Real Time Web Applications  
Comparing frameworks and technologies

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# Abstract

This thesis is the result of a project with two parts. The first part compares five frameworks for real time web applications based on usability. Socket.IO, SignalR, Lightstreamer, Meteor and Play were the five frameworks. I used each framework to implement an auction house with real time capabilities. This allowed me to get an impression of how each framework is to work with. The use of a framework with limited help regarding real time, gave insight into how it is to make real time functionality from scratch.

In the second part, load testing of each framework was in focus. SockJS replaced Meteor of practical reasons in this part. The frameworks went through a series of tests to determine performance in message frequency, latency, network an machine resource usage. This part also focus on comparing the various transport mechanisms for real time. The ultimate goal of this was to find out if WebSockets, a new protocol for real time, is better than HTTP in this matter. I also investigated if WebSockets can replace HTTP for traditional web browsing.

Finally, I summarize both part and make a conclusion. The best framework is a nuanced answer, but Socket.IO was the best all purpose framework. It turned out to be hard to implement real time without much support. WebSockets outperform HTTP for real time, especially with a lot of communication from client to server. It is not, however, suitable as a replacement for HTTP in traditional web.

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# Preface

This thesis is part of my Master's degree in Programming and Networks at the Department of Informatics, University of Oslo. At the University, I have received guidance from the research group "Object-orientation, modeling, and languages" (OMS) at the University. The main part of the work has been overseen by a senior developer from BEKK Consulting.

All opinions made in the thesis reflects the authors opinions only. They are not representative for those of neither the University of Oslo nor BEKK Consulting.

I have received some help from the Italian company Weswit regarding server configuration for Lightstreamer. They also provided a free license for the product. Otherwise, they have not affected my work or influenced my opinions.

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# Introduction

The concept of real time web has existed for quite some time. But over the last few years it has begun to gain popularity. With the introduction of WebSockets we have gotten a new protocol designed entirely for this purpose. I will give an introduction to the motivations behind this protocol and the concept of real time in the background chapter.

Real time features in web applications has become more common. Several frameworks have surfaced to aid the development of such features. This thesis will look at some of these frameworks. I want to develop an application with real time features to test the usability of frameworks. With usability I mean from a developers perspective.

Getting some hands on experience with different frameworks will also help determine if they are needed at all. A real time feature can range from a simple task to a dominating aspect of a web application. I intend to find out if you need a framework no matter the size of the task.

This thesis will also look at various performance aspects. First of all the performance of different frameworks using different methods for real time communication. But I will also look at how these methods compare to each other. This is the most interesting aspect, as it can give insight into the importance of the WebSocket protocol.

It is a clear separation of themes in this thesis. On one hand you have the usability aspects of various frameworks. On the other you have performance of these and of methods for real time. To handle this, the thesis work is separated into two parts. The first part covers the development aspect, while the other handles performance.

Both parts are summarized a common conclusion part. This allows me to make a total evaluation of the frameworks. It also allows me to see trends in the transport mechanisms' performance with the different frameworks.

## Problem statement

The problem statements in this thesis are as follows:

**I: What framework do I consider to be the best for real time web applications?**

No matter the functionality you want to build, you always want the best tools to do it. When it comes to software though, this question will have a nuanced answer. Some frameworks may handle different aspects better than other, while others may have the best performance.

**II: Is it always useful to use a framework?**

If the functionality you need is simple, does your project benefit from an extra dependency? Or can you make it yourself without much extra work? Sometimes you would like as few dependencies as possible. But if it takes up a lot of time to do something yourself, a framework may be a better solution.

**III: Does WebSockets outperform the old, established HTTP methods for real time?**

If it turns out that WebSockets doesn’t perform better, is it something we need?

**IV: Can WebSockets be the foundation for the next generation of HTTP?**

The answer to this is coupled with the previous question. If WebSockets is just as good as HTTP performance wise, it will not be the foundation for the next generation. However, if it does, can it change the way browsers communicate with browsers in more aspects than real time?

## Terminology

This section describes certain terms and phrases used throughout the thesis.

I use the terms "library" and "framework" to describe what I am testing in this thesis. A library is a much smaller entity than a framework [1]. For the sake of consistency, I use framework as a general term. When referring to a library in specific, I use that term.

The term "transports" refer to various mechanisms for real time communication. These include WebSockets, Server-Sent Events, HTTP-streaming, long-polling and polling.

"WebSockets" appears as a plural word in this thesis, but it is not treated that way. It refers to the WebSocket protocol, and is treated just like the term "HTTP". Thus formulations like "WebSockets is..." and "HTTP is..." appear in this thesis.

Automated tests are mentioned several times in this thesis. Such tests involve unit testing, integration testing and functional testing. Unit testing refers to testing a single unit [2], or class in an object oriented language. Integration tests puts several units together [2]. In this thesis, I use the term about testing the data access layer of applications[[1]](#footnote-1). Functional tests test behavior of an application [2]. In this thesis, this kind of testing is done using browsers.

I use "Node" as a short term for Node.js throughout this thesis.

I use “Play” as a short term for Play! Framework throughout this thesis.

## Code base

The code base for this thesis is located on GitHub [3]. You can also find all raw data for the results here.

## Outline

This thesis is made up by seven main parts, each with chapters. The parts mean to make the context of each chapter more clear to the reader. I give a brief overview of the parts and chapters below. Bold text indicates a part, while underlined text indicate a chapter. Chapters are also accompanied with chapter number.

**Chapter 1: Introduction:** The introduction gives an overview of the thesis, its problem statements and terminology.

**Background:** This part contain chapters that describe relevant topics for the thesis.

The World Wide Web: A short chapter that introduce the reader to some changes the Web has seen throughout its lifespan.

Real time: This chapter explains what real time is and how to achieve it with HTTP and WebSockets. It also discuss drawbacks and benefits of both.

Frameworks for real time web applications: This chapter provides a short introduction to a number of frameworks that exist for real time web applications.

Servers: This chapter gives a short description of various servers that can be used to host the different frameworks.

**Project part 1: Hands on development:** This part describes the work done regarding usability of five selected frameworks. It represent the first of the two parts of my work.

Methodology: This chapter describes the methodology of the first part of my work.

Results: This chapter describes my experiences with each of the selected frameworks.

**Project part 2: Load testing:** This part describes the work done regarding the load testing of the frameworks.

Methodology: This chapter describes the methodology of the second part of my work.

Results: This chapter presents the results from the load tests.

Analysis:In this chapter I discuss the validity of the results. I also provide an interpretation of what they mean.

**Conclusion:** This part contains summaries of each of the project parts. It also contain the conclusions to each of the problem statements.

Frameworks: This chapter covers the two first problem statements as well as a summary of all framework related work.

WebSockets versus HTTP: This chapter covers the last two problem statements. It also summarizes the work related to comparing WebSockets to HTTP transports.

Further work: This chapter provide suggestions to projects that can be based on this thesis.

# Background

# The World Wide Web

The World Wide Web has been available for 20 years [4], and is still considered a young technology. But over those 20 years it has changed in almost every thinkable way.

Over the years, the improvements to the Web have changed the way we use it. Visiting a web page before meant reading a page of text that maybe had some pictures on it. Today, Cascading Style Sheets (CSS) has given web pages a more vivid look with various styling options. Asynchronous JavaScript and XML (AJAX) has made them more dynamic. Finally, with HTML5, more revolutionary changes are yet to come.

Along with HTML5 comes a new protocol for the Web–WebSockets. Its purpose is to meet one of the newest aspect of web browsing, namely real time applications. A real time application is when clients receive updates from the server as they occur (for more information see chapter 3). Real-time web applications has been around for some time, but before they have relied on the aging HTTP 1.1 protocol.

## HTTP/1.0

Version 1.0 of HTTP was created in 1996-the World Wide Web's childhood [5]. Besides text, web pages maybe had a few embedded objects at that time[[2]](#footnote-2). But as the Internet grew, it soon became clear that the user experience had to be improved.

At this time, around the mid 90s, CSS too was in its childhood [6]. It caught people's attention and more and more browsers started to support it. Embedding a style sheet in a HTML-file adds another object that the client has to download. This is no problem today, but with the HTTP 1.0 protocol it required quite a lot of unnecessary work for both the client and server.

Downloading one element in a HTML-file, or even the HTML-file itself from the server required one TCP request (Figure 2‑1). The server then replied and closed the connection. Throughout the duration of the request, the client waited [7]. Getting a HTML-file with a style sheet and three images then required five consecutive requests in total.



Figure ‑: Several request to get the whole page.

## HTTP/1.1

After just three years, HTTP/1.1 was released as a standard [8]. It introduced several improvements. One of these was persistent connections. This allowed several request to made at the same time [9]. It was a dramatic change at the time, as it allowed clients to get several objects concurrently.

Another radical improvement was the ability for a browser to cache parts of an object. This allowed an interrupted download to be resumed later by the help of the cached data. Web applications were also given the possibility of sending chunked data [8], letting servers start sending a response without knowing how long it was. In theory, it could be infinite, as we shall see in section 3.1.3.

The authors of the protocol showed great foresight when they made sure that future protocols could be backwards compatible with HTTP 1.1. The upgrade request-header [7] makes it possible for a client to request the use of another protocol. The server can then chose to upgrade, but it is not required.

Updating from version 1.0 to 1.1 may not seem like a giant leap, but it actually was. Looking at the lengths of the different protocol specifications is an indication of how much more detailed the 1.1 protocol is[[3]](#footnote-3).

## HTTP/2.0

During the work period of this thesis a working draft for the 2.0 version of HTTP surfaced. It started out as an initiate from Google called SPDY [10] in 2012. This was then used as the basis for the HTTP/2.0 specification in November that year [11]. The draft has had steady development throughout 2013, and it is planned to be delivered as a proposed standard November 2014 [12].

There are several motivations for a new version of HTTP. Modern web pages contain many elements embedded in its HTML. Today a client parses the HTML and makes requests to get elements as it reads them. HTTP/2.0 allows multiple requests and responses to be sent concurrently on a single connection. The draft referrers to this concept as a stream [11]. With this concept, a server can provide all embedded objects in an HTML page as several responses to a single request (Figure 2‑2).



Figure ‑: Several responses pushed to a client after a single request.

Another vital improvement proposed with HTTP/2.0 is header compression. The goal of this feature is the same as the stream concept. With more concurrency and compression, networks will experience less load. Download times will also be faster, improving the user experience.

Improvements to security are also among the goals of the draft. There are some indications pointing towards a web where unencrypted traffic will no longer exist [13]. It will be exiting to follow the development of the draft.

# Real-time

As mentioned in chapter 2, one of the newest additions to the World Wide Web is real time applications. There are varying degrees of real time content provided by such an application. At the lower end of the scale, there are for example online comment sections that update whenever someone posts a comment. An example of an application with more real time content is Facebook. It displays notifications and your friends’ activities to you as soon as it happens (Figure 3‑1).

“As soon as it happens” is exactly what real time is: providing updates for the client immediately, without the need for refreshing the page on the client side. And as the examples above show, the real time aspect of an application can be either a small feature, or the core concept of the application.



Figure ‑: Screenshot of real-time updates in Facebook.

## The Real-time Web with HTTP

Recently the concept of real time web has become a buzzword. Real time works by pushing updates to clients as they appear. This is not how HTTP works though–the client always has to initiate the communication. To accommodate the growing need for applications of this sort, several techniques have been utilized. Using HTTP in untraditional ways has been the regular way of accomplishing real time until recently. The introduction of WebSockets may render this deprecated.

### Polling

As the first attempt of providing real time updates from a server, polling is a simple approach. It works by having the client make normal HTTP-requests, but at a set interval [14]. The server then instantly sends back a response–either containing new data or just an empty response if there was nothing to retrieve (Figure 3‑2). Polling has obvious flaws like how to set the interval to prevent empty responses while not flooding the server. Therefore, other mechanisms are far more widespread.



Figure ‑: Polling.

There is a way to improve a little upon polling: piggybacking [15]. Polling the server at regular intervals is usually done in parallel to other HTTP-requests initiated by client actions. These actions, of course, also get responses back from the server. Piggybacking takes advantage of this by also sending updated data back via the response. In that way, the client may get new data between the polling interval (Figure 3‑3).



Figure ‑: Polling with piggybacking.

### Long-polling

Long-polling is related to polling. It basically works the same way, but with one rather important difference. By utilizing the keep-alive header in HTTP 1.1, the connection to the server is kept open after the client has made a response [15]. This allows the server hold the response back. It cannot do this forever though, so eventually it times out. The client then makes a new request [16] (Figure 3‑4).



Figure ‑: Long-polling.

### HTTP-Streaming

HTTP-streaming is an old technique introduced by Netscape as early as 1992 - well before even HTTP 1.0 became standard [16]. Two forms of streaming exist, namely page streaming and service streaming. The first of the two has the server streaming content in a long-lived TCP-connection. Accomplishing this requires the server to never send the instruction to close the connection. Instead, it remains open throughout the entire course of a client’s session. This type of streaming is otherwise known as XHR-streaming [17]. The name comes from the use of a long-lived XMLHttpRequest to send new data. Page streaming uses the initial page request to stream data. This gives more flexibility regarding the lifetime of the connection. Otherwise, the two methods are similar.

The most common implementation of this technique today is the so-called forever frame. As mentioned in section 2.2, HTTP 1.1 allows a server to send a response without knowing in advance its length. A forever frame is just an iframe that receives script-tags in an everlasting response from a server [18]. Browsers execute script-tags when they it reads them. The server therefore sends data to the clients wrapped up as such (Figure 3‑5).



Figure ‑: A client using the forever frame technique receives script-tags from the server.

### Comet

Long-polling and HTTP-streaming are often referred to as Comet or Comet Programming [19]. Comet is an umbrella term that captures different ways to have the server as the initiating part in client/server communication. A rather significant effort has been made to create an official standard for Comet [20], but it has yet to become approved by the IETF as a RFC[[4]](#footnote-4). With the introduction of WebSockets, it may never be.

### Server-Sent Events

Let’s move on into the borders of Web 2.0 with HTML5s Server-Sent Events [21]. Server-Sent Events takes advantage of the "text/event-stream" Content Type of HTML5 to push messages to the client [22]. It is, in other words, a one way communication channel from the server to the client.

Still, the client always has to connect first – “subscribe” to the channel. Then the server can send events whenever new data is available. It can keep the connection open forever in theory. But both the client or any intervening proxies can close it. One can also configure it to close after a given amount of time. The time the client should wait to reconnect is also configurable [22]. Server-Sent Events behavior is quite close to long-polling (Figure 3‑6).

