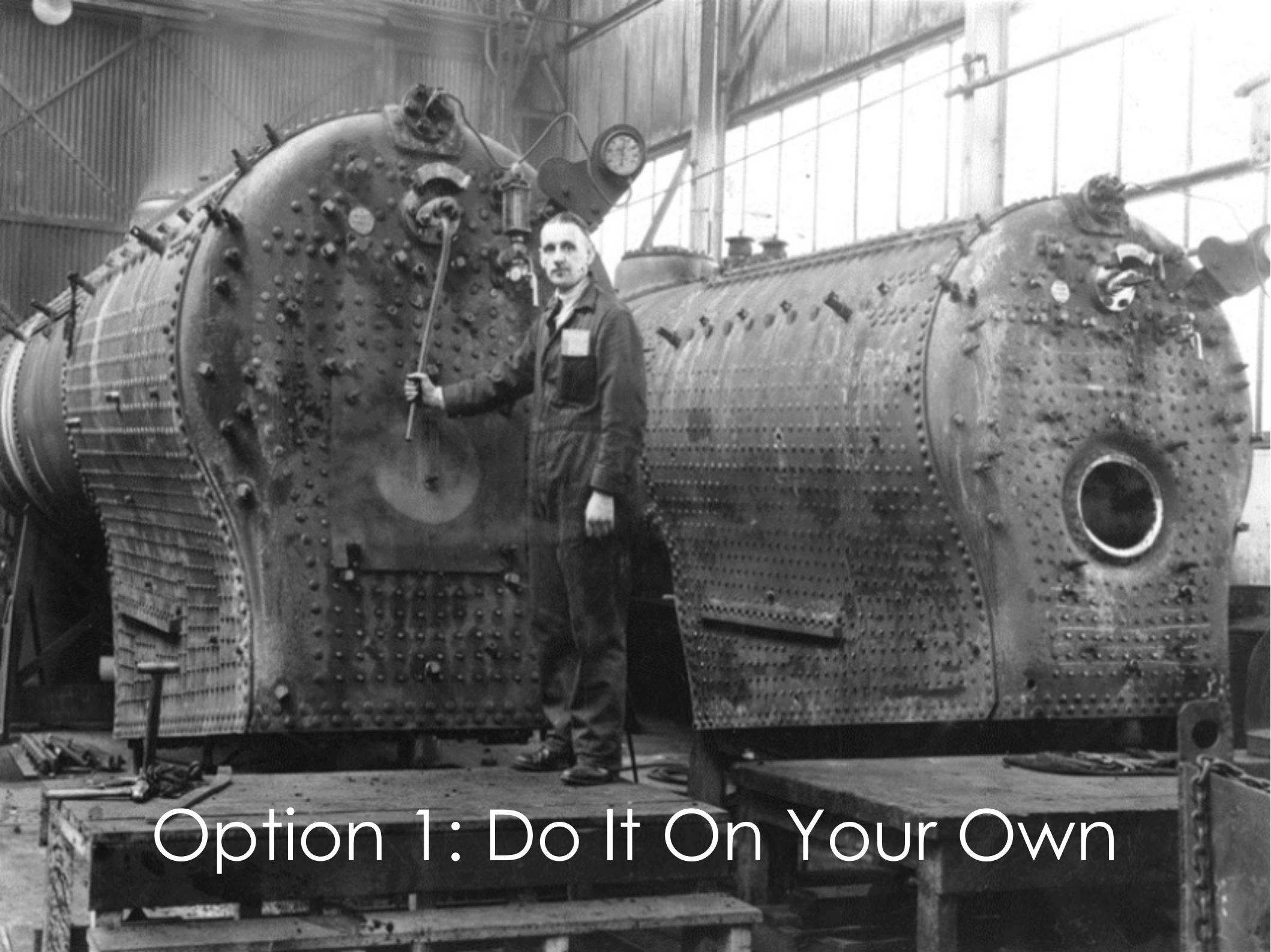


# Towards a server-less web

## Interconnecting web browsers using WebRTC



Option 1: Do It On Your Own



DRIVE  
THRU



Option 2: Let Someone Else Do It

# Migration: Addressing Problem

Oops!  
You found a  
Dead Link



# Take back Control





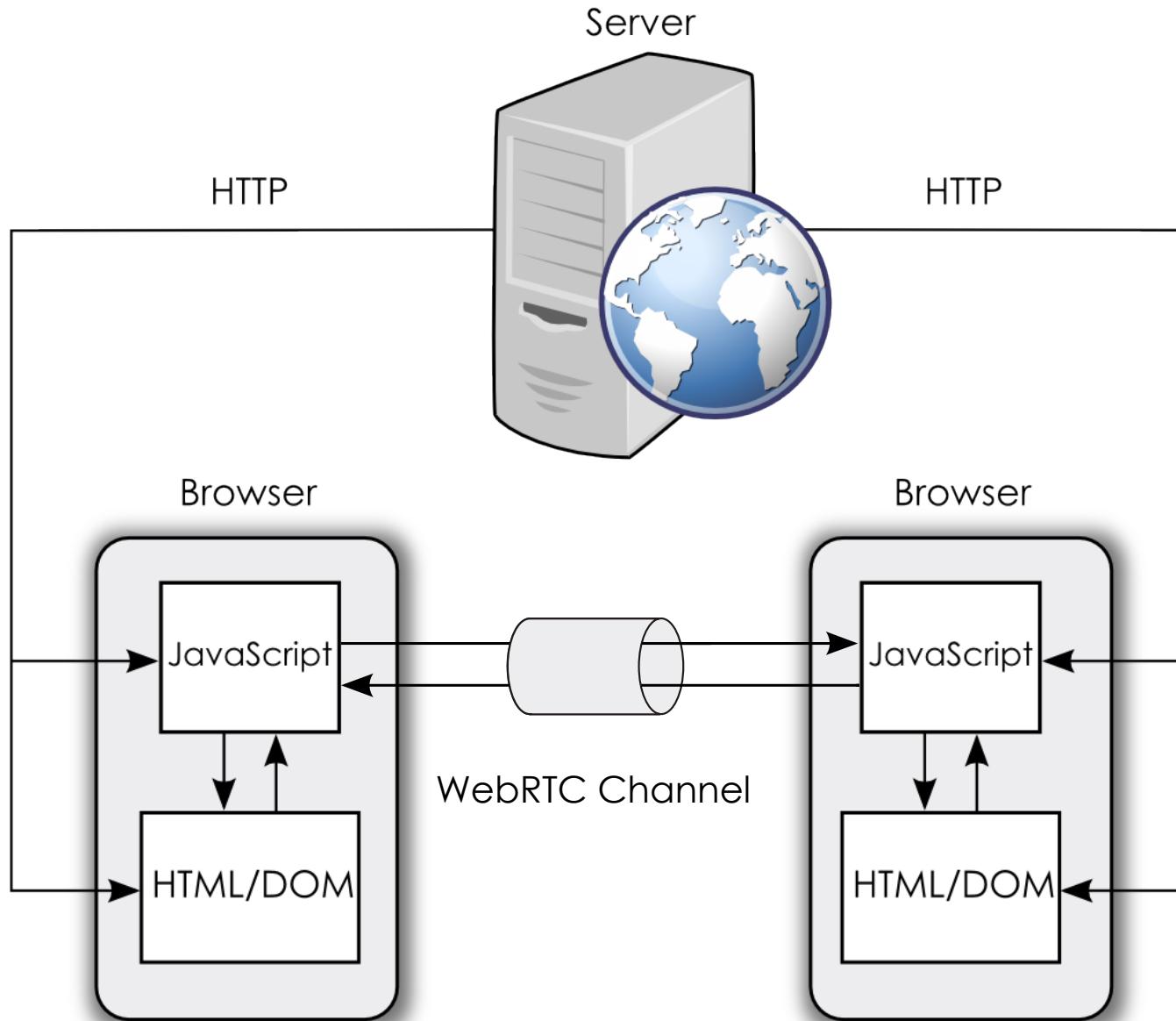
# What is WebRTC?

