 2. Protocol
- 3. Routing Table Construction
- 4. Peer Selection Policy
- 5. Enhancements

2. Network Coding

- 1. Motivation
- 2. Theory
- 3. Example
- 4. Application

3. WebRTC

- 1. Introduction
- 2. Architecture
- 3. Example
- 4. Outlook

4. NAT Traversal

- 1. Background
- 2. Relaying
- 3. Hole Punching



1. Kademlia

Routing Concept, Protocol, Routing Table Construction, Peer Selection Policy, Enhancements

1.0 Kademlia



- Use a parallel iterative lookup to locate data
 - Retrieve data faster
 - Overcome faulty nodes
 - \triangleright Usually a = 3
- DHT-based overlay network using the XOR distance metric
 - Simple operation
 - Symmetrical routing paths

$$(A \rightarrow B == B \rightarrow A)$$

- due to $d_{XOR}(A,B) == d_{XOR}(B,A)$
- Store data with key X on k nodes closest to X according to XOR metric
 - "build in" replication ensuring data availability
 - \triangleright Usually k = 20

- Use lookup messages to maintain the overlay network
 - Learn useful routing information from received lookup requests

XOR Distance Calculation:

```
ID Node A: 110101
ID Node B: 010001
```

$$d_{XOR}(A,B) = d(110101,010001)$$

```
1 1 0 1 0 1

XOR

0 1 0 0 0 1

↓

1 0 0 1 0 0
```

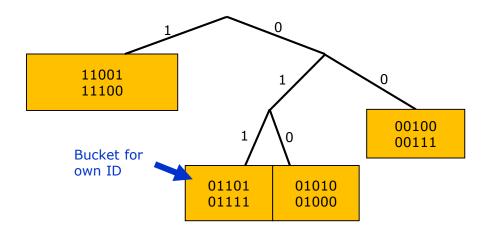
$$d_{XOR}(A,B) = 1 \ 0 \ 0 \ 1 \ 0 \ 0_2 = 36_{10}$$

1.1. Concept of the Kademlia Routing Process



- Structure of routing table
 - Every node maintains a binary tree like routing table
 - Tree branches along the local node ID
 - Every leave is a bucket with k entries

Example routing table for node 01110 with bucket factor k = 2

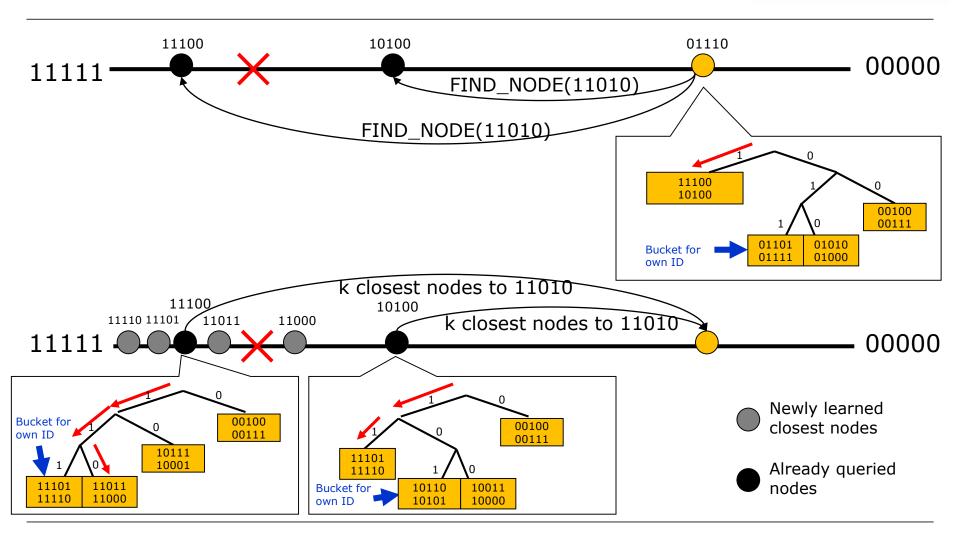


Lookup procedure for a key X

- Step 1: Traverse routing table tree and pick bucket with node IDs closest to key X
- Step 2: Put node IDs in a node list
- Step 3: Send a closest nodes request to nodes in the list closest to X
- Step 4: Put received node IDs in node list
- Step 5: Repeat until Step 3 and 4 until set of k closest node to not change anymore
- Step 6: Pick k-closest nodes from node list and send store or get data request to them

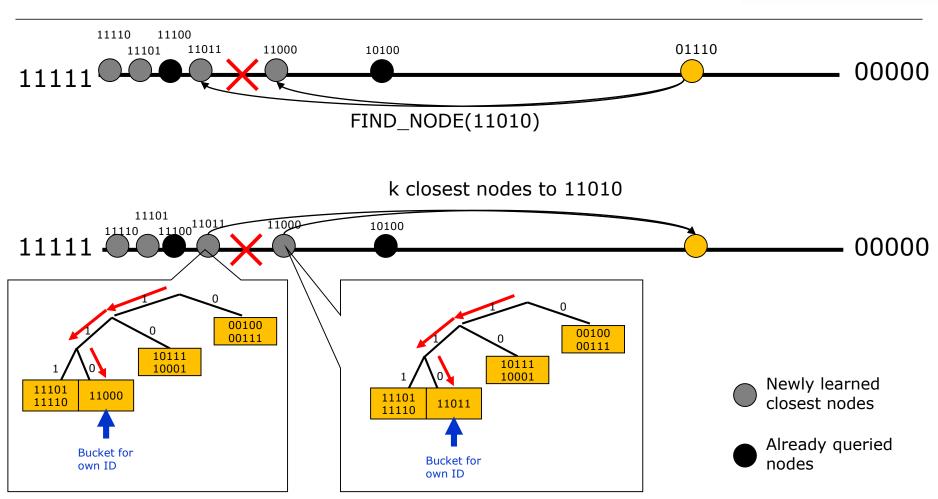
1.1. Routing from 01110 to 11010, a = 2, k = 2