Unlike long-polling, though, developers using Server-Sent Events have access to a simple Application Programming Interface (API) [23]. The API gives access to the EventSource interface, which provides straightforward JavaScript code. It allows the server-side to fire events in the browser and, in turn, update the content on the client-side. With the possibility of setting an ID on each message sent, the client can easily reconnect and continue where it left off. The server just need to look up its ID. This makes Server-Sent Events very robust.



Figure ‑: Server-Sent Events resembles long-polling.

## WebSockets

We have seen that HTTP 1.1, that came only three years after its predecessor, was a significant step ahead. This protocol had backwards compatibility in mind (see section 2.2 about upgrade request-header), but just now a new version is in the works. With WebSockets developers can now finally make use of the upgrade request-header.

In December 2011, the WebSockets protocol became a proposed Internet Engineering Task Force (IETF) specification [24], [25]. The specification document states that the motivation for WebSockets is HTTPs lack of abilities for bi-directional communication between server and client:

*“The WebSocket Protocol is designed to supersede existing bidirectional communication technologies that use HTTP as a transport layer to benefit from existing infrastructure”* [22, section 1.1]

### How it works

WebSockets, like HTTP, use TCP as underlying protocol. While HTTP needs several "hacks" (see section 3.1), WebSockets provides bidirectional communication right out of the box.

The WebSocket protocol use the same ports as HTTP and HTTPS (80 and 443, respectively). This allows the initial handshake to be done via traditional HTTP (Figure 3‑7). The client states that it wants to use WebSockets, and the server sends a response if it supports it. This ensures backwards compatibility with browsers that don't support WebSockets. Developers can use this to make their applications use HTTP methods instead.



Figure ‑: WebSocket handshake as seen in Google Chrome.

Sending messages back and forth once the connection is up, is a lot more efficient than what HTTP can provide. Data in request/response headers in HTTP may accumulate to hundreds of bytes [14], while WebSockets sends messages in frames with only two bytes overhead [26]. Frames can be sent both ways at the same time eliminating the need for more than one request at the same time (Figure 3‑8).



Figure ‑: Frames can pass both way through one connection.

### The WebSockets API

As with Server-Sent Events, WebSockets has its own API [27], that provide the WebSocket interface. This API is a little simpler than the EventSource interface in my opinion. It does not support custom events; just open, close, receiving a message and error.

It provides an easy way to send messages using the send function. Besides, it has an attribute for keeping track of buffered data on the client. This makes the API rather powerful for in spite of being simple. Its simplicity is in accordance with the intention of the protocol:

*"Basically it is intended to be as close to just exposing raw TCP to script as possible given the constraints of the Web."* [22, section 1.5]

## Drawbacks of HTTP techniques

In section 3.1, I gave a rudimentary description of different ways to achieve real time, communication with HTTP. They mostly work in the same way, but uses some different settings for keeping connections open and pushing messages to the client. Most used is probably long-polling, because even the oldest browsers support it. However, there are also some issues that the following sections will describe.

### Really real time?

Long-polling builds upon the idea of polling. But whereas polling is a naïve approach, long-polling is smarter. One of the major issues with normal polling is how to determine the polling interval.

Thinking real time, one might want to say that the client should make a new request each time it receives the response of the last. With frequent updates, this would provide a lot of stress on the server. Dealing with this would require some serious load balancing technology, leading to a expensive solution. A short interval would also increase the number of empty responses if new updates arrive in an inconsistent interval at the server (Figure 3‑9).



Figure ‑: Short poll interval gets new data fast, but demands server resources.

With a longer interval (Figure 3‑10), the longer it takes before new data arrives at the client, thus making the application less real time. Even with piggybacking, one cannot achieve anything close to real time with a longer interval unless the server receives new data at a regular, known interval. As long as this interval isn’t too short, polling may be a good choice for such scenarios. A weather application for instance, might get new updates every hour. This client can use this knowledge to poll the server accordingly.



Figure ‑: Long interval makes the client miss a lot of updates in most cases.

### When long-polling becomes polling

As I said, long-polling is a lot smarter than polling. Keeping requests open for a longer time, ensures that the number of unnecessary requests is a lot less than with polling. Though if the server receives updates at a high rate, the connection will never be able to stay open.

Each time the client tries to initiate long-polling, there is always some data waiting for it and the server respond at once [14]. This effect makes long-polling work just as regular polling at a short interval. Comparing Figure 3‑9 to Figure 3‑11, one can see that long-polling does not outperform polling as long as the server-side updates are frequent.



Figure ‑: Long-polling has no benefit over polling at high update-rates.

Norges Bank Investment Management provided[[5]](#footnote-5) a counter on their homepage, [www.nbim.no](http://www.nbim.no), that showed the total value of the Norwegian Government Pension Fund. The counter updated several times per second. If each change in that number was a response from the server, it wouldn’t matter if it was polling or long-polling in use. The load on their network would be quite substantial in a short time. This little widget, though, faked real time as it polled the server every 30 seconds and got the values from the past 30 seconds.

### Streaming techniques

Using streaming techniques is a different approach than having the client poll for data. With HTTP-streaming and Server-Sent Events, the server is the initiating part. You can configure Server-Sent Events to work almost like long-polling, but it uses push just as streaming does.

A forever frame allows the server to push updates to the client wrapped up in script-tags, but it is far from perfect. Client-side there has to be some extra handling to make the received scripts do something useful. Receiving new data in an ever-growing DOM-element, also creates some challenges related to memory management. If you don't clear the frame at regular intervals, you will have a memory leak in your application. Using XHR-streaming, you avoid this issue. But you still have to handle a persistent HTTP-connection.

Having an open connection that sends a lot of data, gives rise to another problem; proxy-servers and firewalls [14]. The nature of the HTTP-protocol may cause these to buffer the response, thus creating a lot of latency for the client (Figure 3‑12). To avoid this, many Comet-based streaming solutions fall back to long-polling if it detects buffering.



Figure ‑: Intervening proxy buffering response-stream from server.

A forever frame makes the developer write some extra code to handle the incoming scripts. The EventSource interface gives developers a more powerful toolbox for handling incoming events (see section ). Utilizing pure eventhandlers also ensures that there is no need for cleaning up after the incoming data. But, in the end, Server-Sent Events are still HTTP. And as we shall see, the protocol has issues of its own.

## HTTP was never designed for real time

HTTP/1.1 introduced the keep-alive flag, chunked encoding and persistent connections. Claiming that HTTP wasn't designed for real time in spite of these features, may seem rather presumptuous. The next sections cover what I believe to be HTTP's greatest weaknesses compared to WebSockets.

### Overhead

In section 3.2.1, I mentioned that headers in HTTP requests/responses can accumulate to hundreds of bytes [14]. Peter Lubbers and Frank Greco made a simple application for comparing polling to WebSockets [28]. To shed some light over the issue with headers, I will borrow some data from their tests. Their simple stock-ticker application polls a server every second to get new data. The counterpart just uses WebSockets to get the same information.

In this particular case, the header-data for the polling application accumulates to a total of 871 bytes. With just a few clients this is not a lot. But when you have hundreds of thousands clients, the network throughput increases exponentially. A use case with 100 000 users polling every second means that the server's network has to deal with 665 megabits per second of throughput[[6]](#footnote-6). Having the same amount of messages in WebSockets creates only a fraction of that. With 2 bytes of excess data in each frame, it accumulates to a mere 1.5 megabits per second[[7]](#footnote-7).

Using polling to represent HTTP against WebSockets is a little unfair in my opinion. Polling is the naïve approach of achieving real time, while WebSockets is designed for it. However, it shows my point: HTTP-headers have much excess data, but most of the time 99% of this data is irrelevant. Achieving a lot less excess data than this example is possible with HTTP through for example long-polling or Server-Sent Events, though nothing will use as little as WebSockets.

### Unidirectional

HTTP was finished in the 90s and it is still going strong. It’s rather impressive, but it is not unnatural that something that old will struggle with new trends. WebSockets is a protocol designed only for the purpose of bi-directional communication [24]–HTTP isn’t. In fact, no matter how you look at it, or how you try to hack, HTTP remains unidirectional.

As a result of this, most real time applications with HTTP have to use several TCP-connections (Figure 3‑13). Even with Server-Sent Events, one will need one connection to push the events to the client and at least one for the client to send messages to the server. With HTTP 1.0 some browsers used several TCP-connections to get more concurrent loading of web pages [9]. Now the same work-around is used to achieve simulated bi-directional communication. And as with last time this was the case, we need an improvement, namely WebSockets.



Figure ‑: Two TCP connections simulating bi-directional communication.

## WebSockets is still young

With new technology comes the almost everlasting issue of backwards compatibility. The use of the HTTP upgrade request-header ensures this for WebSockets. Implementing it, though, would have been a lot easier if all browsers supported it. As I write this, Internet Explorer has about 14% [29] of the browser market with IE8 and IE9 as the most dominant [29]. None of these support WebSockets. The rest of the major browsers does support WebSockets. But it will be several years before developers can safely assume that every single user out there uses a browser with WebSockets support.

To deal with this, applications have to fall back to other, supported techniques. In turn, this leads to more code. Luckily, frameworks like SignalR and Socket.IO abstract this away for developers. However, sometimes you want more control over the software you create than a framework supplies. And even with frameworks, you might end up having to do some workarounds for certain clients where the fall-back provided by the framework doesn’t suffice.

### Know when to use it

Writing an application with some real time elements is a different task than writing a full-blown dynamic, real time application. Examples of the two is an online newspaper with a live comment-section and a chat room, respectively.

In the first case using WebSockets may provide more pain than gain. Updates in a comment section does not need to be instant. Having the client poll the server will not change the user's perception of the application. Chatting is a different matter. Especially in a chat room, where several people talk to each other at once. This makes real time crucial to the users’ perception of the application. In turn, this makes it worth the extra effort of providing fallbacks for the browsers that don’t support WebSockets.

### Know how to use it

An important thing to realize is that WebSockets is not HTTP/2.0. It is a standalone protocol designed to fill the gap of HTTP regarding bidirectional communication. Failing to understand this might cause developers to use WebSockets in applications that don’t need it. An informative webpage, like Wikipedia, will probably never benefit from using WebSockets. You get less overhead in request-headers, but your application will have to serve mostly idle connections. The only real server to client communication is when the client request a new page (Figure 3‑14).



Figure ‑: No need for exchanging messages leaves idle connections.

Understanding your application’s environment is another vital aspect. WebSockets should handle proxies and firewalls gracefully [14]. But you might still encounter some problems. Especially if the traffic between your server and the client has to go through an older proxy along the way. Peter Lubbers indicates this in a blog-post from May 2010 [30], and even though this post is rather old, it might be a problem for some. His suggested way of handling the issue is the use of a secure connection (wss:// instead of ws://). In my opinion, this is a good practice since it makes data encrypted.

## The use of real time

The World Wide Web has seen many innovations throughout its lifespan. Each time something new comes around, it is hard to determine if it has come to stay. It is always a question of need: Do we really need this? Is it useful to me as a consumer? Real-time is no different from any other new developments; it has to be useful. And without some form of establishment throughout the web, nobody will ever notice it.

There is no doubt that real time content is useful in many aspects, and that in others it is even crucial. An auction site with time based auctions is dependent on delivering the latest bid to all users. Forcing their clients to refresh a web page manually to see the latest bid, would render it completely useless. On the other side of the scale we find web sites that utilizes real time to provide their users with a greater sense of convenience. Getting your friends’ status updates immediately can hardly be seen as crucial, but it does enhance the users’ perception of the experience.

Another interesting development is the increasing use of real time content provided by web sites that are more static. Most of this has to do with integrating social content like live comment-sections, trending articles and such. Again this is purely to make the content seem more dynamic and make the experience better for the users.

Looking at pure web page usage of real time, it is mostly about the users’ experience. But if we expand our perspective a little, it soon becomes clear how much of an impact real time might have on our lives in the future. Live video streaming is a common feature. But the technology is still in its youth, with buffering issues and broadband capacities as bottlenecks (Figure 3‑15). As the technological aspects evolve, I believe we will see a lot more usage of live video streaming across the web. Presumably, WebSockets, with its ability to stream binary data [14], will play a central part in future improvements to video streams.



Figure ‑: Hopefully, this scenario will be a thing of the past in a not too distant future.

## Conclusion

We have seen that WebSockets is superior to HTTP when it comes to bidirectional communication. Sometimes, though, it is not necessary with a bi-directional channel to achieve real time content. In some cases, HTTP-streaming techniques may be better than WebSockets. This is the case if most of the communication is from server to client, and the amount of header-data in the HTTP protocol is no cause for problems. If proxies may cause issues you can use long-polling instead.

Looking as these aspects leads me to say that HTTP methods may still be a better choice than WebSockets for some real time purposes. But ignoring backwards compatibility, there is no getting away from the fact that WebSockets is superior to HTTP for real time applications. After all, that was why WebSockets was created in the first place. Still, HTTP, with Server-Sent Events in particular, remains a strong alternative if you only need real time push. I believe that the other techniques will be outdated in a couple of years. WebSockets will replace them with some use of Server-Sent Events applications as well.

I believe that in the future, when WebSockets has been around for a long while, it most applications will prefer it. Furthermore, my opinion is that any future versions of HTTP will not incorporate WebSockets. The two will remain what they are; two separate things.

Social networks like Facebook, collaboration tools like Google Docs and other real time use cases are already widespread, and that will most likely not change any time soon. Real-time is here to stay, which is good because it provides vast, and yet unseen, possibilities.

Finally, my initial problem was the question of WebSockets’s position in the future of the World Wide Web. Do I believe it is the future? Well, the answer is both yes and no. Yes because it is the future for bi-directional communication applications. It will render HTTP mostly unused for the purpose as soon as the issue of backwards has vanished. Still, HTTP will remain king of the hill in “traditional” web applications that rely on requesting content in a unidirectional manner.

# Frameworks for real time web applications

Many frameworks for real time web applications exist, and it would be close to impossible to introduce them all here. I have chosen to present the 10 that, in my opinion, is the most widespread. Five of these will be selected for further studies.

## SignalR

SignalR is a library for the .NET framework that makes it easy for developers to add real time functionality to applications [31]. It has clients for both web applications, Windows Phone, ordinary Windows Store apps, other Windows applications and even Android an iPhone apps[[8]](#footnote-8).

The library provides two separate levels of abstraction for real time–Hubs and Persistent Connections. Persistent Connections are the most low level, extending the WebSocket API with a few events for reconnection and a few other things [32]. Hubs introduces even more abstraction on top of Persistent Connections. These utilize a Remote Procedure Call[[9]](#footnote-9) (RPC) API that allows for simple real time programming.