WebRTC is a free, open **project** that provides browsers and mobile applications with **Real-Time Communications** (RTC) capabilities via simple APIs.

WebRTC (Web Real-Time Communication) is an **API** definition drafted by the World Wide Web Consortium (W3C) that supports **browser-to-browser applications** for **voice calling, video chat, and P2P file sharing** without the need of either internal or external plugins.

WebRTC is a **communications standard** developed by the W3C in close cooperation with the RTCWeb standard developed by the IETF. RTCWeb functions at a **lower protocol layer**; WebRTC enables the embedding of this functionality in applications and websites. The protocol is commonly used to support **voice or video chat** between peers.

# This is WebRTC



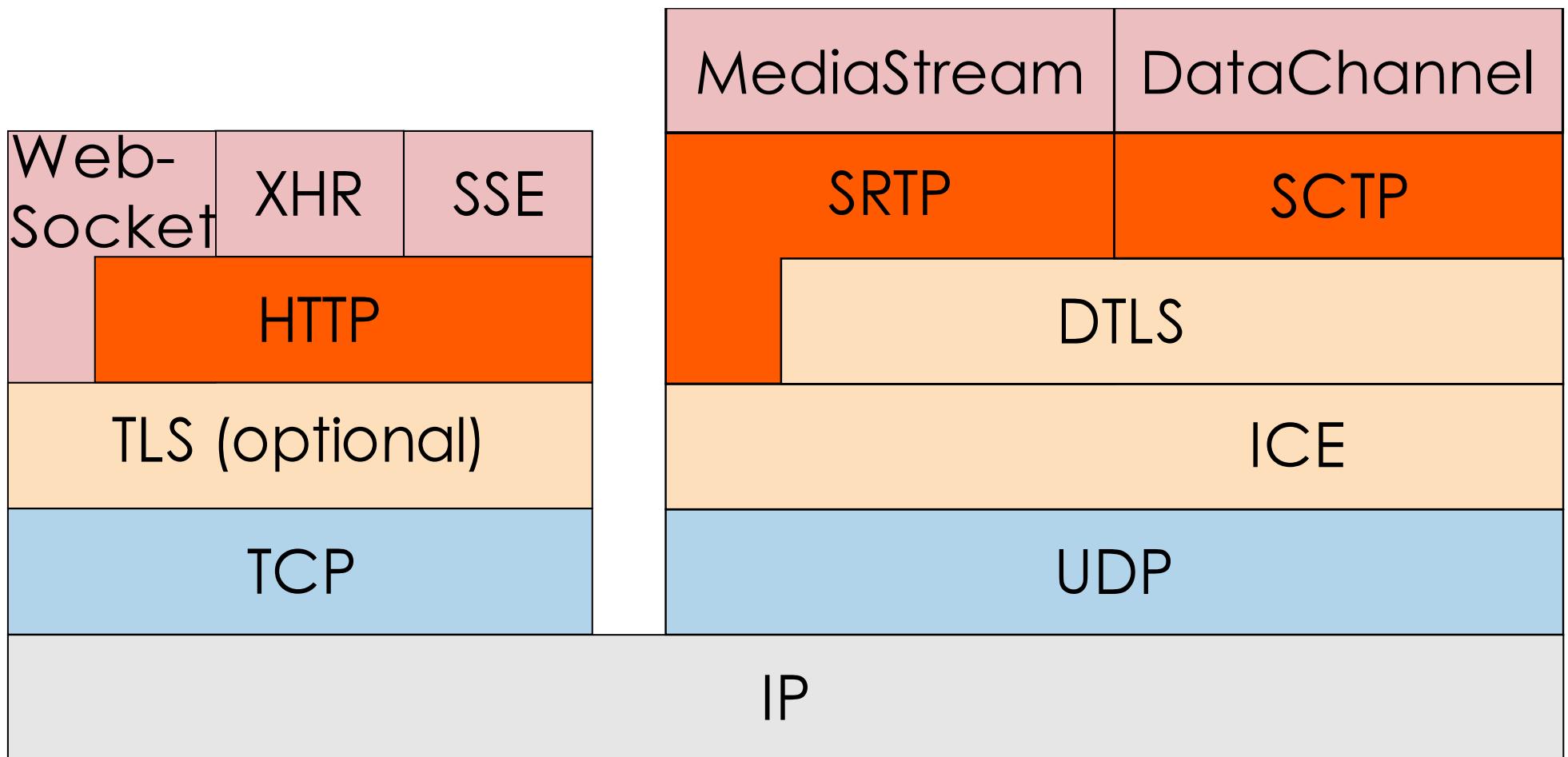
# Networking in Browsers

	Asynchronous	Architecture	Communication
HTTP	no	Client/Server	Unidirectional
Server-sent Events	yes	Client/Server	unidirectional
WebSocket	yes	Client/Server	bidirectional
WebRTC	yes	Peer-to-Peer	bidirectional

# This is WebRTC

- Enables direct communication between Web Browsers
- NAT Traversal
- No intermediary server necessary
- Secure RTP for audio/video data
- SCTP over DTLS for generic data
- Signaling via SDP offer/answer
- Optional source authentication

# The Protocol Stack



# The API: Basics

```
// Alice
var pc = new window.RTCPeerConnection();
channel.onmessage = function(msg) {
    pc.setRemoteDescription(msg);
};

function createOfferCallback(offer) {
    pc.setLocalDescription(offer);
    channel.send(offer);
}
pc.createOffer(createOfferCallback);
```

```
// Bob
var pc = new window.RTCPeerConnection();
function createAnswerCallback(answer) {
    channel.send(answer);
}

channel.onmessage = function(msg) {
    pc.setRemoteDescription(msg);
    pc.createAnswer(createAnswerCallback);
};
```

# The API: Data Channels

```
// Alice
var dc = pc.createDataChannel();
dc.onopen = function() {
  dc.send('hi bob');
};
```