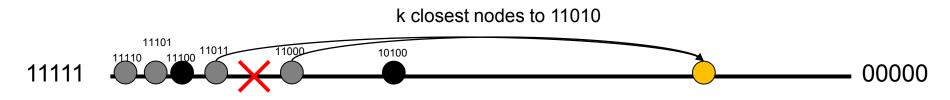
1.1. Routing from 01110 to 11010, a = 2, k = 2



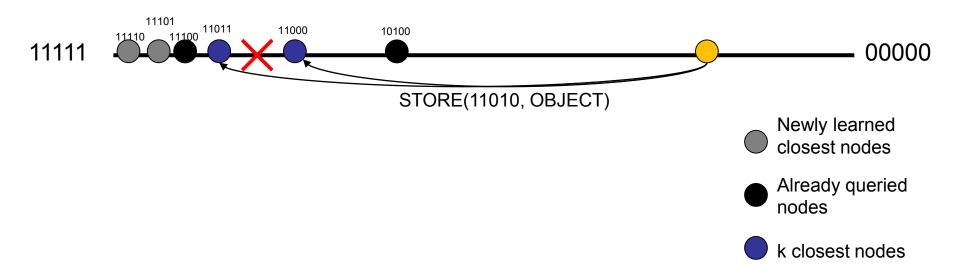


1.1. Routing from 01110 to 11010, a = 2, k = 2





No new nodes discovered → stop lookup process



1.2. Kademlia Protocol



- The Kademlia protocol consists of 4 RPCs:
 - FIND_NODE(KEY):
 - Recipient returns <IP Address, UDP Port, Node ID> triples for k closest nodes he knows about
 - FIND_VALUE(KEY):
 - Like FIND_NODE
 - With exception:
 - If recipient already stores the value, the value is returned instead of k closest nodes
 - > PING(IP ADDRESS)
 - Probes a node to see if it is online
 - STORE(KEY, VALUE)
 - Instructs a node to store a <key, value > pair

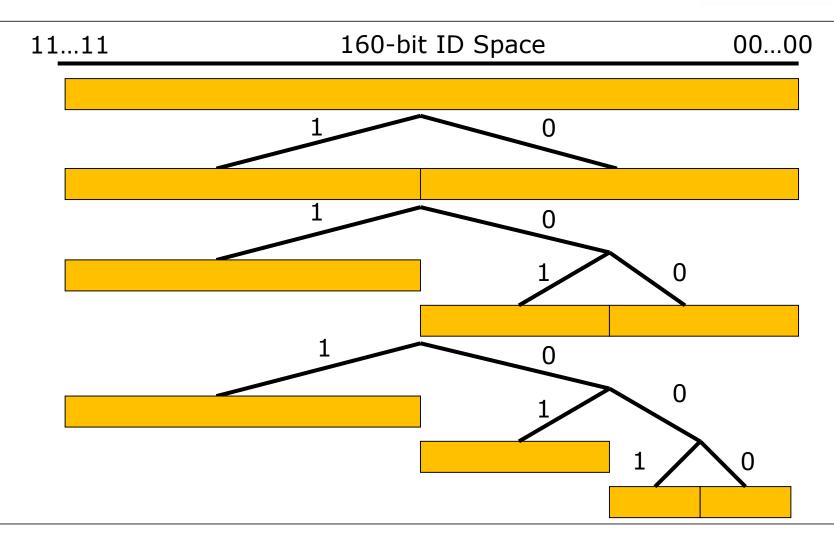
1.3. Construction of Routing Table



- Each node maintains routing table (k buckets)
 - Routing tables of different peers may be different
- ❖ For each 0 <= i < 160 every node</p>
 - keeps a list of <IP Address, UDP Port, Node ID> triples
 - for k nodes within range [2 ^i; 2 ^(i+1)[
 - in total k * 160 contacts
- Nodes learn from
 - messages they receive or
 - using the FIND_NODE method
- Preference towards old contacts
 - Study has shown that the longer a node has been up, the more likely it is to remain up another hour
 - Resistance against DoS attacks by flooding the network with new nodes

1.3. Evolution of the k Buckets







11111 5-bit ID Space 00000

New node 11001:

11001

New node 01101:

11001 01101

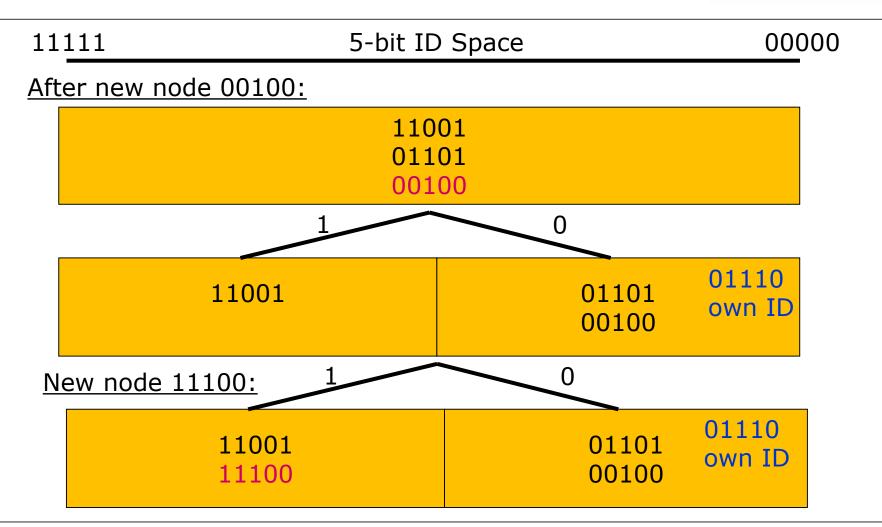
New node 00100:

11001 01101

00100 but ... no because 3^{rd} , and k=2 is max.

k bucket is full → split necessary

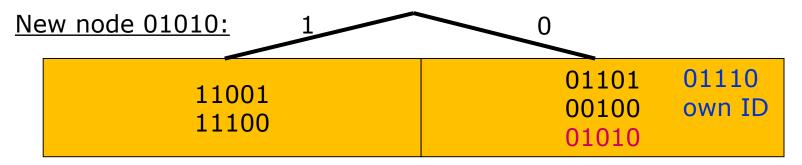






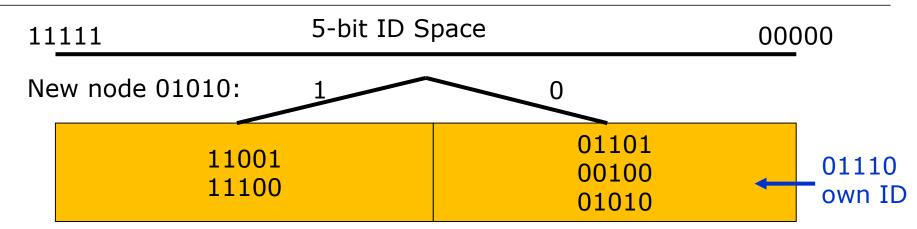
11 <u>1</u>	11 !	5-bit ID Space	00000
New node 11010: 0			
	11001 11100	01101 00100	01110 own ID

Left k bucket full and 11010 NOT in k-bucket-range of 01110 → node is dropped

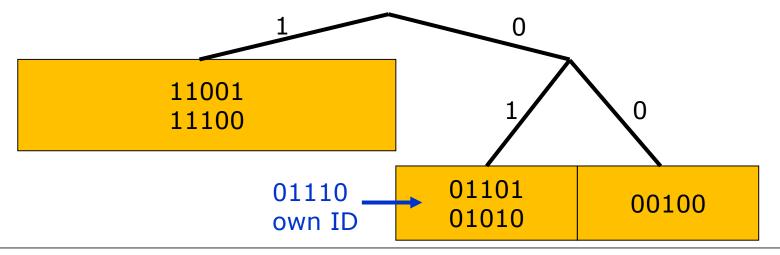


Right k bucket full and 01010 in k-bucket-range of 01110 → split necessary

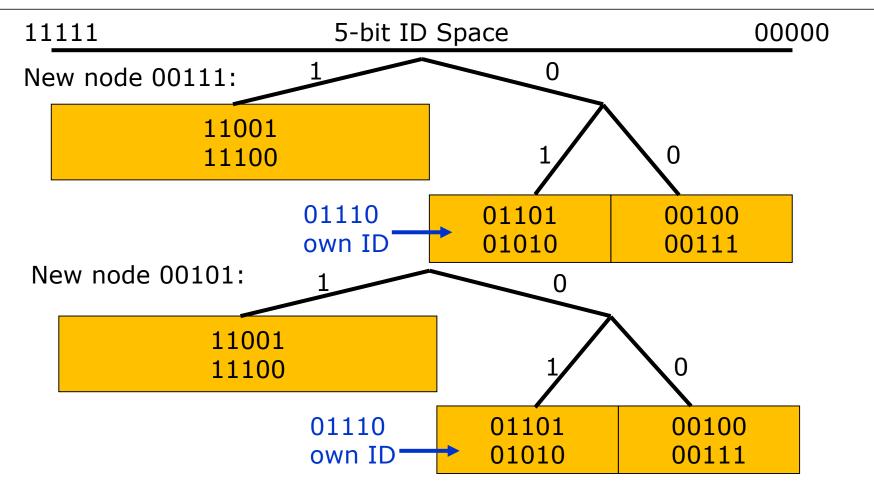




Right K bucket full and 01010 in k-bucket-range of 00110 \rightarrow split necessary