## Socket.IO

Socket.IO is a library for Node.js that *“aims to make real time apps possible in every browser and mobile”* [33]. Since it runs on Node, it only uses JavaScript as a language. Thus, the available clients are browsers, either mobile or desktop.

Using an event driven architecture, it closely resembles the WebSocket API. Socket.IO extends upon this though, by letting developers name their own events as well as using the traditional WebSocket events (see section 3.2.2).

## Atmosphere

The Atmosphere framework is a framework built for Java servers [34]. It uses a JavaScript client that supports all major browsers [35]. It provides an API that is more low level than the Hubs API of SignalR, but a little higher than the Persistent Connections.

On the client, Atmosphere provides something that looks very much like the WebSocket API, only with a few other events.

## Sails.js

Sails.js is a MVC (Model View Controller) framework for Node.js [36] (see section 5.3 for information on Node.js). It has full real time capabilities through the use of Socket.IO. The way this is accomplished, is rather interesting. Through what they call “transport agnostic routing” [37], controllers are allowed to automatically handle Socket.IO messages. Traditionally, this is something that is handled by separate code as with for instance SignalR in an ASP.NET MVC application.

## Play! Framework

The Play! framework, or just Play, is a MVC framework written in Scala and Java [38]. It is built on Akka, a framework that uses the Actor Model[[10]](#footnote-10) to help developers make distributed applications [39].

Play is real time enabled, meaning that it has constructs to help developers make real time functionality. These constructs though are a lot closer to bear metal than what is provided by pure real time libraries and frameworks.

## SockJS

SockJS is a JavaScript library for web browsers that gives developers an object that resembles the WebSocket API. This object can be used to connect to a server. Server side, there is a Node.js implementation as well servers for Erlang and Python [40]. Other servers are under development.

## Meteor

Many different types of web application frameworks exists, but none of these are similar to Meteor. It promises to simplify the process of making web application drastically [41].

Real time is the heart of Meteor, and SockJS is used in order to enable this. All code with Meteor is written in JavaScript, but interesting enough, it is not a Node.js framework (even though it runs on Node behind the scenes [42]).

## Lightstreamer

Weswit is the Italian company [43] behind the oldest real time framework I’ve been able to find (started in 2000 [44]). Lightstreamer is the name of the framework, and from the looks of it, it is also the most comprehensive.

The Lightstreamer server itself is written in Java, but through the use of an adapter based model, you can make server side code with JavaScript, C# or Java. There is also a long list of client side APIs including a JavaScript client for web browsers [45].

## Planet Framework

Planet is a real time framework for Python [46]. It is built with an emphasis on the development of social real time websites. The framework also provides its own Integrated Development Environment (IDE)[[11]](#footnote-11), specially design APIs and scaling mechanisms.

## XSockets.NET

XSockets.NET is an event driven framework for .NET [47] that adopts a model that somewhat resembles that of Socket.IO on the client. Server side it uses an approach that looks like the controllers of ASP.NET MVC.

The framework supports WebSockets and has fallbacks in order to support all major browsers.

# Servers

All web applications has to be hosted on a server. Various servers exist for different languages. In this chapter, I will describe some web servers that are relevant for the thesis.

## ASP.NET

Most applications for ASP.NET uses Internet Information Services (IIS) as an application server. The current version of IIS is IIS8 which was released alongside Windows 8 and Windows Server 2012 [48]. With it came support for the WebSockets protocol [49]. IIS is free as it is a part of the Windows operating system[[12]](#footnote-12). It is a Windows Feature that has to be activated.

For development, it isn’t always necessary to deploy an application to IIS. Along with the .NET framework and Visual Studio, developers get access to IIS Express. As with all products in Microsoft’s Express series[[13]](#footnote-13), it is a lightweight version of the original. It provides what you normally need while developing, but its performance is not enough for production.

## Java

Java applications can run on a number of different servers on any operating system. The most common are Jetty, JBoss and Tomcat [50].

Jetty is an open source server hosted by the Eclipse foundation [51]. It is considered a very lightweight server, but it has support for a wide variety of features.

JBoss is a product suite that includes several server types . The JBoss application server and Netty are probably the most known. The latter is used as Play Framework’s internal web server[[14]](#footnote-14).

Tomcat is an open source server from Apache [52] made for Java applications. Another server from Apache, the Apache Web Server, is used by the Planet Framework as preferred deployment server [53].

## JavaScript

Node.js is a new and exciting piece of technology that, for the first time, allows for server side JavaScript. It is based on Chrome’s V8 JavaScript runtime, and one of its primary goals is to help build fast, scalable network applications [54]. A web server is a fast, scalable network application. In other words, Node itself isn’t a web server, but it provides a simple mechanism for writing one [55].

## Writing your own

There is nothing wrong with building your own web server. In its simplest form it is just a main method with some mechanism for listening to incoming requests. The following example is from the “Getting Started” tutorial for Microsoft’s OWIN [56].



This isn’t completely bare metal, but it is fairly close. Within the “using” clause there is some sort of loop that listens to requests. Another class can handle these incoming requests, thus giving you a web server.

# Project part 1: Hands on development

# Methodology

This part will focus on the first two problem statements (see section 1.1). I will select and compare five different frameworks for real time web application development. Here, the focus will be from a programmer’s perspective. I will therefore implement a simple web application, which is described in section 6.2.

The selection of frameworks made in this part, will be the foundation for the second part, where each will undergo extensive load tests–comparing each framework’s performance (read more in the second part of the project on page 60).

## Selection criteria

As described in chapter 4, several frameworks for real time applications exist. Many have similar programming interfaces and features, but for the purpose of this thesis, only a few will be selected. Each selected framework must support the criteria listed in this section. A detailed justification of the selected frameworks is given in section 6.7.

### WebSockets support

The framework must support WebSockets. Each framework will undergo load testing (in part 2) where the individual transport mechanisms will be compared as well as framework performance. In order to answer whether or not WebSockets is an improvement to HTTP, it must be one of the framework’s possible transports.

### Fallback support

For the same reasons as to why WebSockets has to be supported, at least one HTTP-transport has to be offered as well.

### Presentation

The home page or GitHub repository of the framework should look presentable and give at least basic documentation and/or tutorials. Any piece of technology is hard to use if you cannot find out how to use it. Lack of proper documentation will make a considerable difference on the negative scale.

### Maturity

Real-time applications has been around for many years [57], but most of the frameworks for real time web applications are a lot younger. Maturity will still be given consideration, but if a frameworks offers something unique and potentially revolutionary, it may still be selected.

## Description of test application

An auction house will be implemented with each of the selected frameworks. The application has the following requirements specification:

* Users must receive real time updates regarding all global events.
* Global events are defined as all actions except from logging in and registering a new user.
* Users must be able to register an account and log in.
* Users must be able to add and remove items.
* Users can only remove an item added by themselves.
* An item does at least have the following properties: name, minimum price, info about who added it and who has the lead bid.
* Users must be able to place bids on all items, including their own.
* If the framework does not specify a specific template engine or other means of creating views, the application will utilize a common view implemented in Knockout[[15]](#footnote-15).
* MySql will be utilized as database unless implementing it requires substantial workarounds, that may cause the framework to misbehave.
* All applications should have tests covering the most critical aspects of the program logic.

## Discussion of use cases

Registering and logging in are actions that follow a traditional request/response pattern. These two use cases are present in order to test a frameworks capabilities regarding such communication. The remainder of functionality uses broadcasting; sending the response to all connected clients. This is the most crucial functionality of a real time application.

Another form of real time communication is so-called peer-to-peer communication (client to client) via the server. Implementing such functionality is basically the same as basic request/response, only that the response is sent to another client. As this adds no further, nor less, complexity, I regard it as unnecessary to test this aspect in this thesis.

## Evaluation

Evaluating each framework will be done during and after the development of the test application described earlier. The evaluation will be from a programmers perspective, shedding light on how the framework is to work with rather than how it performs. In order to get a complete picture, this process will follow a preset list of criteria described below. The term “user experience” in the following subsections refers to a programmers experience.

### Documentation

It is probably the most visited page in a programmers browser history while he/she is working with a new piece of technology. How the documentation is written and structured can make a lot of difference when it comes to a user’s experience. The code can be simple enough, but that means nothing if many frustrating hours is wasted looking for reference in the documentation.

Tutorials and examples are one of the most effective means to help a user get started. The presence of such will therefore be considered very positive. Other than that, the documentation will be evaluated based on structure, simplicity and content.

### Maintainability

Being able to write maintainable code has become alpha omega in modern system development. Any framework that introduces unnecessary complexity and dependencies between entities, makes maintainability difficult. In some cases though, there may be possible to write maintainable code even if it is very dependent upon other entities. Such scenarios may render unit testing impossible, but with proper tools, one can test entities using integration testing instead. Normally though, one prefers to have both unit- and integration tests as well as other forms of testing as well (for instance functional). I will measure maintainability based on how the natural structure of the application is with each given framework and how that impacts testability.

### Simplicity

A real time web application framework should act as a layer in a normal web application that handles communication. Other functionality such as session management, database operations and authentication are normally already present server side in a web application. I will evaluate whether a framework offers too much functionality or not.

Serialization an deserialization of data is central in server/client communication, regardless of real time or not. With JavaScript as the client language, the most common data exchange format is JavaScript Object Notation (JSON). In my opinion, a framework should handle this behind the scenes so that the user can focus on implementing the communication part of the application.

Keeping track of connected clients can easily lead to errors, and if you do it in an inefficient way, it can also cause performance issues. Each framework will be evaluated based on how it handles this problem. The more abstraction the better.

Abstracting this away from the user is usually a good thing, but it depends on how the clients are offered back to the user. If one has to write complex or tiresome code just to send a message, the client handling can be as foolproof as it wants–it does not make the overall experience any better. I will therefore also look at what constructs are offered by the frameworks to send messages to clients.

### Browser support

This criteria is directly linked with what protocols a framework offers. WebSockets is not supported by older browsers, nor are some of the fallbacks. How a frameworks detects what transport can be used is crucial for the overall experience. A proper real time framework should handle transport selection gracefully in the background. It should be possible to choose a “lower lever” transport though[[16]](#footnote-16). For most use cases, this is not a demand, but considering the work I will do in part 2, it is important to me.

### Maturity

A lot of real time frameworks are a work in progress and has not yet reached a stable version. Such frameworks often change drastically causing users to change their applications completely to keep up. Hence, these frameworks are not suited for production code.

Another way of measuring maturity can be to look at projects that uses a framework. If nobody is using it, there is often a reason. Furthermore, one can look at the amounts of bugs and errors that appear during development. A lot of errors directly related to the framework’s core, is often a sign of immature code. It is also often a sign of unmaintained code, which many would say is even worse than immaturity.

### Other criteria

There may be other criteria that some deem more relevant than the ones listed above. All those I have listed are based on what is most important in my opinion. Other criteria that I considered were:

* Coding environment: What IDEs a framework allows you to work in can make a difference for some. Especially regarding support for debugging and code analysis. I believe that these things does not matter if a framework is the best tool for a job.
* Community: A large community can be an indication of a mature framework. It does not guarantee it. Furthermore, a small community does not mean that the framework isn't good.

## Limitations

There are some limitations regarding what kinds of frameworks that are suitable for this thesis.

### Cloud based solutions

If my work only consisted of comparing the usability of different frameworks, cloud based solutions could easily been a part of it. But due to the load testing aspect, I cannot test cloud based frameworks. It is impossible to have an equal test between a framework, whose server I do not control, and a normal framework running on a local server.

### Rapid development

Many frameworks for real time web applications have appeared over the last couple of years. Most of these are not among the best, and almost all are very early in the development process. Because of this, there may be some frameworks not considered for this thesis that suddenly has become one of the “buzz-words”. An example of this is the Java-framework Atmosphere (see section 4.3).

## Other choices

This section describes other choices related to this part of the project.

### Common UI

The application that is going to be implemented is the same for all frameworks. It will look the same, and the functionality will be the same. To keep it as simple as possible, I will strive to share as much as the user interface code as possible. For that purpose, I have created a common user interface using Knockout that will be used when possible. If the framework under test comes bundled with some other way of creating the user interface, it will be used instead.

### Choice of database engine

What database the application use is not relevant for the real time aspects of it. Nonetheless, the application I will build has a database component. To keep this as similar as possible, MySql will be used if it is supported. If possible, all application will work against the same database.

### Functional testing

As each application will appear the same to a browser, I will use a common functional test case. This is possible as long as the user interface is similar and all frameworks use the same database table. The tests will use Selenium WebDriver [58] to drive real browsers through a series of tests. Java will be used as implementation language for the tests.

## Selected frameworks

I selected the five different frameworks for some different reasons. There are three “pure” real time frameworks and two “real time enabled” frameworks. A pure real time framework has real time as only functionality. A real time enabled framework has other main functions, but can offer real time functionality. This section describes why each of the five were selected.

### Socket.IO

Node.js is rapidly gaining popularity. Using JavaScript on the server is an exciting thought, and the notion is changing the way developers think of the language. Because of this, it was only natural to chose at least one Node.js based framework.

Several libraries and frameworks exist for real time with Node (see chapter 4). Of these, Socket.IO stands out from the crowd. It seems to have the largest community, and it gets a lot of attention.

Socket.IO has its own homepage that has some documentation. It is not a lot, but everything is presented in a orderly fashion. The project uses GitHub for further documentation and wikis. In my opinion, the library looks very lightweight, which would explain why there isn't a lot of documentation.

The library offers some fallbacks to ensure compatibility with all browsers. WebSockets is the preferred transport. If it is not supported, Socket.IO will fall back to one of the following transports [59]:

* Adobe Flash Socket, which uses Flash to establish a TCP connection between server and client.
* Ajax long-polling
* Ajax multipart streaming (Xhr-streaming)
* Forever frame
* JSONP Polling. Polling with JSONP allows for cross-domain requests.

Currently, the library is in version 0.9, but a 1.0 release is around the corner. Nonetheless, it is considered stable and is used by several projects[[17]](#footnote-17)

### Lightstreamer

Lightstreamer is completely different from all the other frameworks I have found. It is a commercial product from a rather large, European company with customers worldwide.

Being operational since 2000, it is the oldest framework in this thesis by far. Seeing how many years of experience plays out against a lot of the new ideas of the more modern frameworks, will be interesting.

Weswit is the name of the company behind Lightstreamer, which is their only product. This means that all the customers they list, are using Lightstreamer. With names like NASA and Sky on the list, it is obviously a trusted and mature product.

The framework seems thoroughly documented. Separated documents for each API as well as some general concepts, leaves a good impression.