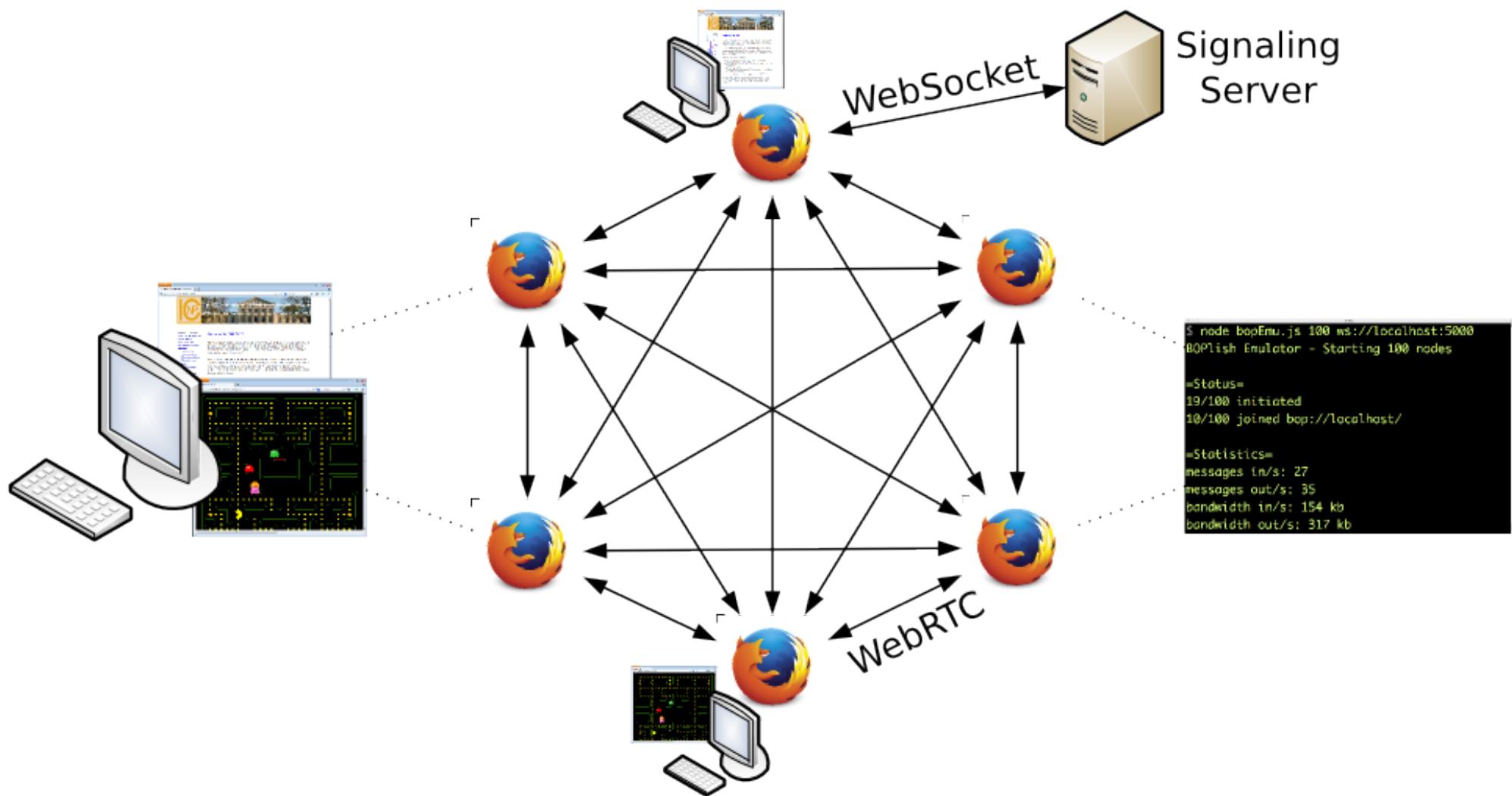
```
// Bob
pc.ondatachannel = function(evt) {
  var dc = evt.target;
  dc.onmessage = function(msg) {
    alert(msg); // alerts 'hi bob'
  };
};
```

# The API (on the wire)

```
1 v=0 // protocol version
2 o=Mozilla-SIPUA-33.0 14557 0 IN IP4 0.0.0.0 // origin
identifier
3 s=SIP Call // session name
4 t=0 0 // may indicate start/stop times; not used in WebRTC
5 a=ice-ufrag:c43d936f // ICE parameter
6 a=ice-pwd:0a0435ce7407bfc0ec43a953278916c7 // ICE parameter
7 a=fingerprint:sha-256 <omitted> // DTLS parameter
8 m=application 49687 DTLS/SCTP 5000 // Request for Data Channel
9 c=IN IP4 84.130.199.184 // Connection Endpoint (overwritten by
ICE)
10 a=sctpmap:5000 webrtc-datachannel 16 // Data Channel parameter
11 a=setup:actpass // Identifies the offerer, wait for answer
12 // ICE candidates following:
13 a=candidate:0 1 UDP 2130379007 192.168.0.20 49687 typ host
14 a=candidate:1 1 UDP 1694236671 84.130.199.184 49687 typ srflx
raddr 192.168.0.20 rport 49687
```