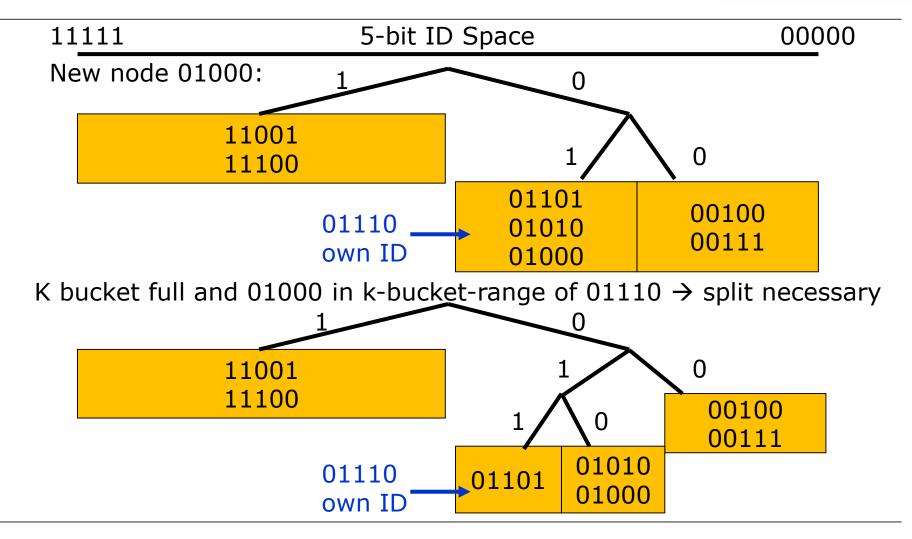






Left K bucket full and 00101 NOT in k-bucket-range of 01110 → node is dropped





1.4. k Buckets: Peer Selection Policy



- If node u receives a message from node v then it adds node v to its k-bucket according to the following rules:
 - IF v is already in a k-bucket
 THEN move v to the tail of the bucket
 - IF v is not in the k-bucket and the bucket has fewer than k entries THEN insert recipient to the tail of list
 - IF the appropriate k bucket is full AND least recently seen node is alive THEN move least recently seen node to tail of bucket and discard node v
 - IF the appropriate k bucket is full AND least recently seen node is <u>not</u> alive THEN remove least recently seen node from bucket and add node v at the tail
- Note: approx. k = 20 in the real world

1.5. Kademlia: Enhancements



- Parallel queries
 - For one query, a (alpha) concurrent lookups are sent
 - More traffic load, but lower response times
- Network maintenance
 - In Chord: active fixing of fingers
 - In Kademlia: learning for bypassing queries
 - Check if peer IDs fit better in routing table
- Large routing tables
 - In Chord: 1 finger per distance 2^i to 2^(i+1)
 - In Kademlia: k contacts per distance 2^i to 2^(i+1)
 - Increased robustness



2. Network Coding

Motivation, Theory, Example, Application

2.1. Network Coding Motivation



- P2P problem area: distribute large data packets to many clients
 - Solution 1: exchange piece/chunk bitmap
 - Request contains the pieces we are looking for, recipient replies with piece or that pieces are not here
 - Peer may also collect piece/chunk bitmaps first
 - Solution 2: use network coding
 - Simple request, when there is nothing new, don't request.
- Random network coding: send linear combination of all received chunks
- → no need to exchange piece/chunk information

2.2. Network Coding Theory



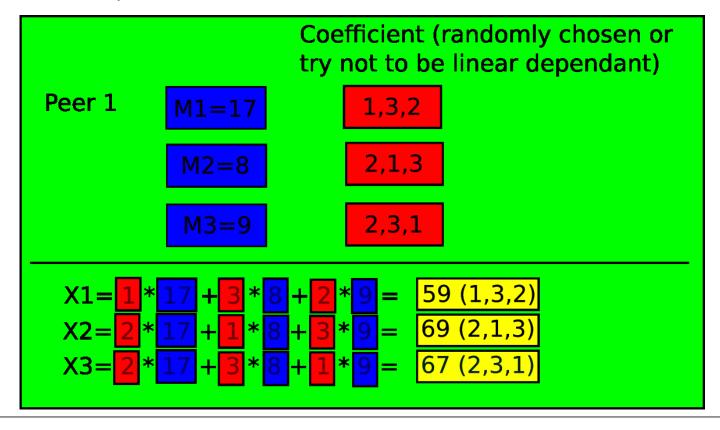
- ❖ M1, ..., Mn → original packets / g1, ..., gn → random coefficients
- \bullet Encoding: $X = \sum_{i=1}^n g_i M^i$
 - Encoding vector g, information vector x
 - > Send x, g
- Re-encoding, h1, ..., hm → random coefficients:

$$X' = \sum_{j=1}^{m} h_j X^j$$

- ightharpoonup Encoding vector ${}_{ ext{h} o ext{g'}}$ $g_i' = \sum_{j=1}^m h_j g_i^j$
- Information vector x'
- > Send x', g'

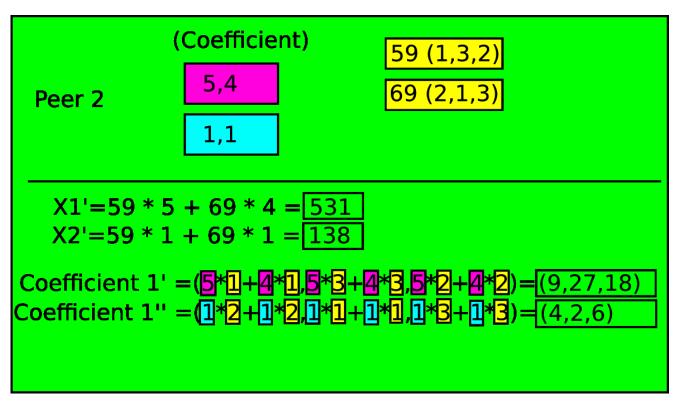


Peer 1 chooses coefficients (encoding vector), calculates linear combination, sends two to Peer 2 and one to Peer 3



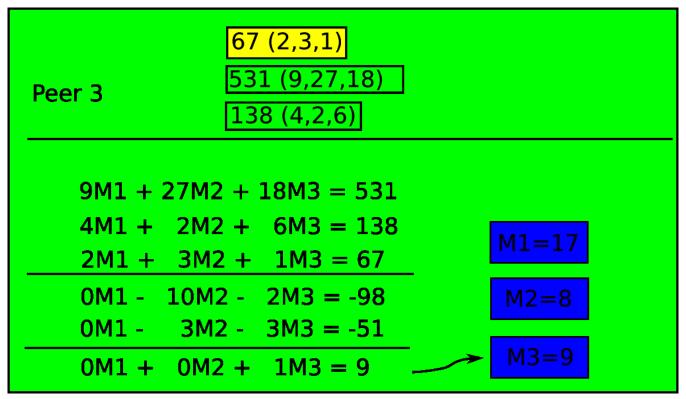


Peer 2 also has random coefficients, gets two linear comb. from Peer 1, creates new linear comb. (adapt coeff.), send to Peer 3