Lightstreamer supports the following fallbacks when WebSockets is not available:

* Http-streaming.
* Http-polling.
* Polling with WebSockets can also be selected.

### Play Framework

One of the questions I seek to answer is whether you need a framework at all to implement scalable real time applications. Play offers some help for real time, but not a lot. It is close to working with bare metal real time development.

Not many web application frameworks I have seen promotes real time features like Play does. Hence, it stands out from the crowd in this matter. The framework has become more popular recently, and some serious actors are using it in production[[18]](#footnote-18).

As a developer, you get two helper classes for real time functionality. One for WebSockets [60] and one for Comet (Http-streaming, [61]).

A clear answer to the initial question will be if the development process with these helpers, turns out to be cumbersome. Then, bear metal will obviously be even harder.

### SignalR

SignalR is one of the few libraries made specifically for .NET. Additionally, it is amongst one of the libraries that has gotten the most attention in recent time.

Two developers on the ASP.NET team started the work on SignalR in 2011[[19]](#footnote-19). The fact that Microsoft supports the project is thus no surprise.

SignalR builds on concepts that are familiar to .NET developers, like the use of Inversion of Control (IOC) containers [62] and Nuget[[20]](#footnote-20).

The library uses an interesting form of abstraction in its programming model. The use of hubs and RPC will be very interesting to get a closer look at.

WebSockets is the preferred transport. If it is not supported, SignalR will fall back gracefully to one of these transports:

* Server-Sent Events
* Forever Frame
* Long-polling

I also have to mention another reason for choosing SignalR. My first experience with real time technologies was through a project with SignalR in 2012. I really sparked my interest. Without this experience, I would probably have written my master thesis with some other topic.

### Meteor

Meteor is not a completed framework. Nor is it close to completion. Nonetheless, I have chosen to test it due to a number of reasons.

First of all, it is radically different from any other similar framework. I uses JavaScript both on the client and server, which is possible through the use of Node. It is not a Node.js framework though.

Meteor tries to share most of its code between the server and the client, blurring the line between them. If this is an application model that is better than the traditional, I intend to find out.

Another interesting feature is how real time is closely integrated in the core of the framework. Actually, a lot of features seems tightly coupled to the core as of now. MongoDB is the only supported database, even though support for others is planned [63]. I will implement the test application using MongoDB. There exists an unofficial add-on that allows you to use MySql [64], but I will not utilize this.

Windows is currently not supported by Meteor. An unofficial fork[[21]](#footnote-21) of the project exists for Windows [65]. I will use this fork for my test. The only real difference is the command line tool that has to be compatible with the Windows command line.

Meteor uses SockJS [42] under the hood to perform its real time updates. SockJS received support for WebSockets only a couple of days before I started my work with Meteor. If WebSockets is not supported, SockJS falls back gracefully to one of the following transports [66]:

* Server-Sent Event (Opera only)
* Http-streaming (iframe or Xhr depending on the browser)
* Long-polling
* Polling (just for really old browsers)

# Results

This chapter gives an in-depth description of my experience with each of the frameworks. Each framework start with an introduction describing some specific choices made for it. For instance what server was used or whether it used some uncommon approaches. The subsections of each describe details from the development process based on the criteria described in section 6.4.

## Socket.IO

As most examples with Socket.IO shows it in conjunction with Express, I used this in my test application. Express is a web application framework [67]. The only aspects of the application that uses Express is the server itself, including serving static files. The Socket.IO application makes use of the common user interface described in section 6.6.1.

### Documentation

With Node already installed, downloading and installing Socket.IO into a project is simple. One simple command is all you need: “npm install socket.io”. While it goes without saying that you need to have Node installed, I think there should have been a link to where you can get it nonetheless.

Documentation is not something Socket.IO has a lot of. Considering the size of the library though, this is not surprising. What it has covers everything in enough detail. Diligent use of examples, makes it easy to read and understand.

The structure isn't perfect. Some pieces reside in the readme on GitHub [68], while you find other pieces in the wiki [69]. There are some logic behind this, with the API documentation in the readme, and other aspects in the Wiki, but there is no obvious flow when browsing it.

As of writing, the current version of Socket.IO is 0.9. The documentation states that it is for the "upcoming" 1.0 release. For the entire duration of the work with this thesis, this has been the case, but the version has yet to be released. I have run into no problems regarding this though, which leads me to believe that most of the API is set already before 1.0.

All the examples are very small, which is good for readability. I missed some larger examples though. Something with more than one html file and a little more complex functionality, would have been beneficial.

### Simplicity

In the spirit of Node[[22]](#footnote-22), Socket.IO is nimble and lightweight. Classical web application elements include authorization and session management. Both of these can be tricky to handle with real-time frameworks, but Socket.IO provides some simple mechanisms [69]. It is actually oblivious to sessions, leaving it up to the server library you use to handle this [70].

Some frameworks and libraries offers a lot of configuration. Usually this is a good thing, but sometimes it is easy to get lost in translation. Socket.IO has many options that you can tweak, but it is far from needed [69]. You manage settings using code instead of one or more files. I prefer the latter, but since the language is JavaScript, one can simply store configuration data in a JSON file and parse it at run-time.

A common use case for a web page is to have several, separated real time features. IGN.com[[23]](#footnote-23) has functionality to show current readers of article, as well as real time comment sections. Socket.IO allows developers to register different channels, or "rooms". This allows implementation of such functionality to be simple.

Sending data back and forth is a dream with Socket.IO. Behind the scenes, it uses JSON. As a result, you can send a normal object and receive it in the callback on the other end (Example 7‑1).



Example ‑: Serialization of objects happens behind the scenes.

You have to be cautious about sending Date objects though. A common problem with JSON [71], is that Date objects don't deserialize too well.

Another nice feature is that you don't need to relate to the concept of a client. Instead, you deal with either one or multiple sockets (Example 7‑2).



Example ‑: Simple response and message broadcast with Socket.IO.

Because of this simple abstraction, sending a message never require more than a single line of code. Socket.IO also provides constructs for sending messages to a specific client. To do this you need the id of the particular client's socket. Associating this with for instance a username, has to be handled manually of course.

A final note on features offered by Socket.IO is that it closely resembles the WebSocket API. While this API only has three events, Socket.IO lets you define your own. These act as a further separation of the “onmessage” event. The result is a code structure where the flow of the application shines through without the need to dig too deep.

### Maintainability

Socket.IO follows the programming principles of Node and provides an event based model. While all code written for Node can be testable, testing events is a little trickier. The event driven architecture conceals all logic regarding sending and receiving within callbacks. Luckily, there are ways to work around this.

One option is to separate all code within the callback to its own module (Example 7‑3). Then you can write tests for this module in separation, as it does not know that it is an event that calls it.



Example ‑: Separate logic within callback to separate module.

Another option is to put the callbacks themselves in a separate module. This makes the code a little less readable in my opinion, which is why I went with the first solution.

To test that specific callbacks execute as expected, one has to use either integration or functional testing. As Socket.IO provides a client library for Node, this is simple to achieve without too much complications (Example 7‑4). This method can also be used for unit testing purposes by injecting mocks and stubs into the module with the Socket.IO logic. In its essence, it will still be an integration test since you have to start the server, but at least you get to test it in isolation.



Example ‑: Testing events with Socket.IO.

Modules can be challenging to keep small enough. Larger modules, just as larger classes in object oriented languages, are harder to maintain. This also applies to the routing of events with Socket.IO. Even if you keep the code within each event's callback short, it can quickly become a mess if you have many events. Using the namespace construct can help provide a stronger separation of concerns in such cases. However, this also introduces extra overhead, which leads to the fact that there is no perfect solution to this problem.

### Browser support

As promised, the library supports all major browsers with no quirks. Transport mechanism is selected automatically during the handshake process of a connection. However, it states that it supports HTTP-streaming, but I did not get this to work. After some investigation, I found out that it has been removed from the core [72].

### Maturity

GitHub claims that Socket.IO version 1.0 is the "upcoming" release, which has been the case for over a year. Commits to the repository has been varying and it had a long dead period. This dead period seems to coincide with another project from the same people, Engine.IO (see commit pages on GitHub [73], [74]).

Engine.IO is a more low level implementation of Socket.IO. Socket.IO is actually an extra abstraction layer on top of Engine.IO. It is reasonable that the creators would wish to separate the most low level functionality into its own library. This makes it easier for other developers to build upon it to make other frameworks.

That there are no more commits to a project is a classic sign of a "dead" project. With the activity on Engine.IO and recent activity on the Socket.IO project, I do not think this is the case for Socket.IO. Nonetheless, little activity does not mean that the product is immature. In this case it is completely opposite as Socket.IO is stable and well suited for production environments. Since the documentation for the 1.0 release is already out, one can trust that no major changes will come soon.

There is the question of Node.js itself though. It too hasn't reached a 1.0 version, and the community surrounding it isn't the largest. As a result, the community surrounding Socket.IO is even smaller. Many questions on Stack Overflow is about Socket.IO though, a clear indication that it is widespread. The tendency with Node is that the community is growing, and more and more developers see it as a technology for the future - an opinion I share.

## Lightstreamer

For the test application I used the free version Lightstreamer offers [75]. In a real life scenario, the Lightstreamer server would only handle real time aspects. My application uses it as a web server as well. It also has database communication. As this is not the intended use of the Lightstreamer server, I have not written about this in the following sections. The Lightstreamer application does not make use of the common user interface described in section 6.6.1.

### Documentation

Getting started with Lightstreamer is a simple and well documented process [45]. The rest of Lightstreamer is also extremely well documented. It is, by far, the most comprehensive documentation of all the libraries and frameworks in this thesis. With such a large scale, it is clear that Lightstreamer is a framework rather than a library like Socket.IO.

With a lot of documentation comes a great responsibility to organize it. Weswit does this well. Each of the many APIs has its own document, while a common document covers all general concepts. These concepts should maybe have been given more room in the documentation. The documentation only offers a single page to one of the most central aspects of the framework: the different subscription modes[[24]](#footnote-24). I didn't get a good sense of what the difference between these were before I found a forum post that explained it [76].

Another shortcoming is the almost complete lack of tutorials - only one exist [77].. That would have been ok if the samples were well documented and written, but this isn't so. Many samples accompany the framework (Figure 7‑1), and these illustrate different uses. Understanding them at a conceptual level is easy enough, but when you start digging into the code, trouble starts.



Figure ‑: The contents of Lightstreamer’s "demos" folder (for JavaScript client)

The samples are "documented" through comments in the code, and these are not abundant. Furthermore, the code is rather messy and hard to follow. I had a hard time figuring out what parts was related to Lightstreamer and what regarded the user interface. As a result, I spent many hours debugging both the client- and server side code in order to understand what was going on.

Lightstreamer’s server comes with a large configuration file where you can tweak it's performance to fit your needs. There isn't too many options, but the file is easy to get lost in. This is because they chose to write all documentation regarding the various options within the file itself as comments. As a result, it is a lot larger than what it could have been. Some xml elements are also hard to spot because they are commented out.

### Simplicity

Compared to the other libraries and frameworks in this thesis, Lightstreamer is huge. It even feels larger than Play and Meteor (see sections 7.3 and 7.5), which both are full featured web application frameworks. Socket.IO provides real time features alongside a web application, running on the same server. This is not the intended use of Lightstreamer. It is meant to be a stand-alone server rather than a layer in a normal application stack. (Figure 7‑2).



Figure ‑: The Lightstreamer server's purpose is real time only.

Other parts of the application (other servers) interact with Lightstreamer through an adapter infrastructure. There can be any number of data adapters and one so-called metadata adapter. Data adapters handle subscriptions and sending updates. The metadata adapter handles incoming messages from the clients, session management, authorization and Quality of Service (QoS) [78]. This is a nice separation of concerns, but it also introduces some complexity. There has to be some form of connection between the adapters, meaning that they either have to depend on each other, or have some common dependency. (Figure 7‑3).



Figure ‑: How the various adapters interact.

Lightstreamer uses an application model that resembles a Service Oriented Architecture (SOA) [79]. More specifically, it uses a publish/subscribe model. It doesn't follow all SOA principles though [80]. Some examples of this is that it is not discoverable and the strong coupling between clients and server.

Clients subscribe to different items rather than listening to certain events. To me, this feels a little old fashioned, as it creates a very tight coupling between the DOM and the items. The items, which "live" on the server, needs to have fields that corresponds to the fields in the DOM client side (Example 7‑5).



Example ‑: Tight coupling to the DOM.

One can work around this by using a "message" DOM element with only one field. If this field receives updates in JSON format, you can mimic an event driver architecture. There are several drawbacks to this technique though. Serialization and event routing has to be handled manually. The first is manual no matter (Example 7‑6), but converting to JSON adds another layer of complexity. It also makes you unable to benefit from the different subscription modes.



Example ‑: Serialization of data bound for clients.

Lightstreamer is not intended to used in this way though, and doing it may influence performance. You will be better off using the subscription modes. These are actually powerful, allowing for updates of single fields, deleting and adding items. However, it is a lot more complex than the more "modern" approach given by for instance Socket.IO and SignalR (see sections 7.1 and 7.4).

The tight coupling to the DOM makes using popular MV\*[[25]](#footnote-25) frameworks like Angular or Knockout little useful. As a result, it may be hard to integrate Lightstreamer into an existing application where such frameworks are present. With a new application, Lightstreamer can fall through because of this in my opinion. Either that, or the application has to have a clear separation between the "Lightstreamer-pages" and the rest.

As I mentioned earlier in this section, there has to be some connection between the metadata adapter and the data adapters. Directing an incoming message to its destination is something you have to handle yourself. The same goes for concurrency. The following figure shows the flow in my simple test application from an incoming message, to a broadcast is sent (Example 7‑7).



Example ‑: The flow of a message from it is received to a broadcast is sent.

Lightstreamer is broadcast by default. Given its old age, it is not unnatural that the main use case is push. In fact, it performs best if it can function purely as a publisher (in SOA terms). It would be even better if some other entity than the clients functions as producers. This would let the Lightstreamer server do only push, which is what it handles best.

A clear indication of this is the fact that the default update mechanism is broadcast. To send to individual clients, even back to the caller, you have to have a separate subscription for each client. Request/response features is, in other words, not Lightstreamer’s strong suit.

### Maintainability

Despite the complexity, there is nothing stopping you from writing maintainable code. Both data- and metadata adapters are based on interfaces, so that you can mock them in tests. If an application manipulates the data injected to the adapters by the Lightstreamer server, this is also testable. Since the server uses method injection for this, it is just a matter of creating some dummy data and inject it in the test.