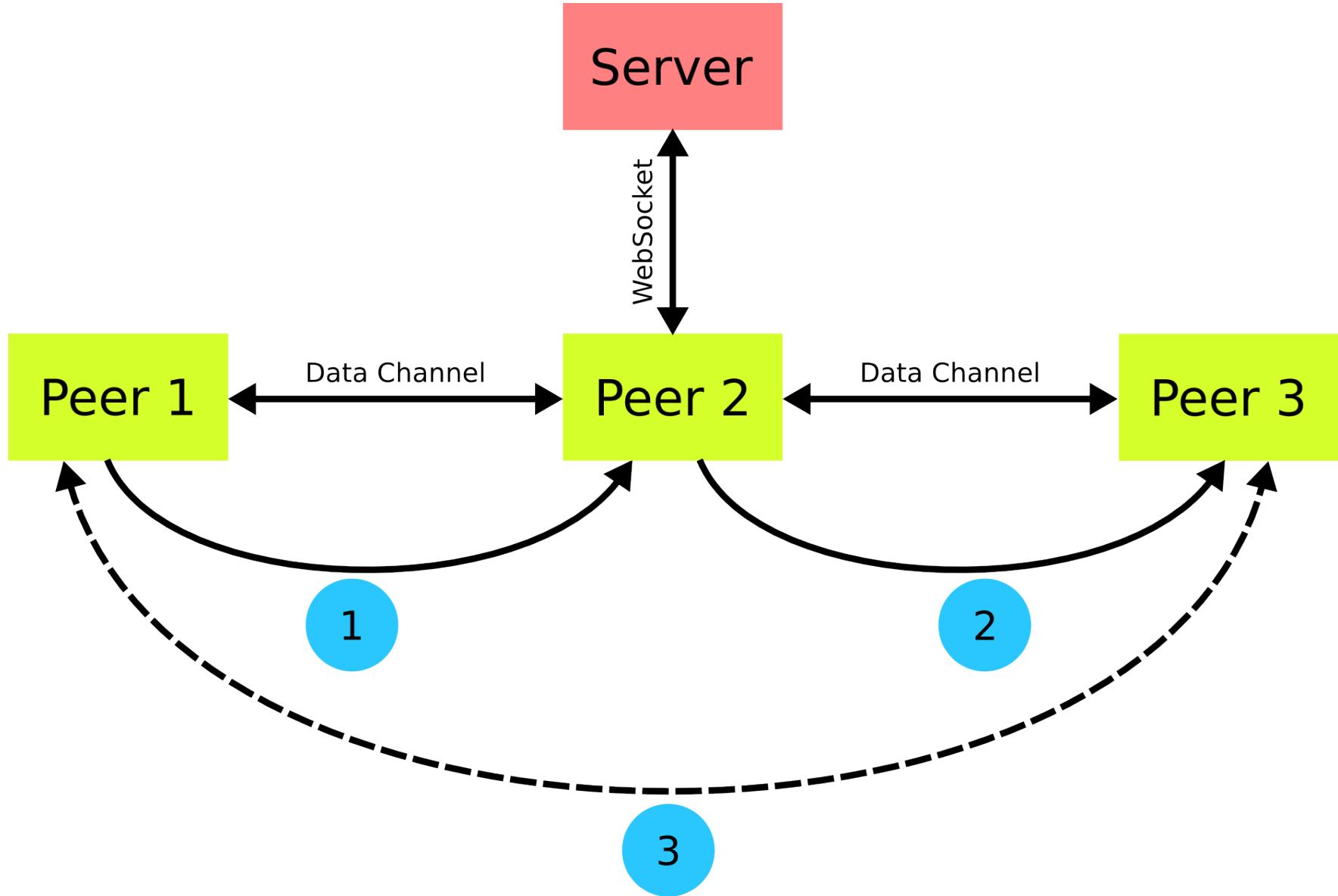
# BOPlish: Overview



# BOPlish: What do we need?

- Routing of Data
- Content Names and Name Resolution
- Simple API

# Routing



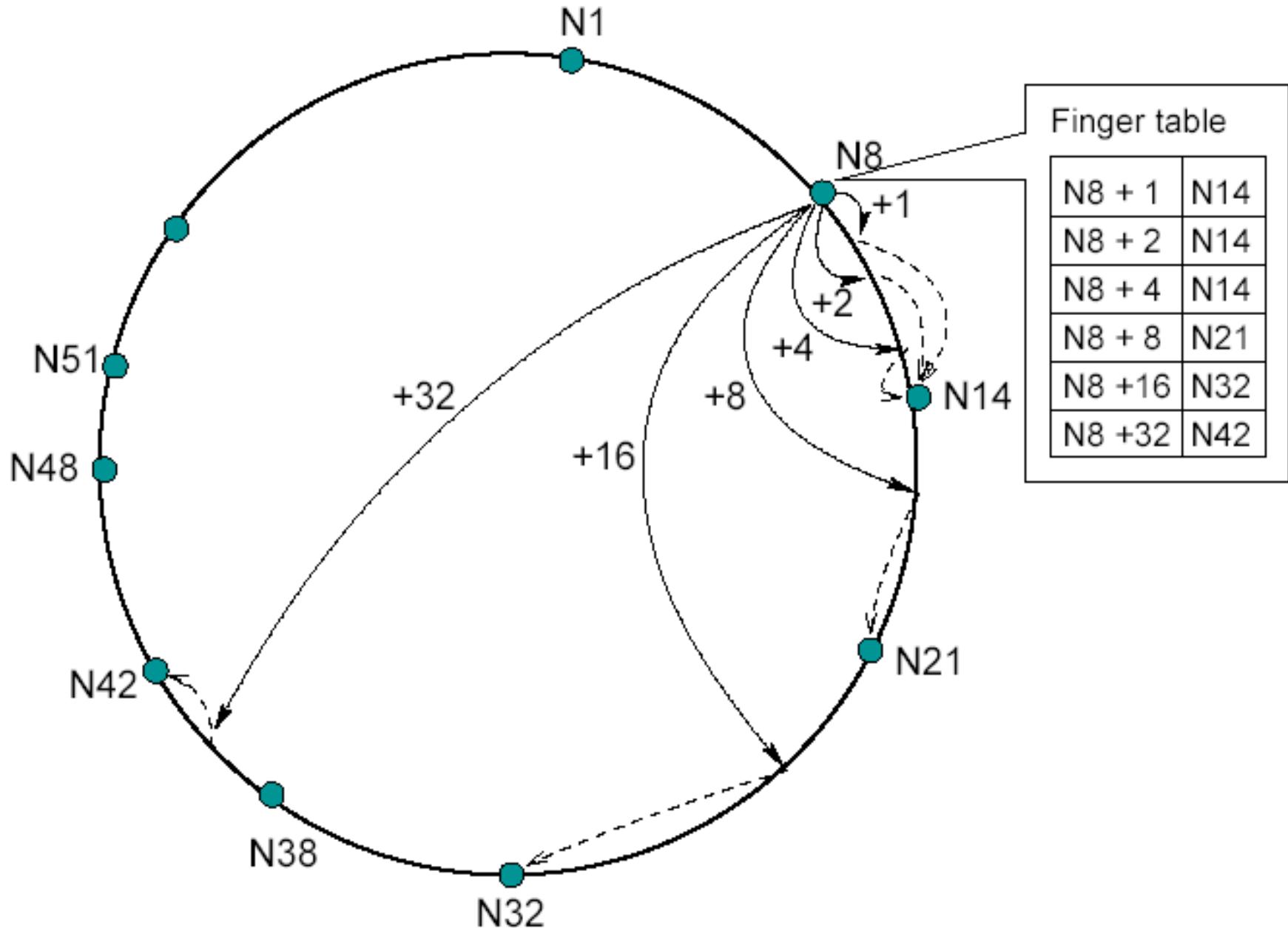
# Content Names

- Location-independent names
- Easy replication and migration
- URIs, the common scheme for names on the Web:

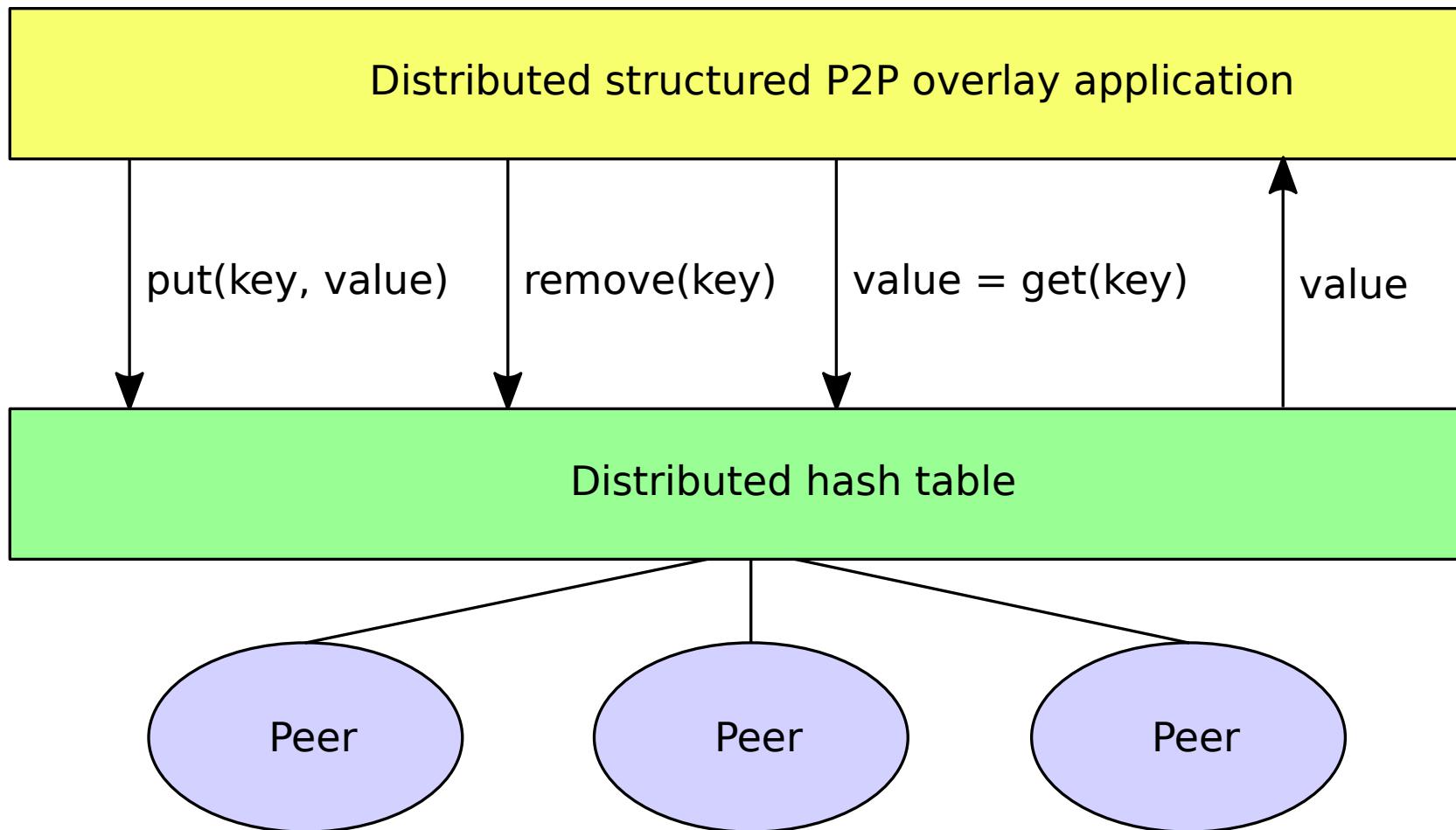
bop:username@idp:protocol [/path [?parameters] ]