Peer 3 gets all 3 linear comb. with coding vector, does Gauss elimination (linear system with 3 equations & 3 unknowns) [1]



[1] http://www.gregthatcher.com/Mathematics/GaussJordan.aspx



Peer 3 gets other linear comb. with coding vector, does Gauss elimination (linear system with 3 equations & 3 unknowns) [1]

[1] http://www.gregthatcher.com/Mathematics/GaussJordan.aspx

2.4. Network Coding Application



- Send packets as long as linear independant
 - No exchange of bitmap
- Random coefficients: low probability of collision
- Overhead of transmitting the encoding vectors is small, size of block is Kbytes
 - \triangleright CPU overhead O(n³), (n = number of blocks)
- Avalanche: File Swarming with Network Coding [1]
 - Network Coding improves robustness and throughput

[1] http://research.microsoft.com/en-us/projects/avalanche/



3. WebRTC

Introduction, Architecture, Example, Outlook

3.1. WebRTC - Introduction (1)



- WebRTC for browser to browser communication
 - P2P, no server involved (~mostly)
- Google bought in 2010 GIPS and open sourced WebRTC



- Protocol standardized by IETF (codec requirements, media protocol), JavaScript API by W3C
- Supported by Chrome, Firefox (and others)
- Compatibility [1]
 - WebRTC support since Chrome 26+, Firefox 23+
 - SCTP: supported by Firefox, Chrome 31+
 - Binary data: supported by Firefox, Chrome 31+

[1] http://peerjs.com/status

3.1. WebRTC – Introduction (2)



- Filling gap in the Web-Experience
 - Video Chat

 - Customer Service
 - Online Games
 - Real-time Feeds
 - File Sharing

- → Google Hangouts Plugin, Flash, Java
- Multimedia / Confernces → Expensive, proprietary 3rd party apps
 - → Chat only, 3rd party plugins/apps
 - \rightarrow Flash
 - → Proprietary software
 - → Requires Server / BitTorrent
- WebRTC widely deployed, no client necessary!

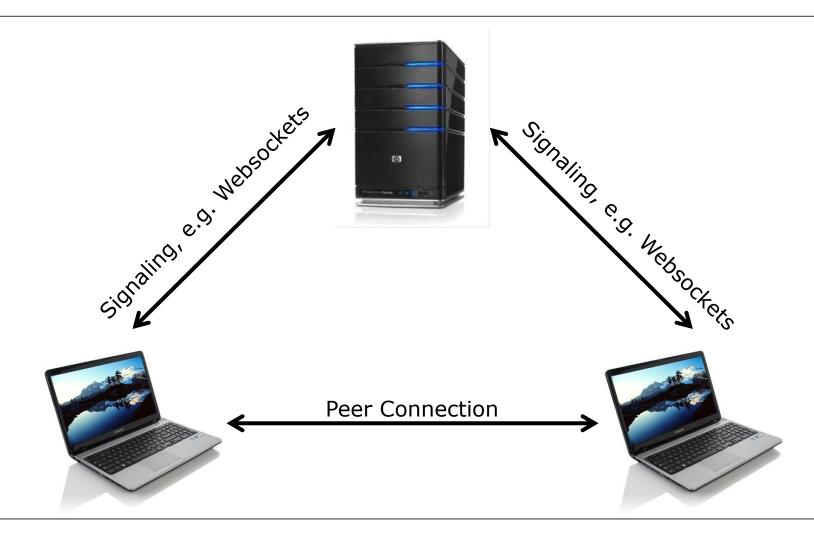
3.1. WebRTC – Introduction (2)



- Developer does not need to care about NAT
 - Abstraction, using STUN, ICE, TURN
 - STUN: session traversal utilities for NAT (detect which kind of NAT)
 - TURN: traversal using relays around NAT (relay)
 - ICE: interactive connectivity establishment, uses STUN and TURN
 - UPnP / NAT-PMP setup by the browser optional? [1]
 - Bugzilla@Mozilla Bug 860045 [2]
- Once connection is established easy API
 - sendChannel.send("hallo")
 - sendChannel.onmessage = function ...
- Mandatory AES encryption
 - SRTP for Media, DTLS for Data, HTTPS for Signalling
 - [1] http://tools.ietf.org/html/draft-kaplan-rtcweb-api-reqs-00
 - [2] https://bugzilla.mozilla.org/show_bug.cgi?id=860045