The samples uses an ExecutorService [81] to handle concurrency. Testing methods that uses this can be a little messy, but no more than testing callbacks of Socket.IO events. As long as you can inject a mock of the listener that the ExecutorService triggers, you are good to go. In other words, the seemingly preferred method for Lightstreamer, is the same I used for Socket.IO (see section 7.1.3).

Client side on the other hand, is an whole other story. When testing this code, the first thing you'll want to do is to mock out dependencies to Lightstreamer. But before that, you need to understand how this code works. With only a minified version of the code available, this is easier said than done. What Lightstreamer does client side is somewhat advanced, so mocking its behavior isn't the easiest of tasks.

Still, as long as you keep a firm line regarding separation of concerns, on both server and client, you should be able to write maintainable code. After all, Lightstreamer's intended use is to be a separate server in your application stack. Keeping this in mind, you would want to keep this part as light weight as possible. Other aspects, like database handling and such, should be left out of the Lightstreamer server. But even with all of these precautions in place, the real time specific code of Lightstreamer will be more comprehensive than most other real time frameworks.

### Browser support

Lightstreamer supports all major browsers without any trouble. Fallbacks are handled gracefully behind the scenes so that the optimal transport is always used. One thing I have trouble understanding though, is the transport mechanism that uses polling over WebSockets. As far as I can tell, there is no real benefit to this as any browser that supports it, also supports streaming over WebSockets.

### Maturity

With 14 years of experience, it is safe to say that Lightstreamer is a mature product. In a way, it has become too mature. 14 years ago the focus of real time applications was pure push. This is still what Lightstreamer does best, as the mechanisms for receiving input from clients are somewhat complicated.

Lightstreamer uses a completely different approach for real time than any of the competition. The push orientation is probably due to its old age. But the use of a separate server for all real time functionality is not a bad thing. Depending on the needs of an application, this technique may be the best no matter what framework or library you are using. With Node based solutions for example, the most likely approach is this, since very few applications uses only Node on the server.

Weswit is actually one of the pioneers for real time technology [82]. Their main problem now is to keep up with modern day trends. The competition are lightweight libraries rather than large scale frameworks. In my opinion, they would benefit a lot from making a "light" version of Lightstreamer that introduces a higher level of abstraction. Using for instance an event driven model, this would result in a simple library. Decoupling it from the Lightstreamer server may also be beneficial. This would make it more attractive for Java developers that want some simple real time features in their application.

Nonetheless, the company has a large customer base, and they have received a lot of appraisal [83]. Changing something that many people use and love, isn't always a smart thing. The expression "If it ain't broke, don't fix it" exist for a reason. What I think they should do though, is keep everything they got, but also offer what I suggested above.

## Play Framework

I used version 2.1.1, which was the newest at the time. Play comes with a server side template language for rendering of HTML. I minimized the use of this in order to use the common user interface described in section 6.6.1. Real time communication require JavaScript on the client side. This means that the server side template language is not applicable.

### Documentation

Obtaining Play is a simple and well documented process [84]. Following installation, the documentation provides a step-by-step guide to get you familiarized with Play. There are also two tutorials: one simple and one more comprehensive [85]. These are mostly easy to follow, but I ran into some issues with links to certain files that didn't work anymore.

As Play lets you write in either Java or Scala [85], [86], it is a good thing that the documentation is separated accordingly. The exception is the getting started part, but this covers no code. Splitting this as well would have been a duplication rather than a separation. With this separation, developers can focus on the language they want and have no knowledge of the other.

Play is an open source framework with a rather rapid development process. Instead of referring you to a change log, Play's documentation lets you browse previous versions. This simplifies the process of migrating to a new version a lot.

The framework builds on many other technologies from third parties. Play's documentation does a good job of pointing the reader to resources regarding these technologies. It also explains these itself, but not in too much detail.

Using examples, the documentation is easy to read and learn from. Learn by doing is, in my opinion, the best way. With Play, there was little need for diving into the source code. I just followed the examples.

### Simplicity

Play is a MVC framework. Hence it resembles most other frameworks in this genre. Still, it has some interesting features that separates it from others:

* A route configuration file that gives you IntelliSense[[26]](#footnote-26) in the editor. It gets compiled at run-time, providing you with feedback just as with code files. You have to manually list all routes though. Compared to the route configuration of ASP.NET MVC, this is a little elaborate.
* Hot code push allows you to run the server once. Each time you change a file, the code recompiles and you page refreshes.
* The template language is Scala. It resembles the template language of other frameworks. Compilation errors from all template files, are also shown as with other files (Figure 7‑4)
* Play lets you use Akka Actors. Actors is an old, mathematical concept [87], but it has become increasingly popular recently. As with the rest of Play though, you are not forced to use it.



Figure ‑: Compilation error with Play.

The real time features of Play are close to bare metal, but it provides some nice abstractions. There is two utility classes, one for WebSockets and one for Comet (HTTP-Streaming with forever frame). The WebSocket class gives you two channels: in and out. As WebSockets is bidirectional, this is reasonable (Example 7‑8).



Example ‑: Plays WebSocket provide an "in" and an "out" channel when it's ready.

With the in-channel you get access to the "onclose" and "onmessage" events of the WebSocket API. "Onconnect" is handled by the "main" WebSocket class (the onReady event in Example 7‑8). The out-channel handles sending of messages only.

Since HTTP-streaming is from server to client only, the Comet class has no in-channel. A separate route has to be set up to handle incoming messages. Using a POST route is the most applicable. Outgoing messages uses a similar out-channel as the WebSockets class provide. One handy thing the Comet class provides is a callback for disconnects (Example 7‑9). This, along with the out-channel and incoming POST requests, you have a similar API for Comet as for WebSockets.



Example ‑: The Comet class provides an event for disconnection handling.

On the client you stand without support. To access WebSockets, the standard WebSockets API is used. Comet depends on the presence of an iframe on the client in order to receive messages. The utility class on the server wraps outgoing data into a function within a HTML script-tag. You specify the name of the function, but it has to be globally available or provided an absolute path[[27]](#footnote-27).

Both of the utility classes allow for sending either strings or Jackson JsonNodes[[28]](#footnote-28). JSON is the natural choice since it allows for easy access on the client. Play even has a helper to extract a request body as a JsonNode, which fits well into the real time stack of the framework [85]. There are some drawbacks though, as serializing more complex objects yields more overhead (Example 7‑10) With a lot of nested arrays and objects, results in many loop constructs. Overall, this can lower the performance of the application.



Example ‑: A lot of work to serialize complex objects to JSON.

Besides the manual fallback handling and serialization, you also have to keep track of clients yourself. A hashmap is probably the best solution for this. No matter what you use, you need a reliable source of an unique id for each client. Luckily, Play gives you a connection id that can be used for this purpose. The value in the hashmap has to be some kind of entity that contains the means to send messages. I solved this by separating all common methods into an abstract super class. Then I made wrappers for the WebSocket out-channel and the Comet class that inherited from this super class (Figure 7‑5). With these two transports only, this is a simple solution. If I had introduced long-polling as well, it would have been much more complicated.



Figure ‑: Socket helper class hierarchy.

### Maintainability

Writing clean and maintainable code with Play is mostly the same as for any other good framework. There are some small issues though.

The utility classes for WebSockets and Comet are both based on events. They use Java's counterpart of JavaScript’s anonymous callbacks (Example 7‑11). This leads to the same issue as the anonymous callbacks of Socket.IO's events. With Socket.IO I proposed a number of solutions (see section 7.1.3). These can also be provided for Play. Another aspect where this problem surfaces is when working with Actors. You can build huge hierarchies of Actors that work together. Testing this is not straight forward, but again, the above solution is a simple workaround.



Example ‑: Anonymous callbacks in JavaScript and Java.

All applications needs to have tests that work from end to end. An event based architecture makes this even more important. Play offers some utilities to help with this issue.

You can make a fake server and use Selenium to do functional tests against it. It is a handy feature, but I don’t see that you have too much need for it. Most development environments features a shared development server that tests can run towards.

Play also supplies a simple way to use an in-memory database to test your models. A traditional Java web application using Spring MVC would have a separate data access layer. Play encourages the use of Ebean which puts data access into the models themselves. Testing is similar though, as a normal process with Spring MVC is to set up an in-memory database just as with Play. This process in Spring require a lot of configuration, while Play gives you a single method to call [88]. Play also offers methods to help you test controllers, templates and even the routes, meaning that the hole stack is testable.

### Browser support

Using just WebSockets and Comet provided support for all major browsers. Handling fallbacks manually can be achieved in two different ways. I chose to use the user agent string to determine what browser the client was (Example 7‑12). In retrospect, this was not the best solution. Instead I could have the client find out if it has the WebSocket or MozWebSocket (for Firefox) objects. If it doesn’t, it doesn’t support WebSockets. This information could then be used to connect to the server. I used this method in the load tests (as you can see from the source code on GitHub [3]).



Example ‑: Code to determine WebSockets support based on browser type.

### Maturity

The framework was introduced in 2009, but Play as it is today came out in 2012 as version 2.0 [89]. With it, the core had a major overhauling. Scala support came through an external module in the 1.x versions. The 2.0 version integrated Scala with the core, providing full support for the language.

One might say that the framework matured a lot with this change in the core. It made it easier to provide full support for both Java and Scala as well as providing more consistency in the way you build an application. But, this also means that the core is quite new and unproven. Furthermore, Scala is a relatively new language. Although it’s popularity is increasing, it still isn’t used that much. The TIOBE Programming Community Index , places Scala as number 42 in March 2014 [90].

Nonetheless, Play keeps gaining popularity, and it is considered a stable piece of software. This is my opinion as well, but there were some issues. Outdated versions of third party software and broken links on the homepage was the most prominent.

There is no arguing that Play follows modern trends at least. With support for CoffeeScript, LESS and other popular languages and tools, it offers a lot of freedom. It may not support the same stability as for instance ASP.NET or Spring, but it strives to make development clean and simple.

## SignalR

This part of the thesis does not require high performance of the server the application runs on. Hence, I chose to use IIS Express 7.5 to host the application. An ASP.NET MVC 4 application wraps the SignalR components. No other communication than serving static files uses the MVC specific aspects of the application. The application uses the common user interface described in section 6.6.1.

SignalR offers two levels of abstraction: Persistent Connections and Hubs. As Hubs are most high level, and what most people are likely to use, I focused on this aspect only in the test application.

### Documentation

SignalR's has a very comprehensive introduction part. It talks you through all from what it is to making your first application. As with all other .NET applications, you use Visual Studio. Nuget is the package manager, and the documentation does a good job describing how to use these with SignalR.

During early development, the only documentation was up on GitHub [91]. Even so, there were still a lot of documentation and many, simple examples. There are still some topics that are covered only on GitHub, but most of the documentation now reside in one place [31].

The new documentation has a lot of tutorials and examples. It is way more comprehensive than what it was on GitHub. Just as Play does, SignalR's documentation lets you browse previous versions, though not with the same granularity[[29]](#footnote-29). The content itself is also separated into logical bulks with lots of examples. A nice feature is that all class names appear as links to the class reference[[30]](#footnote-30).

SignalR is the only framework that covers IOC in its documentation. This is probably because it is the only framework that has its own dependency resolver. It is therefore only natural to tell you have to use this or change it to something else. Still, the term "IOC" isn't even mentioned in most of the other frameworks' and libraries' documentations.

There are some things I missed in the documentation. Hubs are thoroughly , but Persistent Connections are not covered in detail. The GitHub pages covers this though, so it may be in the making for the new pages. Furthermore, I missed some documentation regarding testing hubs. Finding information of this required me to search elsewhere [92].

### Simplicity

The developers made SignalR to be compatible with ASP.NET web applications. Whether you use ASP.NET MVC, WebForms or Self Hosting, using SignalR is the same experience. It is compatible with mechanisms for authentication, IOC and sessions provided by ASP.NET. A developer only has to find out how to match those things with SignalR. This is an easy task as there are many examples for almost everything.

SignalR comes in two different forms: Hubs and Persistent Connections. I have focused on hubs, as I predict this to be the most common usage of the library. Persistent Connections are a lower level of abstraction than Hubs. It resembles the WebSockets API, but with some extra methods for handling sending, broadcasting, reconnecting, groups, etc. A benefit of using this API is that you can access the real time part of an application from several places. This allows for instance a controller action method to broadcast data to clients.

The Hub API strives to be very simple, providing a RPC model which makes the code simple and understandable. A Hub exposes all its public methods to the client and they can be "called" directly (Example 7‑13).



Example ‑: SignalR's RPC model makes it simple and understandable.

C#'s convention is to have method names with capital first letter, while JavaScript has the opposite. As a result, the name of the method or function you "call", doesn't match the actual name[[31]](#footnote-31). It takes a little while to get used to, but in the end it makes sense to follow conventions.

Hubs manages clients through an abstraction that is the most straightforward and easy to understand I've seen. You have two main choices on how to get data out to the clients. Either via returning from a method. This sends the returned data back to the caller, just like a response to a request. On the client, the returned data is sent to the "done" callback of a promise, just like a jQuery get call. Clients can also specify functions that are callable from the server in the same RPC style as clients call server methods (Example 7‑14). Furthermore, the "Clients" object also has constructs to access the Caller of a method and to access groups. Groups are similar to Socket.IO's namespaces (see section 7.1.3).



Example ‑: Clients can specify functions for the server to invoke.

Each method handles serialization behind the scenes using JSON. This means that you can send objects back and forth. You can also annotate properties to shorten property names (to reduce bandwidth). To get this back in readable form on the client though, you have to write some manual code [93]. The serialization even handles date objects. I ran into a problem with this though, as dates deserialized as UTC time rather than GMT + 1. Problems with dates therefore persists, even though it seems like SignalR has solved it at first glance.

### Maintainability

SignalR is just a real time layer within a normal web application, just as Socket.IO. This means that the only entities you have to test are the Hubs or the Persistent Connection classes. While I will focus on the Hubs, most of the principles I will discuss applies to Persistent Connections as well.

The Hub class, which is the abstract class all Hubs inherit from, is made with testability in mind. Testing a Hub class directly though, is not feasible. This is because you have to mock the "Clients" object as this has dependencies you don't want in your test. Such a scenario is not uncommon when writing tests. What you do is write a class that derives from the Hub you want to test and make mocks within it [92]. Setting up tests like this allows you to verify all logic within a hub. (Example 7‑15).



Example ‑: Unit test code for a SignalR Hub method.

The site also claims that you can test that the Hub sends out correct values as well. I was not successful in doing this though. This is not critical, as such functionality is more usually tested through functional tests.