# Name Resolution: DHT

a distributed key/value store

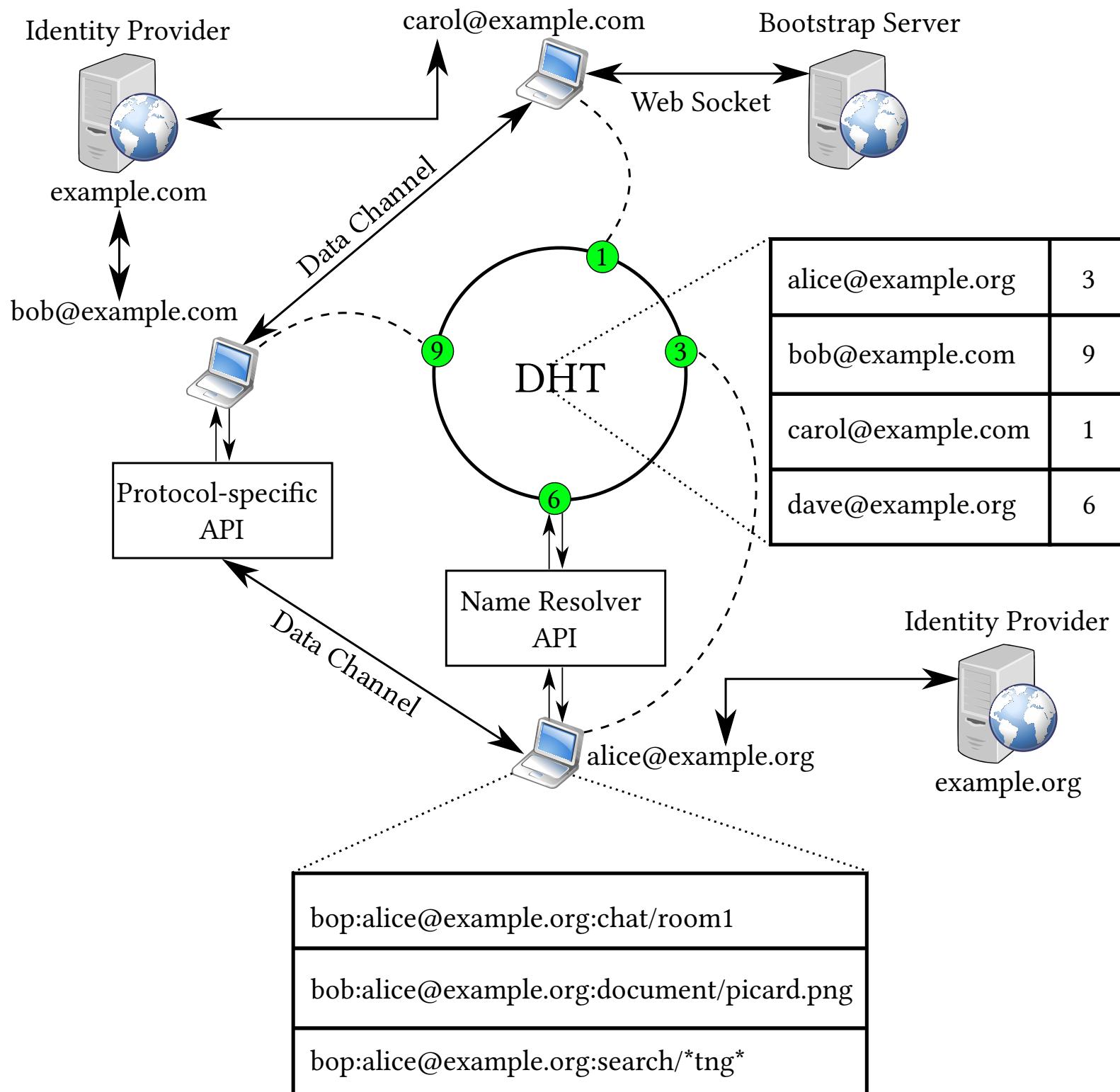


# Name Resolution: API



# Simple API

```
var client = new BOPlishClient("ws://boplish.com",
    function onconnect() {...},
    function onerror() {...});
var protocol = client.registerProtocol('groupchat');
protocol.onmessage = function onmessage(msg) {...};
```



# Use Cases

- Document Sharing
- Content Search
- Real-time Chat

# Use Cases: Document Sharing

“Every time you email a file to yourself so you can pull it up on your friend's laptop, Tim Berners-Lee sheds a single tear.”

```
bop://max@example.org/family.png?\nchecksum=sha256:a2bd. . .
```

# Use Cases: Content Search

bop://max@example.org/Music/\*johnossi\*

# Use Cases: Real-Time Chat

The image shows two side-by-side browser windows. The left window is titled '141.22.28.166:5000/admin.html' and displays a user interface for managing rooms. It includes a 'room name' input field, a 'Create Room' button, a 'Rooms' section listing a room named 'bop:dqbtcqkws@id.com:chat/star-trek-chat', and a 'Log' section showing a transcript of messages between two users. The right window is titled '141.22.28.166:5000/client.html' and shows a user interface for joining a room. It has an input field containing the room URL 'bop:dqbtcqkws@id.com:chat/star-trek-chat' and a 'Join Room' button. Below the input field, the log shows the user joining the room and sending a message.

**Left Window (admin.html):**

- Room Name:
- Create Room:
- Rooms**
  - bop:dqbtcqkws@id.com:chat/star-trek-chat
- Log**

```
Sending "I'm Captn. Kirk" to unxxqrtuik@id.com
mtiqdulxiggb@id.com: {"type": "JOIN", "room": "star-trek-chat"}
{"star-trek-chat": ["unxxqrtuik@id.com", "mtiqdulxiggb@id.com"]}
mtiqdulxiggb@id.com: {"type": "CHAT", "room": "star-trek-chat", "message": "I don't
believe you!"}
Sending "I don't believe you!" to unxxqrtuik@id.com
Sending "I don't believe you!" to mtiqdulxiggb@id.com
```

**Right Window (client.html):**

- Room URL:
- Join Room:
- Log**

```
Joining bop:dqbtcqkws@id.com:chat/star-trek-chat...
Joined star-trek-chat
unxxqrtuik@id.com
unxxqrtuik@id.com I'm Captn. Kirk
mtiqdulxiggb@id.com joined this room
mtiqdulxiggb@id.com I don't believe you!
```
- Chat Input:
- Send Button:

bop:alice@example.org:chat/room1

# Ressources

Project Home

<https://github.com/boplish/>

Core Library

<https://github.com/boplish/core/>

Demo Applications

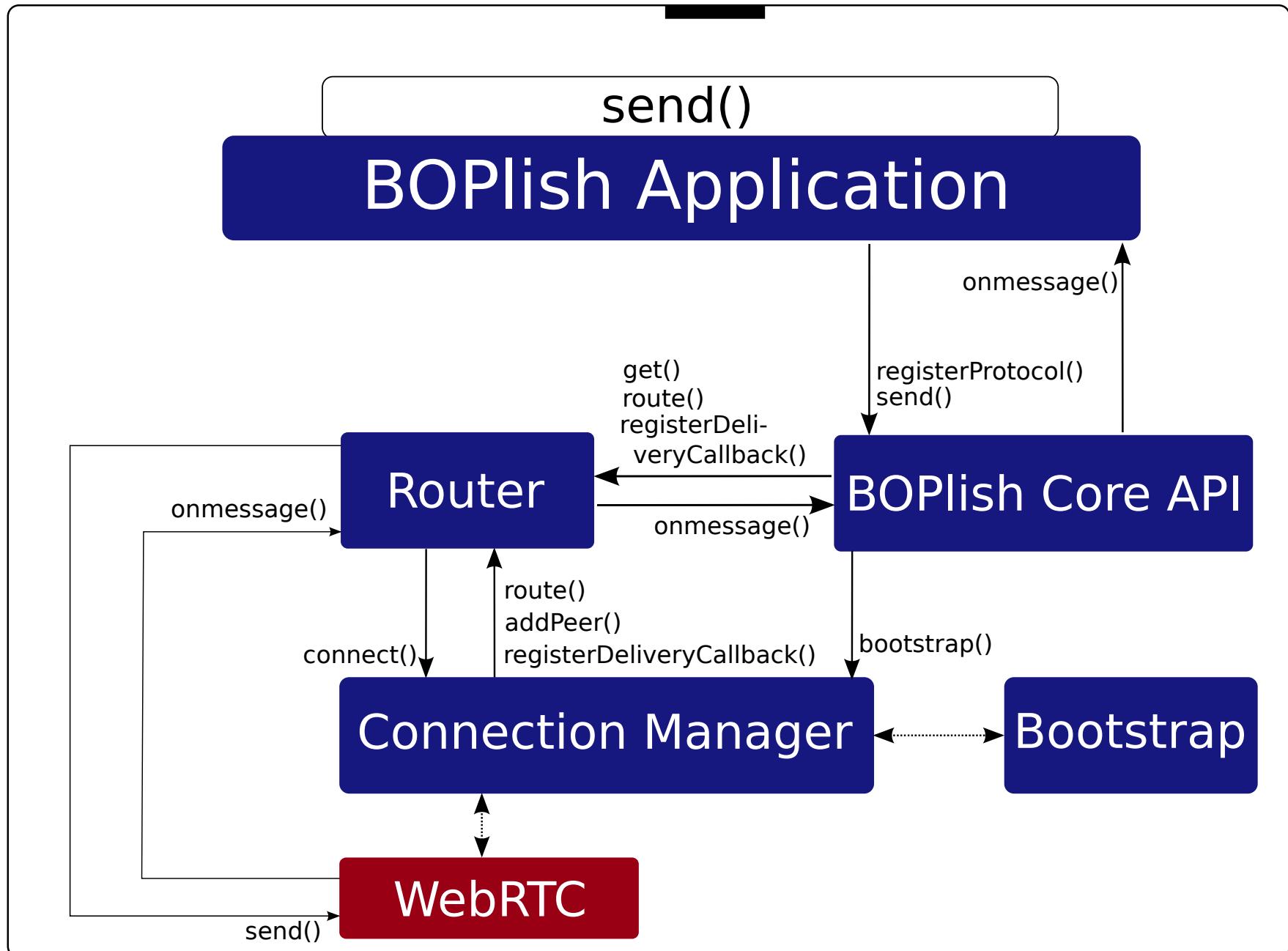
<https://github.com/boplish/demos/>

# Recap/Outlook

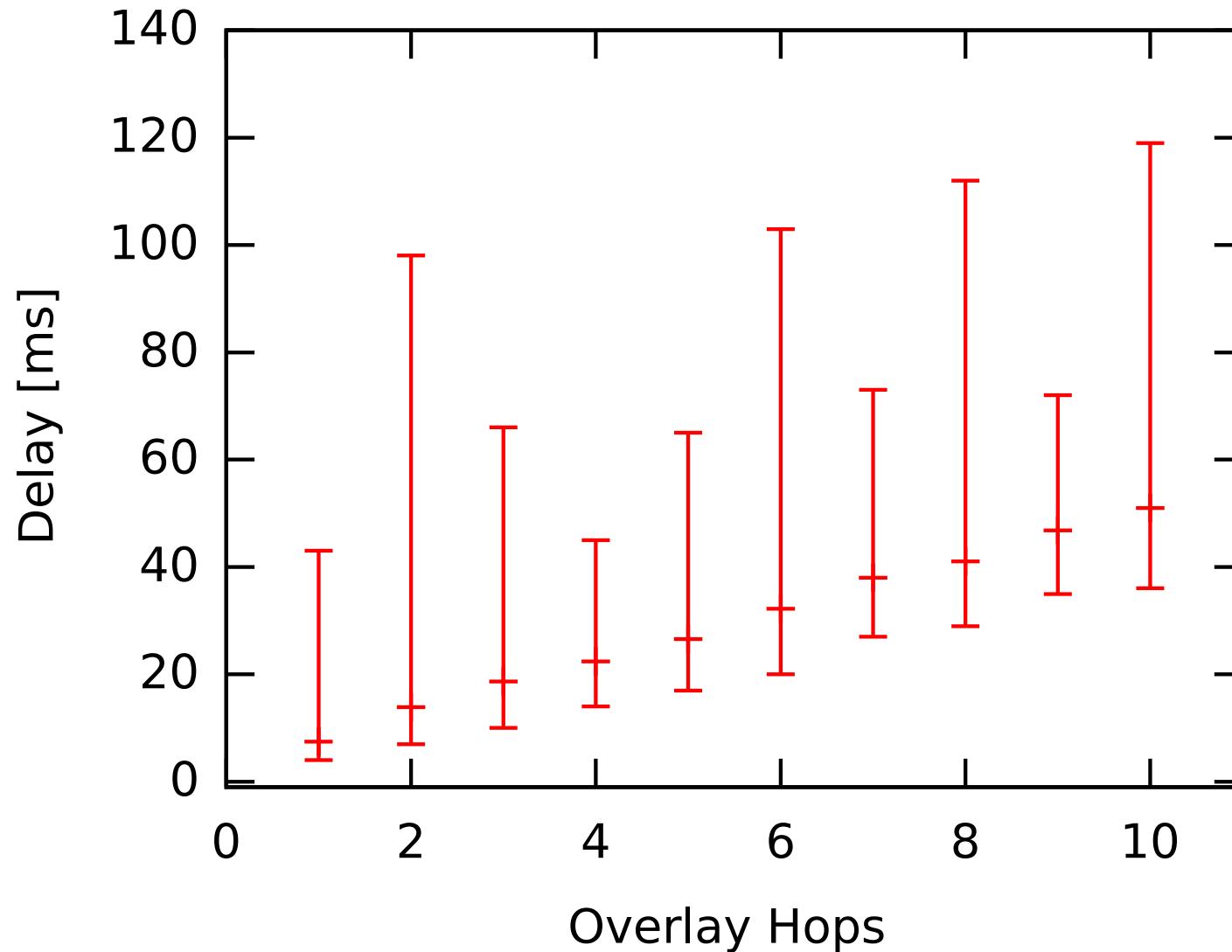
- BOPlish provides a framework for distributed content communities
- It's as simple as pulling in `boplish.js`
- WebRTC is cumbersome at times
- TODO: mobility, content offloading, network/P2P stability, security