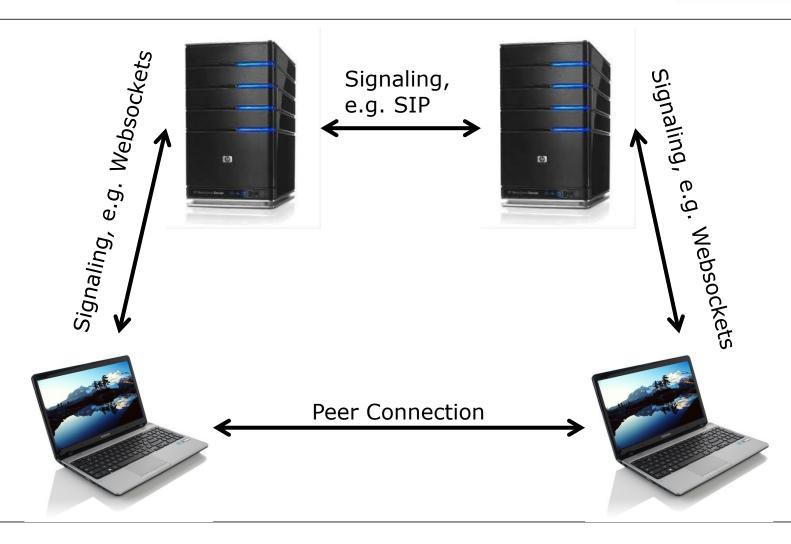
3.2. WebRTC Architecture - Triangle





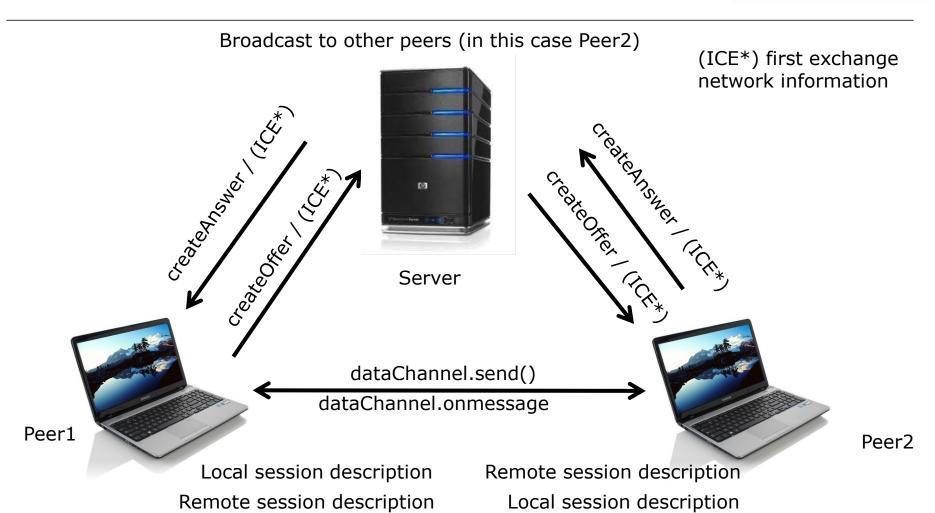
3.2. WebRTC Architecture - Trapezoid





3.3. WebRTC Architecture - Example





3.4. WebRTC - Outlook



- Examples:
 - WebRTC Plugin-free Screen Sharing [1], RTCPeerConnection Simple Demo [2]
- Strong focus on VoIP
 - Skype competitor?
 - Microsoft / IE and WebRTC?
 - SDP / signaling overhead if using with raw P2P data
- Fewer plugins (flash, java), fewer registrations
- Mandatory Codecs? [3]
 - Video codec in JavaScript [4]
- [3] http://gigaom.com/2013/10/30/google-sticks-with-vp8-opposes-ciscos-push-for-h-264/[4] https://brendaneich.com/2013/05/today-i-saw-the-future/

[2] https://www.webrtc-experiment.com/demos/client-side.html

[1] https://www.webrtc-experiment.com/

Pluginfree-Screen-Sharing/#CA134CZY-1FQD7VI

- Web-based P2P frameworks
 - http://peerjs.com make the API simpler
- New types of applications Conferencing, gaming, P2P file sharing
 - PeerCDN [5], serve static content from browsers of other visitors

[5] http://peercdn.com/



4. NAT Traversal

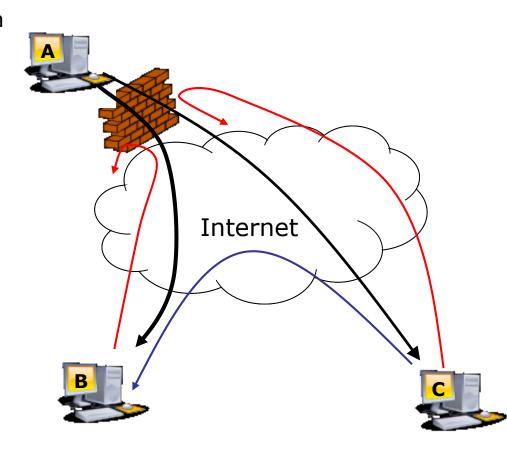
Background, Relaying, Hole Punching

4.0. NAT Traversal



- Nodes must be able to contact each other
- Does not work for peers hidden behind NAT boxes and firewalls
 - Peers cannot be contacted arbitrary
- In real systems often 60-80% of peers are hidden
 - Less available resources
 - Smaller user base

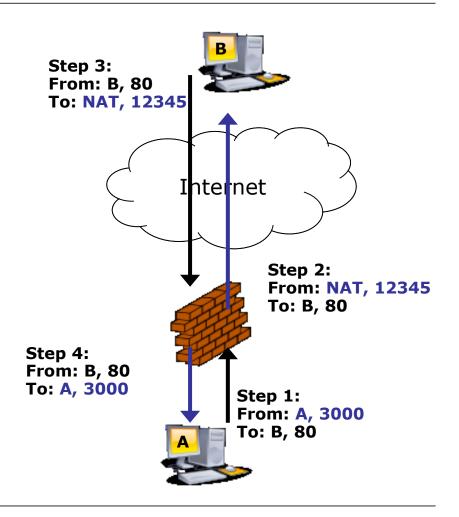
→ P2P applications have to deal with NAT issues, but often this is done only later, as low priority, cf. Skype



4.1. Background: NAT



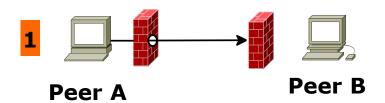
- Reuse one IP address for many hosts
 - Solution for sparse IPv4 address space
 - Security features
- Functionality
 - Address rewriting in IP header
 - Works only if hidden host initiates the communication!
 - Applies for Client-Server
 - Not for P2P!