### Browser support

All major browsers are supported by the library. As with all the other frameworks, you can specify what transport to use. However, SignalR makes an annoying choice for you. If you provide “foreverFrame” (streaming) as transport in a browser that supports Server-Sent Events, you’re not allowed [94].

### Maturity

The development has gone on for a couple of years, but the library didn't reach 1.0 until early 2013 [95]. One of the benefits of being a Microsoft supported project, is that it has a strong team working with it. As a results, a 2.0 release surfaced within a year of 1.0 and both 1.x.x and 2.x.x versions receive updates frequently.

Another sign that Microsoft provides heavy support for SignalR is that it has become an integrated part of the ASP.NET framework. Newer versions of Visual Studio has project templates for it. The IDE itself actually uses SignalR for hot code push functionality [96]. Furthermore, a lot of attention goes into promoting the library at conferences. All this shows that Microsoft has confidence in the technology.

After the 1.0 release, I find it unlikely that the code platform and structure of SignalR will undergo any drastic changes. There may be additions, but overall, the current form of the library is very stable.

One sad thing about SignalR is that you must use IIS8 to harness its full potential. IIS8 is the only way to get support for WebSockets. This isn't due to the implementation of SignalR but due to limitations in the Windows operating system.

## Meteor

The unofficial Windows version is a little behind the current version of the official version. During my work, this was not the case, and both versions was 0.6.6.3. Meteor did not have support for MySql at this time. I used the supported MongoDB instead. This means that I was not able to use the common functional test case described in section 6.6.3. Meteor also have integrated support for a client side template language. The common user interface described in section 6.6.1 was thus not used.

### Documentation

Even if Meteor is available for Mac and Linux only, the documentation mentions the unofficial Windows version. It is a little hidden though, since you have to go via the "supported platforms" site [97]. The fact that Meteor only supports UNIX based operating systems may seem a little weird. But it is only natural, since Node didn't support Windows in the beginning [98]. Maintaining only one type of operating system is easier, and makes the development process go faster.

The framework is a work in progress, and parts of it is changing all the time. This goes for the documentation as well, but I have the impression that updating it isn't the most prioritized task. Nonetheless, the constant changes makes it a little hard to follow sometimes. "Does this apply to my version?" was an often asked question. I knew that I would run into this problem when I chose Meteor, but it could have been more clear about what's set in stone and what is not. Node is also a work in progress, but their documentation does a thorough job at showing the status of different aspects. Meteor has some red text here and there, a feature that doesn't do the job as well as Node does (Figure 7‑6).



Figure ‑: Hard to keep track of stability of features in Meteor's documentation.

Meteor's documentation features many examples, but also quite a lot of text. This would be fine if it weren't for the design of the documentation page. All white and black makes things blend together, making it a little hard to read. Other than that it is well structured, showing a dynamic menu to the left that shows you where you are at a given time. This functionality breaks sometimes though. But since you can navigate by clicking on any element in the menu, this isn't too much of an inconvenience.

Overall, the documentation is impressively detailed for something that isn't complete. Maintaining it require a lot of work besides driving the project itself forward. There are also a few, simple samples and a few videos that show certain aspects of Meteor. The samples are well enough, but I found the videos hard to follow. They show a lot of code in a short amount of time, which isn’t good for learning purposes. It leaves me to believe that they are more for promotional purposes. You can see one of the videos here: <https://www.meteor.com/screencast>.

### Simplicity

Simplifying developing web applications is the motivation behind Meteor. Right now though, it is not simple. But I can see that if the smart package system [42] works well, it will make it possible to write a lot of functionality fast. The concept is that you can build applications from a collection of packages. Then you write some code to wire it all up.

With such a high level of abstraction, it feels like there is a lot of magic going on. Sometimes this is a little confusing. More than once I had to look twice at my code in the debugger to make sure that it was what I wrote. Meteor handles bundling and wrapping of your code before it serves the client. This means that your code is always wrapped into a scope for you (Example 7‑16). I must say that I prefer to handle this myself, as I think that it makes the code more readable.



Example ‑: Meteor wraps your client side files automatically in a scope.

Meteor encourages the declaration of global level[[32]](#footnote-32) functions and variables. By doing so, it isn't always clear what the effects will be. In the above example, the function, "addItem", ends up in the global scope. If this code was for a package, it would be global to the package only [42]. Any front-end developer would have second thoughts on this practice. General JavaScript development discourages the use of global variables.

Another thing that is a little tricky because of Meteor's "magic" is structuring files that depend on other files. Files load in a specific order, based on how deep they are in the tree and alphabetical. As a result, you sometimes have to make an extra folder just to ensure that one file loads before another. Meteor suggest using packages as a solution for this. I believe that this may solve the issue. Since support for making your own packages was limited when I tested Meteor, I was unable to test this.

Many of Meteor's features are revolutionary, but they also feel a little weird. For instance, publishing a record in the database, makes it available to all clients. The result is that any change to this data is broadcasted to all clients. As real time goes, this connection to a data set is unlike anything else. It somewhat resembles Lightstreamer’s item (see section 7.2.2), but these items are not necessarily connected to a database. Another typical real time feature is the ability to send a simple message from server to client or vice versa. With Meteor it is not possible to do so without involving a data set. In the test application, I implemented request/response functionality through Meteor methods (Example 7‑17). For messaging from one client to another, this does not work.



Example ‑: Meteor method used for request/response functionality.

Having direct access to the database from both client and server is something no one has done before. It lets you write database queries on the client. Security is an obvious issue here, and Meteor has introduced a package for authentication. It is also encouraged to keep sensitive code on the server. Clients emulate the "happy day" scenario of a database operation by default. As a result, you may see some changes starting to happen, before they snap back as the server responds [99]. From a users perspective this is a little odd. The functionality can be overwritten, but in my opinion it should be off by default.

All in all Meteor offers many new features to web application development. It will be interesting to follow the project in the future. I think it can get a lot of users, but I don't see large corporations throwing out the more traditional frameworks to use Meteor instead.

### Maintainability

As of now there is no official testing framework for Meteor. I find it a little worrying that the roadmap lists this as "In 1.0 if time permits" [63]. Testing is essential to keep maintainable code, and should be a part of any framework from the beginning. Especially with a framework with as many tight couplings as Meteor, where almost everything work together.

Still, there are ways to test an entire Meteor application. As of now though, they feel a little unnatural, like your hacking the framework to get it working. My approach was using Node with Mocha as test runner. Then I used a module called “unit-testling” [100] that allows you to inject mocks into another module (Example 7‑18).



Example ‑: Using "unit-testling" to inject mocks.

Meteor files are not Node modules though. With some files, this was not an issue. But testing files with global functions was another issue. To make these tests work, I had to add some test specific code into the files I wanted to test with this method (Example 7‑19).



Example ‑: Making a Meteor file testable as a Node module.

Testing Meteor is something that will best be done by integration testing of some form. Laika is a testing framework for Meteor that does just this [101]. Unfortunately, I was unable to get this working on Windows, but it looks promising. Laika proves that it is possible to write tests for Meteor in a simple manner. Whenever an official testing framework surfaces, it will probably resemble Laika in a lot of ways.

As Meteor shares a lot of code between the server and the client, a new problem occurs. Should you test the code on the server, the client or both? And how do you test code that triggers a database update on the server from the client? There are many unanswered questions with Meteor as of now. I believe that the answers will come soon and that they will be satisfactory. Many developers have fate in the project[[33]](#footnote-33), something they wouldn't if it didn't look promising.

### Browser support

Meteor supports all major browsers. I had some issues regarding the WebSocket support of SockJS though. While it does handle graceful fallback, it refused to use WebSockets in Chrome when the address was localhost. This turned out to be a bug with SockJS [102], and pointing the browser to 127.0.0.1 solved the problem.

### Maturity

The development of the project has been steady since 2011. Recently, it has started to receive a lot of appraisal from developers, thus boosting the community. In 2012 it received a substantial financial contribution [103]. This assures that the team can work on Meteor full time rather than besides other work.

Meteor introduces a new way of thinking–there is no other web application framework like it. Real time features becomes more popular every day, and Meteor puts this in the center of its applications. A repercussion of this though, is that the framework will not be suitable for every task. Then again, what framework is? The developers promises to solve a lot of problems and revamp web applications. Do they promise too much? Only time will tell, but if they deliver, Meteor will become a popular choice for small to medium sized projects.

Currently, Meteor is far from ready to be used in production code. Drastic changes to the APIs may still occur, and there are no guarantees offered by the documentation regarding any aspect of the framework. This means that the current state of the project is only suitable for case studies and hobby projects. With so many innovations, the framework will also need time to prove itself after it reaches 1.0. Having database access from the clients is the aspect that really needs to prove itself. If it turns out to be secure, and if they can support all major databases, it will no doubt change the way we think of front-end forever.

# Project part 2: Load testing

# Methodology

This part will focus on the last two problem statements (see section 1.1). That means it will be about performance. Both for the individual frameworks and the different protocols. I will strive to answer questions about what frameworks has the best performance. The question of WebSockets versus HTTP will also get a lot of focus.

## Test scenario

Real time applications are about distributing updates at once to multiple clients at the same time. The scenario I have designed follows this concept. A set number of clients will follow a strict message flow, recording data along the duration of the test. One of the clients will be dubbed "master" and handles starting the test. Figure (Figure 8‑1) describes the flow of a test. Section 8.2 provides further insight into this.



Figure ‑: Message flow in a test run.

Along with the "initTest" message, the master client sends information about:

* What test to run. The "echo" option starts a test where each message received by the server, is sent back to the caller.
* How many clients are connected. Instead of letting the server count this, I find it easier to just tell it.
* The spacing of the x-axis in generated graphs. Numbers on the axis represents time intervals in seconds. For instance, the value “0”, represents the interval from zero to one second. I ended up with running short tests, so this value was always set to one.
* The start time as recorded client side. To eliminate time differences, the server will record the start time as well. Calculations for client data will use the first, while server data will use the latter.

All messages from the "complete" message handle exchange of data. At the end of a test, clients will have access to an object with the complete dataset of the test. This allows me to separate calculations and displaying of the results from the tests themselves.

## Test data

While the tests run, both the clients and the server will collect different data. Message frequency data is the server's responsibility. When a client sends a message to the server, it gives it a timestamp. The server uses this to calculate how many messages all clients sent in a given interval. It also registers how many messages it received and sent in a given time interval. One goal of recording this data is to find out if any messages are dropped. The other is to determine if the server manages to keep up with the message stream.

Messages received by the server will be registered immediately after receiving a message. The registration of messages sent from the server, will be registered as close to the send event itself as possible.

The other telemetry measured automatically by the tests is latency data. Each client need to calculate the latency of its own messages. To achieve this, a message will be given an unique id. A client then registers that it has received a message only once, and calculates the latency using the timestamp for when it was sent. It is one drawback to this technique though. I have no guarantee that a framework sends a message to the caller first when it broadcasts. This introduces some insecurity in the values, but not much. The extra latency will never exceed the time it takes to send a message to all clients.

In addition to the data described above, I will monitor some telemetry with other tools. What a certain framework or transport require of the server in terms of memory and processor will be measured. Network usage is the data I expect to see the clearest differences between transports, along with latency. Hence, it is important to measure how many bytes gets sent during a test.

## Test setup

There are several ways to implement the test scenario:

* Using an existing tool.
* Using console applications as clients.
* Using headless browsers[[34]](#footnote-34).
* Using real browsers.

I have not found any tool that can handle testing real time applications in the manner I want. The first alternative is therefore not suitable for this thesis.

Most load test setups I have seen uses some sort of desktop or console application to emulate clients. Some examples of this are Crank [104], Apache JMeter [105] and HtmlUnit [106]. This test however, has many aspects that I have yet to see in any other scenario. I am going to test five different frameworks. All will use WebSockets, but I will also test the different fallback transports. Each framework expects messages in a different format, and they all have different ways of connecting a client to the server. Achieving this with console applications would need a lot of overhead to support it all. WebSockets would need different code than HTTP as they are separate protocols. Furthermore, I would have to manually construct each message to fit the format of a given framework.

A better solution is to make use of the JavaScript clients each framework provide. Using headless browsers, I could still use a console application. But rather than acting as clients, this application would just launch a given number of headless browsers. Sadly, few viable headless browsers exist for my purpose. Phantom is the most widespread, but the current version (1.9) does not support the newest version of WebSockets [107]. Phantom uses Google WebKit. Slimer is the Gecko (Mozilla) counterpart. This does support WebSockets, but it is an immature piece of technology. Also, it is not completely headless yet.

The final option is to use real browsers and have multiple clients in each open window. This also allows for use of the JavaScript clients each framework provide. But it is a solution that demands more resources of the client machine. In the end, it will not be possible to have as many clients with this solution as with the others.

## Choice of setup

I chose to go with real browsers. Writing many different console applications could easily have become a time trap, resulting in uncompleted tests. Using headless browsers would have been the optimal solution, but I find Slimer to be too immature. Any bugs with the testing software could have given false results.

To start up the browsers I wrote a simple Selenium application that I named “ClientLauncher”. Firefox turned out to be the ideal browser as Selenium has good support for it. It was also the browser that handled several clients best.

I used only 60 clients running in 30 browser instances. Message frequency was set to two messages per second per client. The load this produces is not substantial, but it enough to generate differences between both frameworks and transports.

I used my own Asus K55V laptop to host the clients. It has the following specs:

* Intel Core i7-3610QM CPU with 2.30 GHz.
* Eight Gigabytes of RAM.
* 64 bit Windows 7 Home Premium.

To host the servers I used a desktop computer from HP with the following specs:

* Intel Core 2 Duo-T7600 CPU with 2.33 GHz.
* Eight Gigabytes of RAM.
* 64 bit Windows Server 2012 R2 Standard Edition.

One of the major benefits of using browsers is that it allows for a lot of shared code on the client. Eight JavaScript files make up the client side code. Of these, seven are common to all frameworks. Socket.js handles communication with the server and had to be rewritten for each framework. There is a total of 50 tests for the three main files that handle the tests. This ensures that the client works as specified.

Each of the frameworks need different servers, but I wanted them to follow a strict implementation. There are two main entities for registering data: the "loadhub"[[35]](#footnote-35) and the "monitor". 19 and 16 tests respectively ensures that these entities perform to specification. In addition, most frameworks needed some extra code to wrap these entities together with sending messages. Play Framework and Lightstreamer, for instance, needed to serialize data to JSON. They share the class JSONHelper.java for this purpose (also unit tested).

## Number of runs

To get viable results, I did several runs and calculated the average values afterwards.. The results turned out to be mostly stable from run to run. Therefore, I decided that 10 runs of each combination of framework and transport would suffice.