# BACKUP

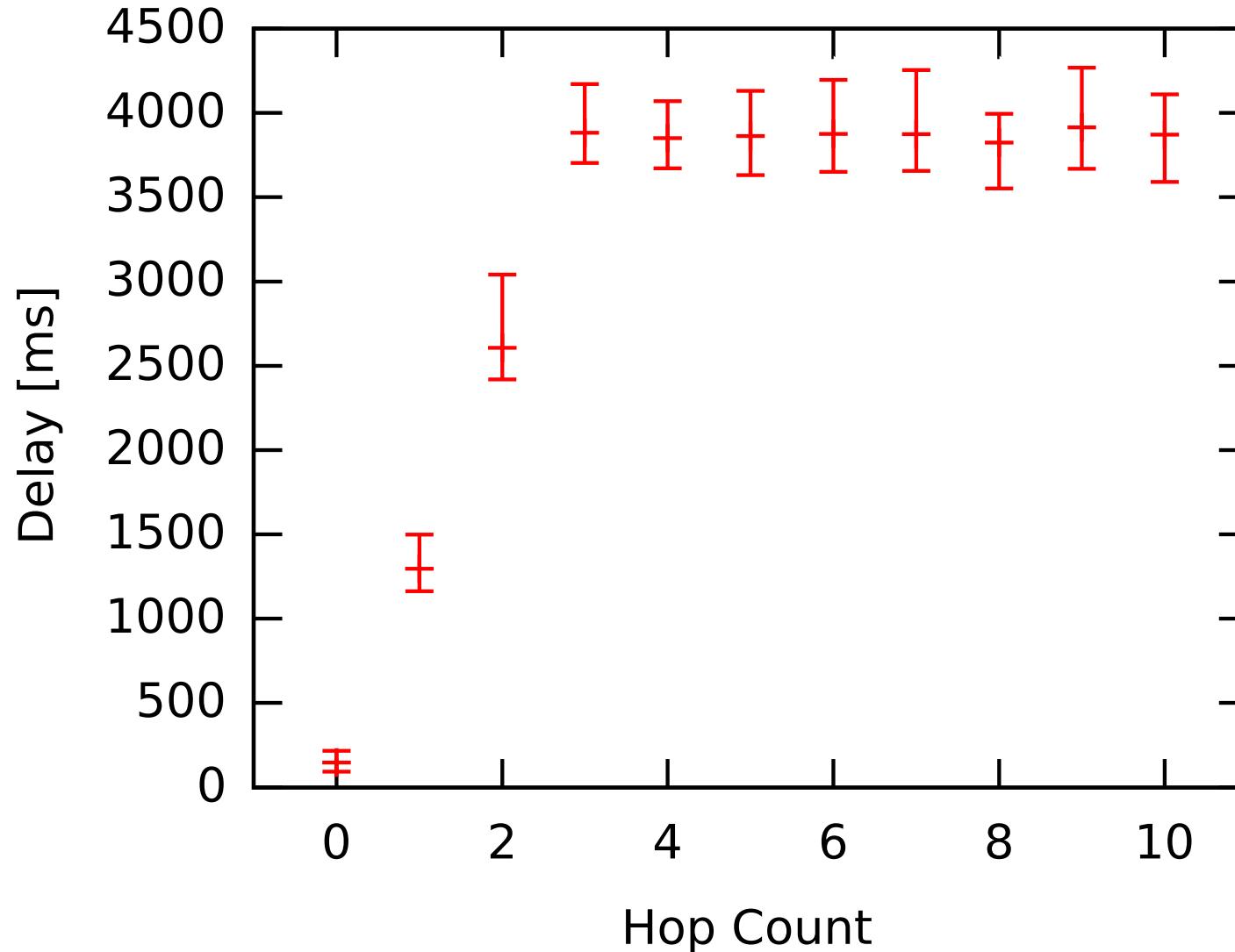
# Architecture



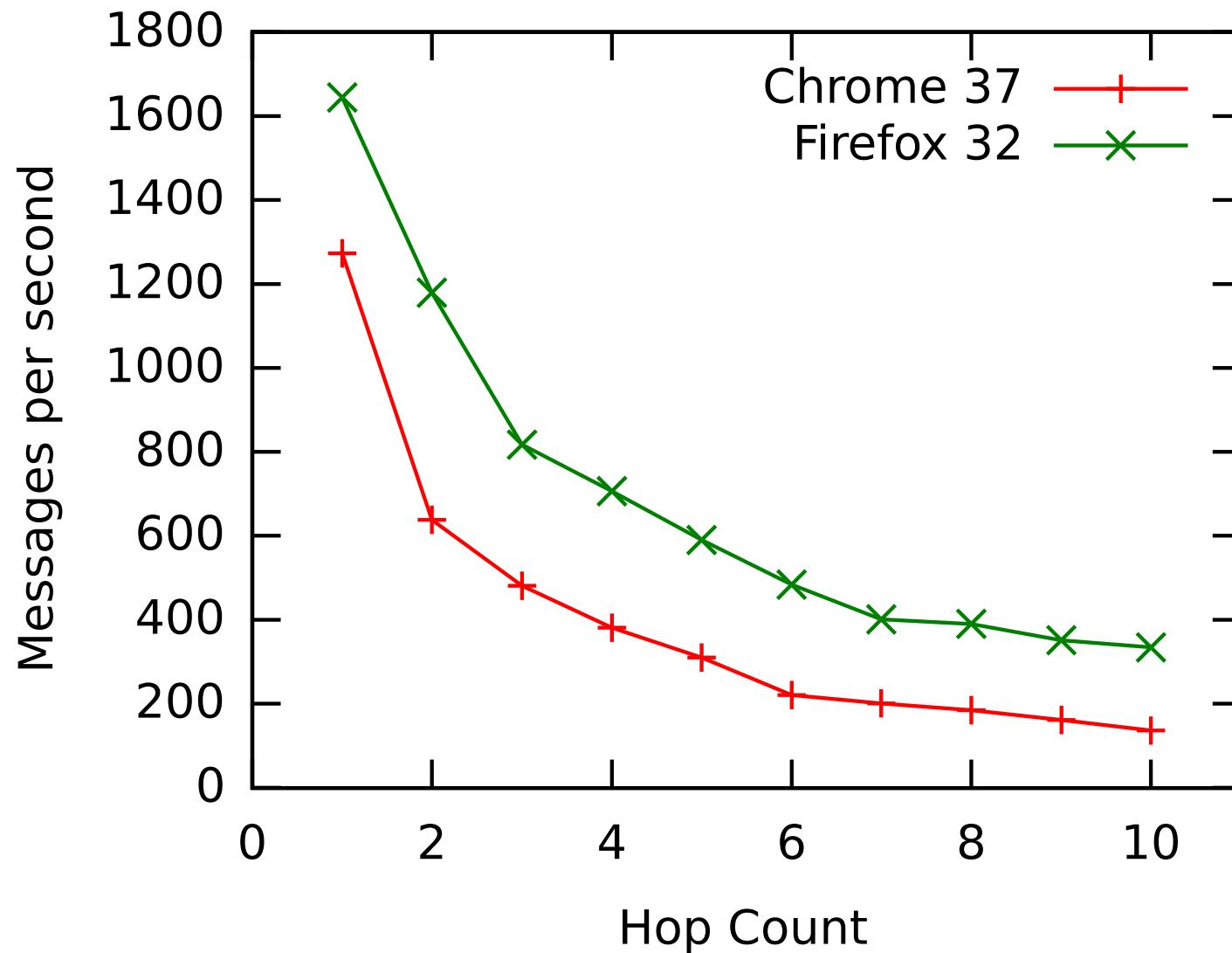
# Lookup Performance



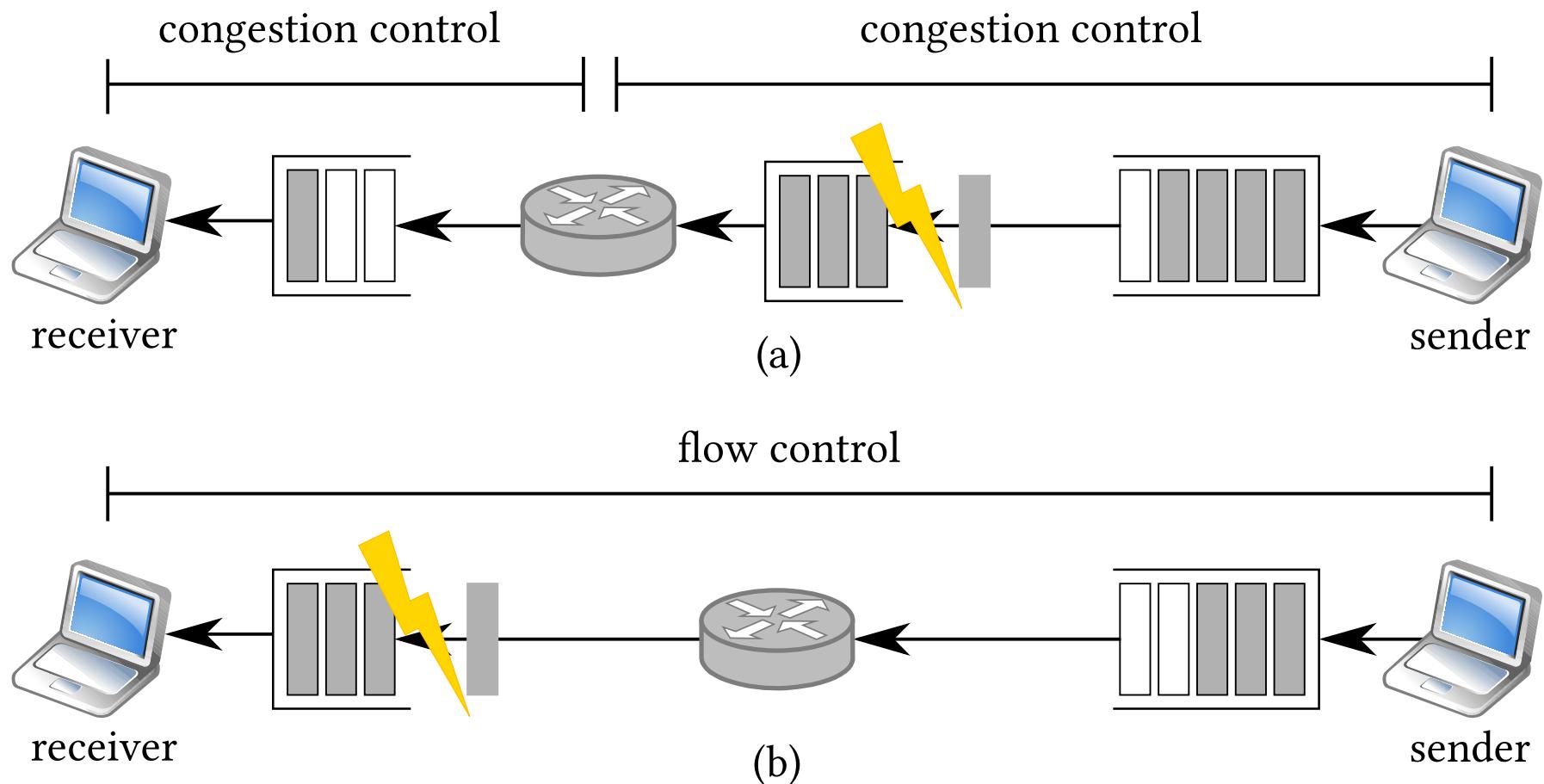