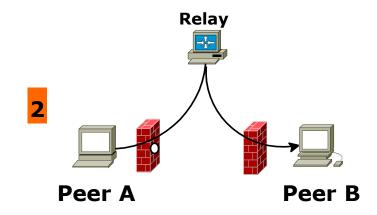


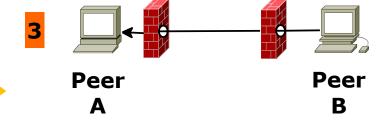
4.1. Background: NAT Traversal



- Port-Forwarding
 - Static mappings in the NAT device
- Universal Plug and Play (UPnP)
- Application Layer Gateways (ALGs)
- Connection Reversal
 - Let hidden host initiate the connection
- Relaying
 - Forward data through publicly accessible hosts
- Hole Punching
 - Exploit NAT functionality
 - Easy for UDP, hard for TCP ...

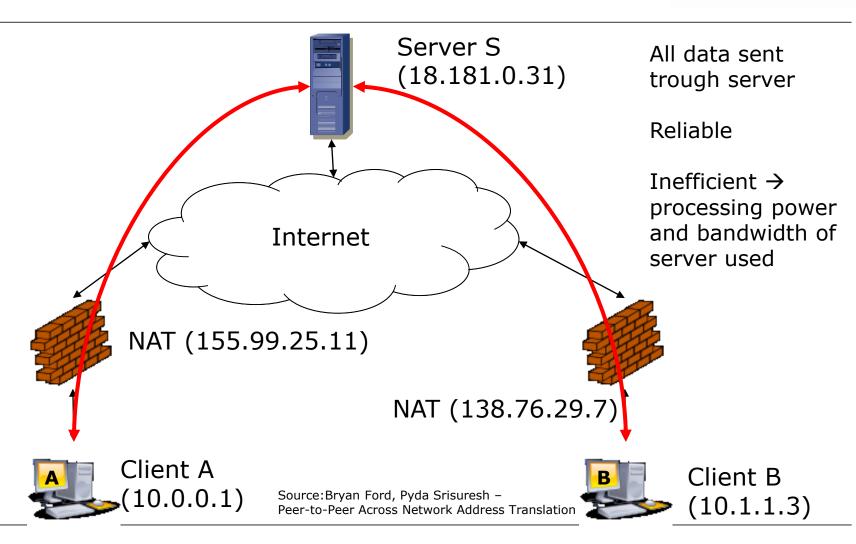






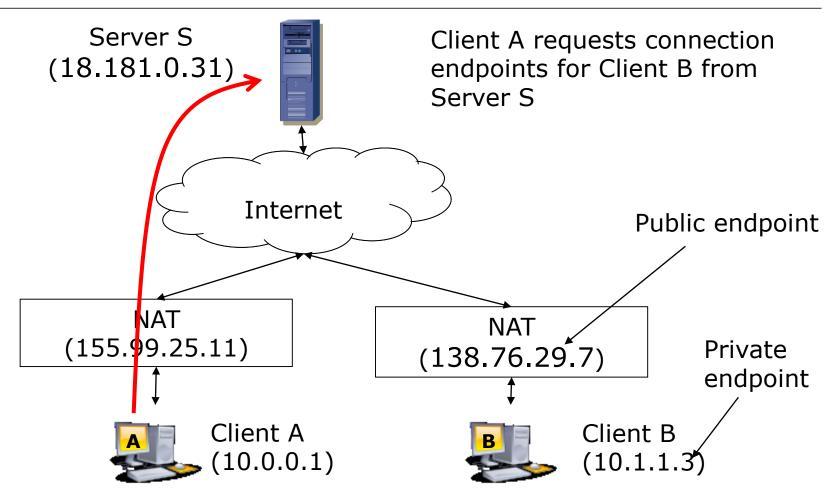
4.2. Relaying





4.3. UDP Hole Punching – First Step

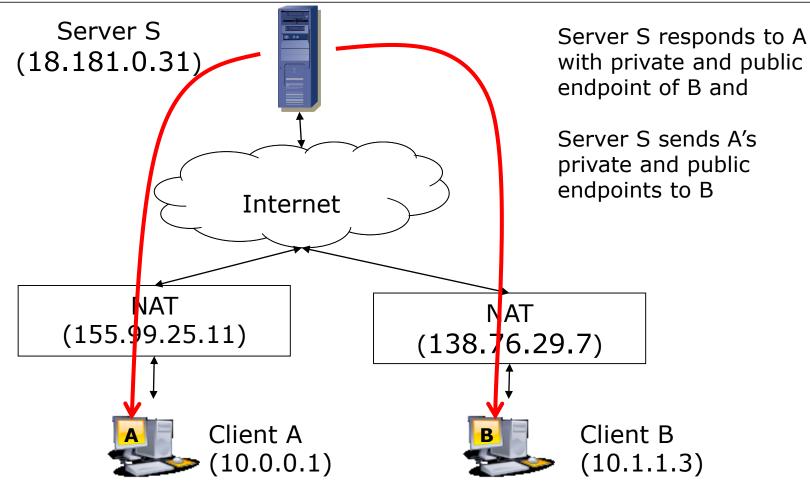




Source: Bryan Ford, Pyda Srisuresh - Peer-to-Peer Across Network Address Translation