## Displaying data

I chose to use Highcharts [108] to display charts with all collected data. This is a simple and powerful API that provide graphs that suit my need. To format the data gathered by the clients and servers, I designed a simple Web API. After a test is done, the master client is in possession of all the raw data. This is then sent to the Web API. The API then processes the data and returns three objects that the Highcharts API can make use of. A total of 33 tests ensures that also this part of the application stack performs as expected.

While the Web API provide charts for each run, it is the average values of several runs I need. To help with this, I designed a "Chart merger". This contain a series of (unit tested) functions to extract the average values of different types of tests. It then presents this in Highcharts graphs. These graphs will be used in the thesis. The chart merger also handle the data I gather manually.

## Configurations

With SignalR, IIS benefits from a few configurations. I followed the steps given by the wiki of SignalR's GitHub page [109].

Lightstreamer required some more work before it was ready for testing. First of all, I had to upgrade to the best edition available, the Vivace edition [75]. Weswit proved very helpful, giving me free access to this version for the duration of my work. They also helped with a few settings needed for the server. Not all of these setting made a difference for me, so I didn’t use all they suggested. The only tweak I made was setting the attribute “max\_delay\_millis” to 0.

My initial results showed that Lightstreamer lagged behind all the other frameworks, so I turned to Weswit for help. Appendix A shows the whole mail correspondence with an engineer at their office. One of the critical things I learned from them is that the JVM needs to warm up before load tests. As a result, tests done with Play and Lightstreamer ran for 30 seconds before I started recording data. A long running test showed that the run stabilized around this mark. Hence, there was no need for a longer run.

## Monitoring network traffic

Several tools were tested to find one that fit my needs. I was interested in a tool that could show me the total amount of bytes sent given a certain filter. It also needed to have functionality to inspect packages for both WebSockets and HTTP traffic. Network Monitor [110], Wireshark [111] and Fiddler [112] were the most prominent that I considered. In the end, the choice fell upon Wireshark as it provided exactly what I needed.

## Monitoring of processor

While I could have chosen an advanced tool with pin point accuracy, this was not what I wanted with this data. The most interesting for me was to see if there were clear differences between different transports and frameworks. A simple tool that could have been used is Perfmon [113]. This tool offers counters for processor time, not for the total usage. The server is the main program running on the test machine. The counter for processor time shows how many percent of the total used percent a process uses. Because of this, the server would always show close to 100% usage[[36]](#footnote-36). Windows Server 2012’s Resource Monitor was therefore chosen for this purpose.

## Monitoring of memory usage

Memory usage falls into the same category as processor usage. I wanted to see if there were any clear differences. The task manager in Windows was more than enough of a tool for this job. Readings were taken at the end of each test, as this proved to be the maximum usage during the test runs.

## Different servers and platforms

The frameworks run on quite different servers and platforms. Node is for instance a lot more lightweight than Java or C#. This is because the V8 engine it builds on, is implemented in C++. C++ compiles to machine code, and it does not need a runtime to function. Both Java and C# code compiles to byte code that has to be interpreted by a runtime. For Java this is the Java Virtual Machine (JVM) and C# has the Common Language Runtime (CLR). As a result, C++ is a faster language than both C# and Java [114].

I chose a coarse grained method for monitoring resource usage of the different frameworks. As a result, the results for processor usage and memory usage shows more than just the frameworks' usage. A substantial part will belong to the server in some cases.

This method allows me to see what the different servers require of the machines they run on. In my opinion, this data is more relevant than what the frameworks themselves use. The server is a part of the total package, and I believe that it is interesting to see how these perform compared with each other. I will perform an analysis of this results in chapter 10.

No matter what method I use to measure machine resources, I still get an overview of how each transport perform. This comparison cannot be done across multiple frameworks anyways. For instance, it would be wrong to say that long-polling is better than WebSockets based on Socket.IO with long-polling and SockJS with WebSockets.

## Use case of test setup

I should point out that my setup is tailored to suit the needs of this thesis. It is the web application aspect of each framework I am testing. The use of JavaScript clients is thus the most natural. Results should be read with this in mind. Some of the frameworks support clients on other platforms than browsers and have other use cases than web applications. These frameworks may prove to perform differently under such circumstances.

## Limitations

This section describes various limitations I ran into using the chosen test setup.

### Meteor

Meteor has the real time component tightly embedded in its core. This made it impossible to use more than one client pr. browser without fiddling around with Meteor’s source code. As this could easily could have broken certain aspects of the framework, I decided to test Meteor’s real time component alone. SockJS therefore replaces Meteor in this part of the thesis.

### Using browsers

There are certain drawbacks to using browsers as clients. Most obvious is the fact that you cannot have more than a few open connections at the same time. The HTTP 1.1 specification states that no client should have more than two connections to a single host [8].

Nonetheless, most modern browsers have increased this limit to somewhere between four and eight [115]. Consequences of this are that I cannot have more than two or three clients per browser, and I cannot have a high message frequency. If I do, I risk reaching the maximum number of connections, thus introducing extra latency.

Another repercussion is that the number of clients will be limited by the client machine. A browser takes up about 100 megabytes of RAM when idle. Considering that each browser will store some data as well, this number will grow close to 200 megabytes. My machine was pushed to the limit with 30 browsers running.

### Network capture

I started a new capture when I started the tests. This turned out to have huge drawbacks that I did not discover until all tests were run. Open WebSocket connections was not recorded correctly. Messages going from server to client are there, but they show up as TCP traffic. The other way is masked, so there is no way of knowing if the data is correct or not. HTTP-streaming suffered from similar problems. It also seems to be some issues with long-polling with SignalR. Some messages that I expect to be in the capture are not there. This leads me to believe that the captured data is not accurate enough to be presented as results. It can give an indication, but that is all.

To present some more believable results I have done calculation of the theoretical throughput of each test kind. The basis for each calculation is a capture with one client sending 10 broadcast messages in two seconds. For these captures, I started the capture before I navigated to the test page, thus ensuring that all packets got captured.

There are obviously sources of error with this approach as well. Some of the frameworks may compress data more or send multiple messages in one package during higher loads. I must thus stress that the results from my calculations do not take this into consideration. The actual performance of each framework may be better than what the calculations indicate.

A benefit of doing this is that it allowed me to show results for transports that are not supported by the frameworks in Firefox. This applies to HTTP-streaming with SignalR and Server-Sent Events with SockJS. Keep in mind though, that there may be some differences between how browsers handle different transport mechanisms. Still, it gives an indication.

The basis for the calculations is on GitHub [116].

### Streaming with Play Framework

Streaming with Play Framework required two forever frames in each browser. I have not been able to understand why, but this did not work. Even when I used one client per browser, connections failed after six. For this reason, I excluded HTTP-streaming with Play from the tests. I was able to get a basis for calculations of network traffic for it though.

## Testing idle connections resource usage

To find out whether WebSockets has potential to set a new standard for client/server communication, I will measure how much resources idle WebSockets connections use. If it is to have any chance of replacing HTTP as transport method in static web pages, it should use close to no resources. There is no such thing as an idle HTTP connection in such cases, as the client requests content and gets it via a response. The connection is then closed. Using WebSockets for such a case would require the connection to remain open.

I will perform two tests with each framework to test this. One will connect 1000 clients evenly spread across 10 browsers to the server. The other will connect 4500 clients. This will be done using 25 browsers with 180 clients in each. 4500 clients may seem like an odd number, but there are some reasons for it. I ran some tests, to see how many clients I could connect in a single browser. This was stable up to around 200. After this, some connections were refused. With a safety margin on 20, I ended up with 180 clients pr. browser. My computer was able to handle about 30 instances of Firefox. Since these test would use three times as many clients per browser, I wanted to add a safety margin here as well. To be safe, I landed on 25 browsers.

## Raw data

All raw data can be viewed on GitHub: <https://github.com/kjohann/MasterThesis/tree/master/Loadtests/Results>

# Results

This chapter describes the results of the tests. Each graph show the average values of the 10 test runs for each framework/transport combination. For the results regarding messages, I only show a selection of graphs. The graphs I show, highlight some general trends in the results as well as anomalies I experienced. All graphs can be reviewed in Appendix B. An analysis of the results is given in the next chapter.

## Messages sent from clients

With 60 clients and two messages pr. second, I expected a stable graph with 120 from 0 to 14 for a perfect run. A lot of the runs show this behavior, but a lot more than expected have certain anomalies. Figure 9‑1 shows the results for WebSockets. It displays one example of such anomalies. Some runs went one second past the expected 15 second duration of the test, resulting in a sudden drop during this extra second.



Figure ‑: Messages sent from clients using WebSockets.

Some tests had a lot more overtime than just one second. The graph below (Figure 9‑2) shows results from runs using polling. As you can see, the results from Socket.IO stands out from the rest with as much as ten seconds overtime. This behavior was consistent through all the test runs. The Lightstreamer results was also consistent.



Figure ‑: Messages sent from clients using polling.

The results from HTTP-streaming show a lot of overtime for Lightstreamer (Figure 9‑3). This reflects only one run that lasted seven seconds too long. None of the other runs replicated this effect, but they still ran one or two seconds of overtime.



Figure ‑: Messages sent from clients using HTTP-streaming.

## Messages received by server

I expected the server to receive the same amount of messages that were sent from the clients in a given interval. The results show that this is the case for most of the tests. The graph below (Figure 9‑4), shows both messages sent and received for all frameworks using long-polling. As you can see, the results are so close that there is almost one line in the graph. This is the expected behavior of a perfect run.



Figure ‑: Messages sent from clients and received by server using long-polling.

Again though, there are some anomalies. Figure (Figure 9‑5) shows the WebSockets results for Lightstreamer. There are some deviations between messages sent and messages received. The difference is not significant, but it is enough to show that it is not a perfect run. I will discuss possible reasons for such variations in the Analysis chapter.



Figure ‑: Messages sent from clients and received by server with Lightstreamer WebSockets.

While most of the variations are small, polling is, once more, the largest source of anomalies. The clearest example of this is Lightstreamer (Figure 9‑5). It seems like the server worked in intervals, with peaks of more than 130 and bottoms of 105 received messages. These are variations of up to 12,5%[[37]](#footnote-37), which is quite significant.



## Messages sent from server

A perfect test should show that the number of messages received multiplied by 60 is a perfect match to the number of messages sent. It broadcasts every message to the 60 clients, thus the multiplication by 60. The following graph (Figure 9‑6) show this behavior for SignalR using Server-Sent Events. It is a small deviation the first second, but otherwise, the lines are almost 100% matches. This behavior was consistent throughout all the tests.



Figure ‑: Messages sent by server (divided by 60), corresponds with messages received.

## Average latency

This section will present the results for each transport. An analysis of the effects of a transport on latency will be given in the next chapter.

I did not warm up the servers for SignalR, Socket.IO or SockJS before any of the test runs. Looking at the results in this section, it seems that they may have benefitted from a warm up after all. The most accurate results are therefore the ones recorded from five seconds and onwards.

### WebSockets



Figure ‑: Average latency using WebSockets.

Both Play and SockJS relies on some manual client handling. Broadcasting messages with these frameworks relies on a loop construct. The other three have internal mechanisms that handles this. Figure 9‑7 shows a clear divide between the first two and the other three. Of the three best, Socket.IO had the best performance as it stabilized around five milliseconds. Lightstreamer and SignalR follow each other closely, but did not stabilize in the same manner. On average, they had a latency of a little less than 20 milliseconds, with some peaks going above 25.

### Server-Sent Events

Only SignalR provide Server-Sent Events as one of the transports. The graph below (Figure 9‑8) compares the results of this transport to SignalR's results using WebSockets. This is a very interesting result as it shows that Server-Sent Events matches WebSockets when it comes to latency.

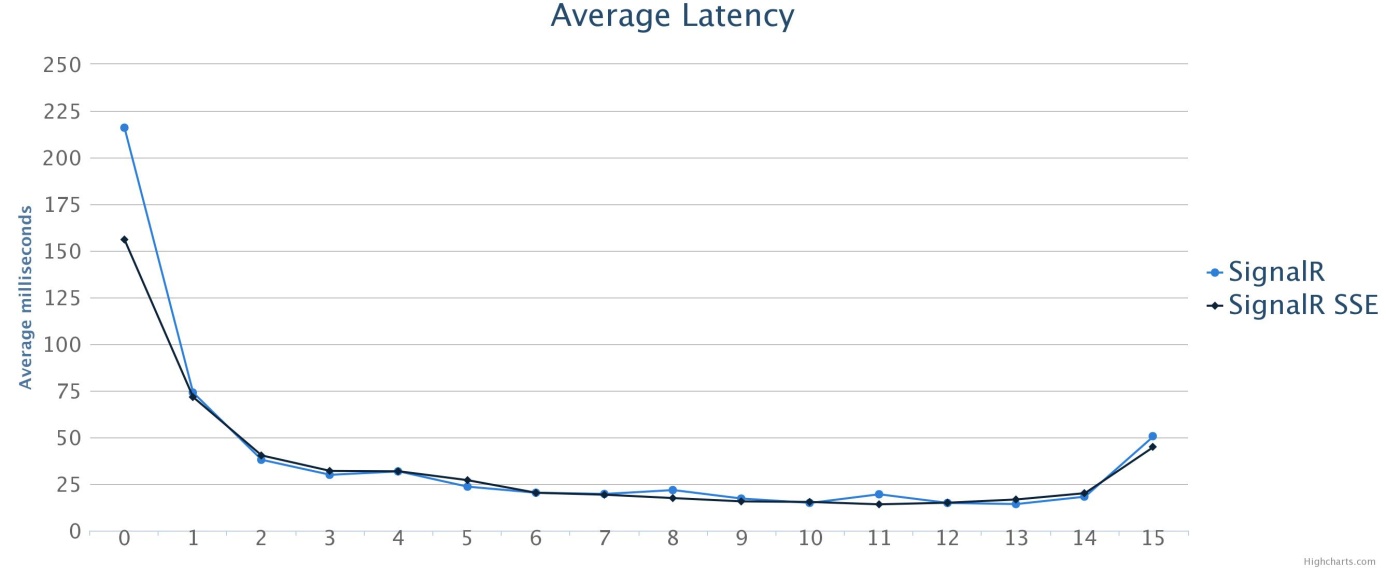


Figure ‑: Average latency of SignalR with Server-Sent Events and WebSockets.