# Gross Bootstrap Performance



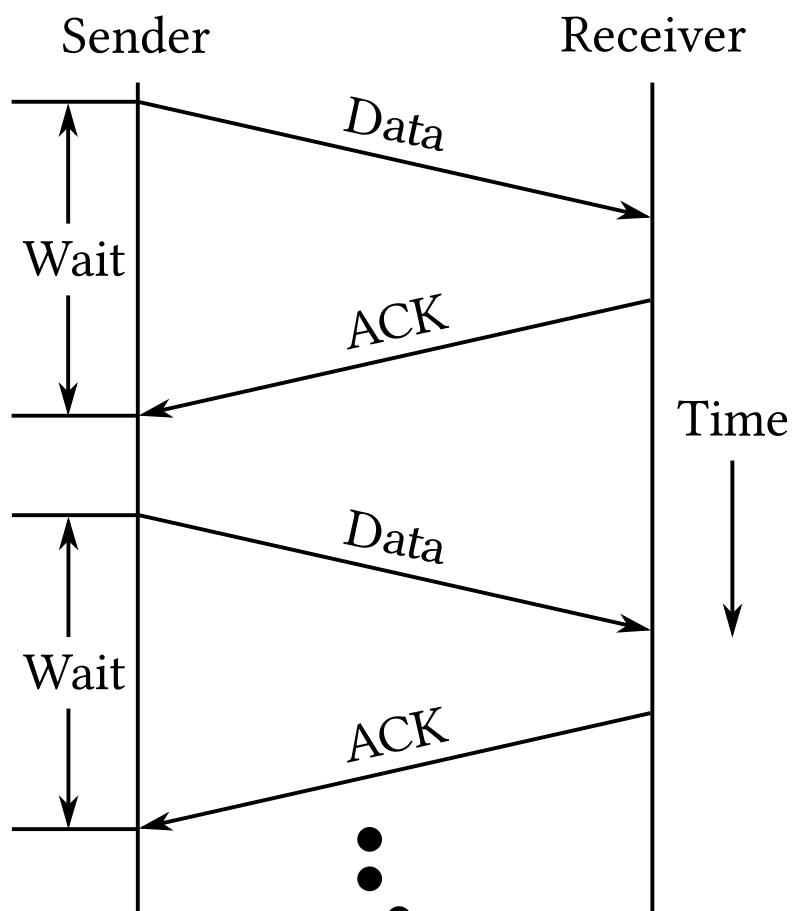
# DHT Performance



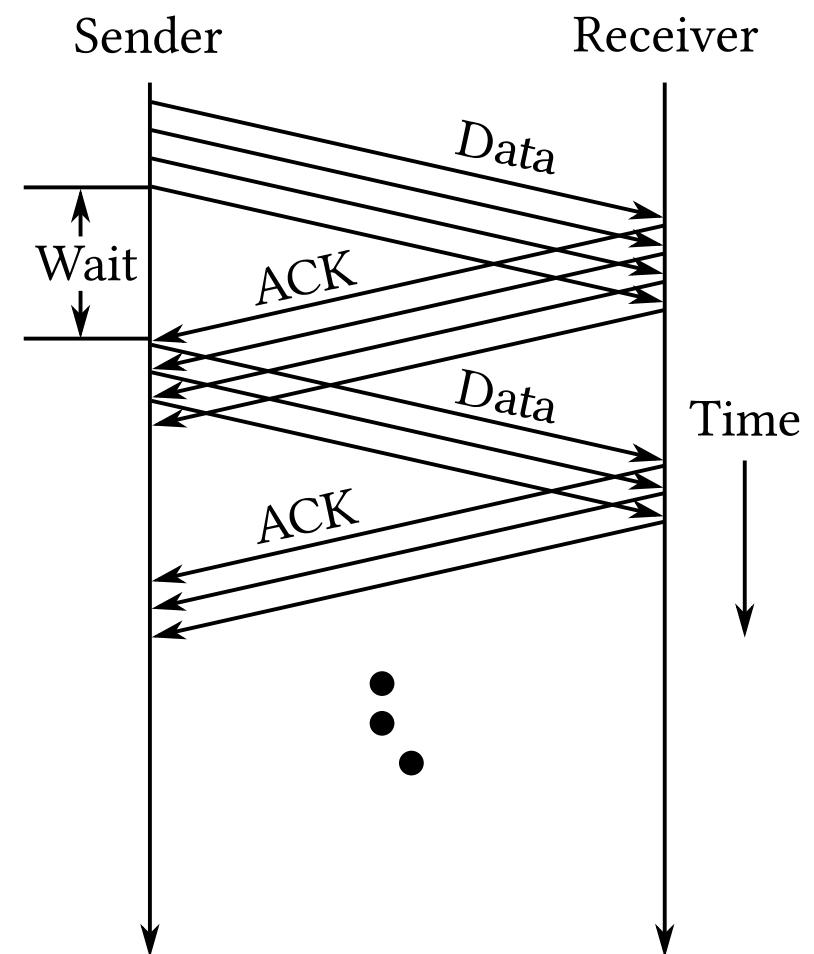
# Flow Control



# Flow Control



(a)



(b)

# Flow Control