4.3. UDP Hole Punching – Second Step

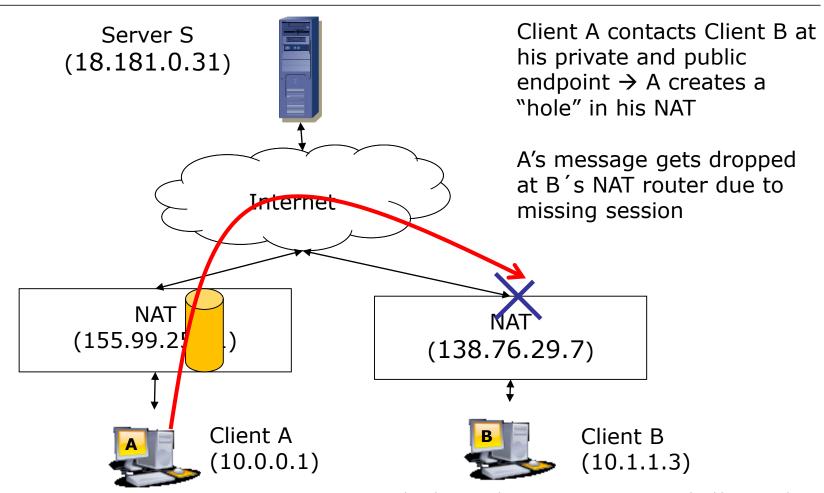




Source: Bryan Ford, Pyda Srisuresh - Peer-to-Peer Across Network Address Translation

4.3. UDP Hole Punching – Third Step

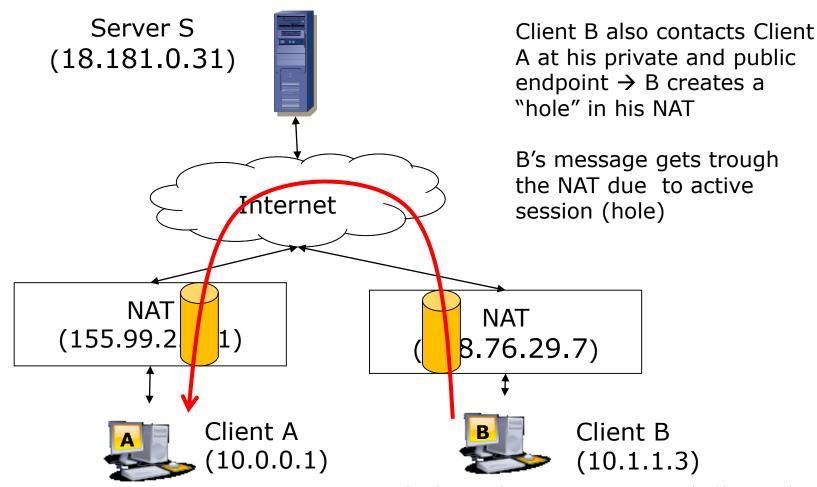




Source:Bryan Ford, Pyda Srisuresh – Peer-to-Peer Across Network Address Translation

4.3. UDP Hole Punching – Fourth Step

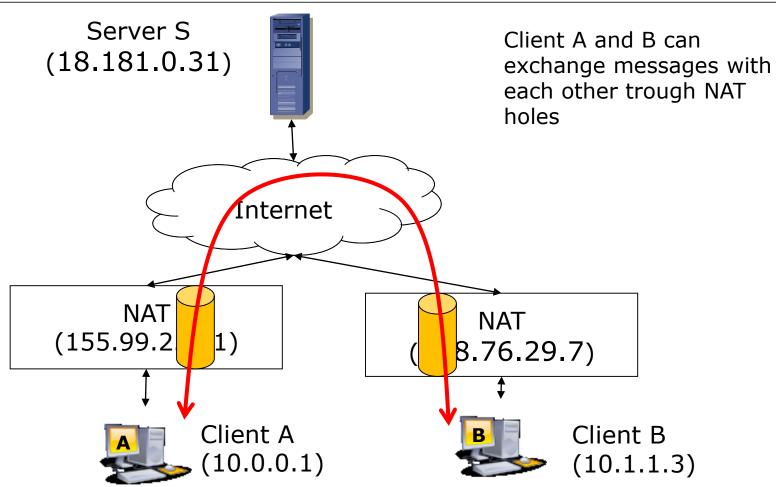




Source:Bryan Ford, Pyda Srisuresh – Peer-to-Peer Across Network Address Translation

4.3. UDP Hole Punching – Fifth Step



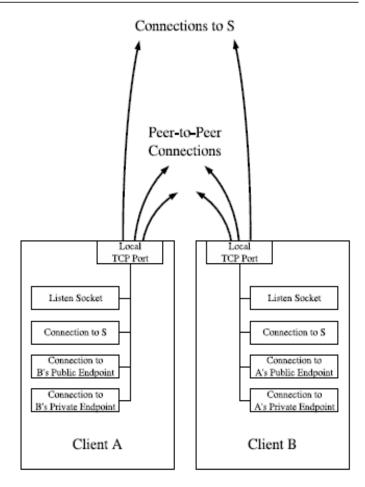


Source:Bryan Ford, Pyda Srisuresh - Peer-to-Peer Across Network Address Translation

4.3. Hole Punching



- UDP-NAT holes are closed after specific life time
 - Resending packets necessary
 - Re-opening hole on-demand
- Hole Punching also works for TCP
 - TCP socket option SO_REUSEADDR option necessary in order to bind multiple sockets to the same local port
- Success of hole punching cannot be guaranteed
 - Different NAT vendors
 - Symmetric NAT vs. cone NAT
 - Drop SYN packet vs. TCP RST packet



Source: Bryan Ford, Pyda Srisuresh - Peer-to-Peer Across Network Address Translation