### Http-streaming

In section 9.1, I mentioned that only one run with Lightstreamer had seven seconds of overtime. The graph shown in Figure 9‑9 reflects this fact, as the latency increases a lot towards the end. As the other runs never passed two seconds of overtime, the data from 18 to 21 seconds should be disregarded. Prior to the 18 second mark, Lightstreamer performs well with HTTP-streaming. The latency is higher than WebSockets, but not much. SockJS does not come close to the same. Here though, the difference to WebSockets is more clear than with Lightstreamer.



Figure ‑: Average latency with HTTP-streaming

### Long-Polling

As with WebSockets, Socket.IO outperforms the other frameworks with long-polling. However, as the graph shows (Figure 9‑10), SignalR's latency decreases throughout the duration of the test. As discussed in the introduction to this section, I did not warm up the server for SignalR. The other transports with SignalR showed a stabilization after five seconds. With long-polling, this is not the case, even though it seems to be stabilizing around the ten seconds mark.



Figure ‑: Average latency with long-polling.

### Polling

Once again, polling is the greatest source of deviating results. As the graph (Figure 9‑11) shows, Lightstreamer has a dramatic increase in latency towards the end, even with WS-polling. Socket.IO also has a peak with twice as much latency as the rest of the test. In the stable part of the graph, we see that Socket.IO does not handle polling well.

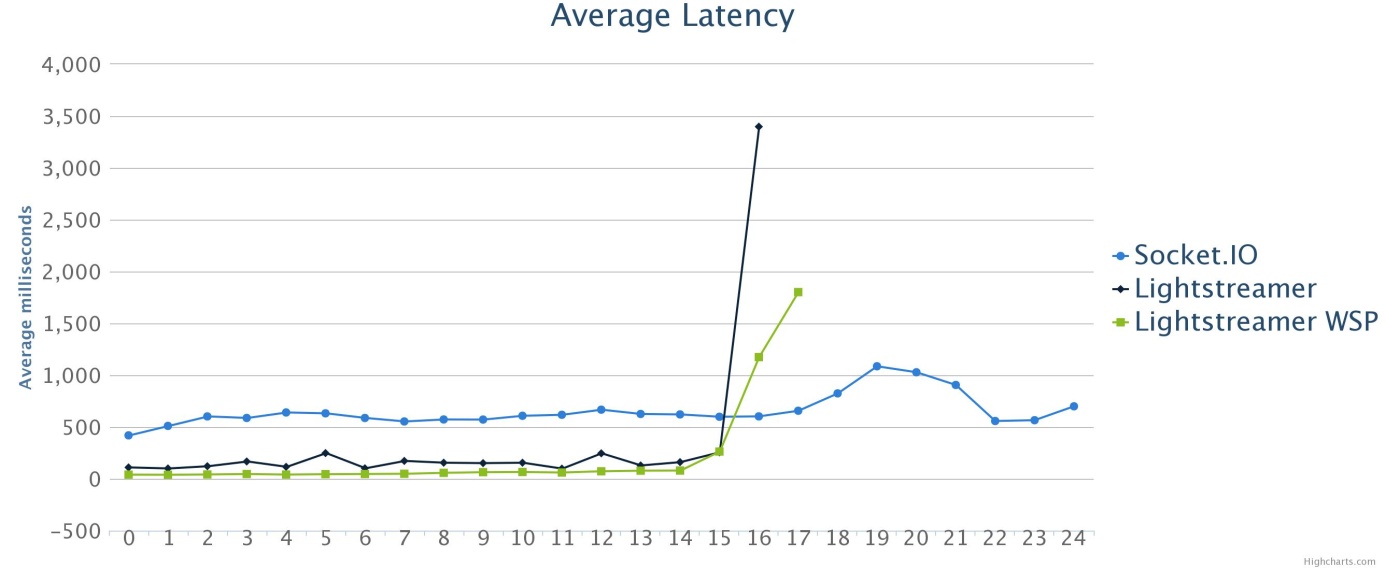


Figure ‑: Average latency with polling.

Polling using WebSockets is more efficient than over HTTP by about 100 milliseconds. This is not surprising, but I still don't see the use case for this technique. As a browser has to support WebSockets to use it, it should be compared to streaming using WebSockets. Figure (Figure 9‑12) compares the two, and it is obvious that WebSockets perform best when streaming data. Polling over WebSockets degrades performance more than three times on average.



Figure ‑: Average latency with Lightstreamer using WebSockets and polling over WebSockets.

## Median processor usage

As discussed in the methodology (see chapter 8), these result reflect the usage of the servers, not just the frameworks. The lightweight nature of Node shines through as both Socket.IO and SockJS takes up about half as much as any other (Figure 9‑13). Otherwise, it seems that the Java solutions perform a little better than SignalR.



Figure ‑: Median processor usage - all frameworks and transports.

## Maximum memory usage

Again, the results reflect the resource usage of the servers as well as the frameworks. As with processor usage, one can see that Node outperforms both Java and C# (Figure 9‑14). Otherwise, we see that SignalR, Socket.IO and SockJS have almost the same memory usage regardless of transport. Polling with Lightstreamer though stands out from everything else. It uses about nine times as much as WebSockets.

There is one source of error in the results though. I measured the memory consumption at the end of the test. While I did not see any drops during the tests, it was some drops that occurred just after the test finished. Some of the raw data stands out from the rest because of this.



Figure ‑: Maximum memory consumption - all frameworks and transports.

## Bytes sent/received

In section 8.13.3, I mentioned an error regarding my usage of Wireshark to monitor network traffic. I still want to show the recorded data, as there are some implications in it when compared to the calculated network traffic. The following graph (Figure 9‑15) is a result of the recorded network traffic.



Figure ‑: Bytes sent/received (captured) - all frameworks and transports.

The next graph is the data I calculated. (Figure 9‑16). It shows that SignalR sends a lot more data than any of the others. It also gives a clear indication that the streaming techniques use a lot less network traffic than polling and long-polling. There is one result that stands out from the general trend. Server-Sent Events with SockJS uses more than twice that of WebSockets and HTTP-streaming.



Figure ‑: sent/received (calculated) - all frameworks and transports.

I expected Play framework to be the best framework in this category. It is the framework with the closest to a "raw" WebSockets implementation. But the results show that both the Node frameworks perform a little better. Lightstreamer is also not far behind.

## Idle clients with WebSockets

Table 9‑1 shows how much memory each framework used with 1000 and 4500 clients. Processor usage was mostly 0 for all. Deviations are due to small peaks that each framework had at set intervals. The true processor usage was therefore nothing for all frameworks regardless of how many clients were connected.

|  |  |  |  |
| --- | --- | --- | --- |
| Framework | 1000 clients | 4500 clients | Increase factor |
| Play | 118932KB | 223812KB | 1,88 |
| SignalR | 188892KB | 537064KB | 2,84 |
| Socket.IO | 42724KB | 81780KB | 1,91 |
| SockJS | 38800KB | 85996KB | 2,2 |
| Lightstreamer | 166,788KB | No result | No result |

Table ‑: Memory usage with 1000 and 4500 idle WebSockets connections.

I was not able to get any results with 4500 clients using Lightstreamer. At around 2500 clients, the browsers used so much memory that my computer crashed.

# Analysis

In this chapter, I will discuss the results of the tests. I will try to provide plausible explanations to some of the anomalies in the results. The meaning of the results will also be discussed. Explanations are the result of my own experiences and opinions.

## Message frequency

This section discuss the results from sections (9.1, 9.2 and 9.3). Most aspects applies to all, so I found it natural to keep the discussion in one section. All tables show the highest and lowest recorded data from their respective results. These numbers are from the raw data. The tables also show an average value. This is calculated based on the values in the corresponding graph.

Looking at any graph about message frequency, one can see some minor drops. These are most likely the result of either concurrency issues or timing. Thread proofing any application is not an easy task, even for experienced developers. It is likely that my code has one or more weak spots in this matter. This may have caused some messages to be skipped, resulting in a minor drop in the graph.

Time issues can have surfaced even though I used a separate start time for clients and the server (as described in section 8.1). The clients start time was set before the call to the "initTest" procedure. With SignalR, I observed that this took some time to respond, which explains the one second overtime each test with this framework had. The other frameworks also had some occurrences of this, resulting in an average result with one second overtime.

The table below (Table 10‑1) shows data for messages sent by the clients. Most cases show an average number that is smaller than the expected 120. All these are the result of tests that went into overtime. A lower number, represents more overtime. The case with the runs that only used one second extra, is that they often registered less than 10 messages during this second. Low numbers for SignalR and Socket.IO (except for polling) is due to this effect.

|  |  |  |  |  |  |  |  |
| --- | --- | --- | --- | --- | --- | --- | --- |
|  | | Polling | Long  polling | Streaming | Server-Sent Events | WebSockets | WS-polling |
| SignalR | High:  Low:  Average: | - | 120  2  (113) | - | 120  18  (112) | 120  2  (112) | - |
| Socket.IO | High:  Low:  Average: | 120  2  (72) | 120  4  (113) | - | - | 120  120  (120) | - |
| SockJS | High:  Low:  Average: | - | 120  120  (120) | 120  4  (107) | - | 120  4  (114) | - |
| Play | High:  Low:  Average: | - | - | - | - | 120  120  (120) | - |
| Lightstreamer | High:  Low:  Average: | 120  2  (111) | - | 120  2  (88) | - | 120  44  (114) | 120  1  (109) |

Table ‑: Messages sent by clients.

While Socket.IO shows a highest number of 120 for polling, its average is a lot lower. As you see from Figure 9‑2, it was far beneath the expected frequency throughout the test. A probable explanation to this behavior is a combination of two things. First, that the server took a long time to respond to a poll request. The captured network traffic indicate that several messages often was bundled into a single response. This operation requires time on Node's singe thread. This may have blocked other incoming requests for a short period of time. As a result, the clients can have reached six outgoing requests which is Firefox's maximum [115]. Together, these phenomena result in a queue of request forming at both the clients and the server. Then, 10 seconds of overtime isn't too improbable.

The clients single threaded nature, can cause some issues. If the server has a peak in messages sent, which some results show, the client suddenly receives a lot more work. This can push subsequent “send”-calls far down the call stack, resulting in a frequency drop.

Explaining what happens towards the end of each run with Lightstreamer is rather hard. All runs show unstable frequencies. I ran tests with all frameworks to see the resource usage with many idle clients using WebSockets. With Lightstreamer, I was not able to connect more than about 2500 clients before the browsers used all my computers memory. None of the other frameworks displayed this behavior. This may indicate that there is a memory leak in the Lightstreamer client library. If this is the case, it can impact the clients abilities. But, as I have no clear indications towards this, it remains pure speculation.

Another possible explanation is the usage of an ExecutorService [81] with a cached thread pool to handle concurrency. This is the only difference between the concurrency aspects of Play and Lightstreamer. As you can see of the data, Play had no drops in frequency what so ever. I have found indications that a cached thread pool used in conjunction with synchronized code, degrades performance [117]. This is exactly how I registered data in the Lightstreamer application (Example 10‑1). If this is the case, more than just the frequency data may have been affected.



Example ‑: Code executed by an ExecutorService that triggers synchronized code.

The table (Table 10‑2) shows data for messages received by the server. There are some isolated cases of more than 120 messages received for several frameworks and transports. SignalR had this behavior for both Server-Sent Events and WebSockets. Both cases had this as the number for the interval from one to two seconds. Both followed a first second with close to 110 received messages. In section 9.4, I mentioned that SignalR could have benefitted from a warm up of the server. This behavior seems to support this.

|  |  |  |  |  |  |  |  |
| --- | --- | --- | --- | --- | --- | --- | --- |
|  | | Polling | Long  polling | Streaming | Server-Sent Events | WebSockets | WS-polling |
| SignalR | High:  Low:  Average: | - | 120  4  (113) | - | 127  2  (112) | 127  2  (112) | - |
| Socket.IO | High:  Low:  Average: | 110  1  (72) | 120  2  (113) | - | - | 120  120  (120) | - |
| SockJS | High:  Low:  Average: | - | 120  120  (120) | 133  4  (107) | - | 120  27  (113) | - |
| Play | High:  Low:  Average: | - | - | - | - | 120  120  (120) | - |
| Lightstreamer | High:  Low:  Average: | 269  0  (111) | - | 130  4  (88) | - | 120  2  (117) | 122  2  (100) |

Table ‑: Messages received by server.

With HTTP-streaming though, we see some high peaks occurring more than once. Looking at the network captures for these cases, it appears that some responses to incoming POSTs was sent rather late. It is not possible to tell for sure, since the capture say nothing about what browser a request and response belonged to. Nor does it say what specific request a response belongs to. If some responses were slow, it can have caused the clients to reach the connection limit.

The final table (Table 10‑3) regarding message frequency shows messages sent from the server. This data correspond to messages received, and indicate that each server managed to send messages out at the same pace as it received them. The most obvious anomaly is also seen in the previous table.

|  |  |  |  |  |  |  |  |
| --- | --- | --- | --- | --- | --- | --- | --- |
|  | | Polling | Long  polling | Streaming | Server-Sent Events | WebSockets | WS-polling |
| SignalR | High:  Low:  Average: | - | 7320  60  (6746) | - | 7620  120  (6741) | 7620  360  (6777) | - |
| Socket.IO | High:  Low:  Average: | 6540  120  (4350) | 7200  120  (6785) | - | - | 7200  7200  (7200) | - |
| SockJS | High:  Low:  Average: | - | 7200  7200  (7200) | 7560  240  (6411) | - | 7200  1620  (6823) | - |
| Play | High:  Low:  Average: | - | - | - | - | 7200  7200  (7200) | - |
| Lightstreamer | High:  Low:  Average: | 15060  0  (111) | - | 7920  120  (5262) | - | 7200  120  (7013) | 7320  120  (6001,41) |

Table ‑: Messages sent from server.

Lightstreamer had some occurrences of no messages sent or received in a given interval. In the same interval, there is no corresponding drop in the messages received. Something happened on the server that caused it to skip a whole second. Immediately after, it sends the “skipped” messages, resulting in the high peaks you see in the tables (as much as 15060 sent messages with polling). Network captures from these runs show that it is not because of registration error. In the intervals in question, there actually are no messages going out from the server. At the same time, only a few POST requests are sent as well. Since there is no drop in send frequency, it seems that the send timestamp has been set correctly. The only explanation then is that something happened with the “send” routine of Lightstreamer’s client library. This may have resulted in a delay for most of the POST requests.

## Average latency

This section discuss the latency results. The results presented each transport individually. Here I will discuss the performance of each framework, but the main focus will be on comparing the transports.

The table and graph presented in this section used the average values from the fifth second to the 14th–a ten second span. There are two main reasons for this: Some frameworks had anomalies in the first couple of seconds, indicating that they could have used warm up. Others had anomalies towards the end with a lot of over time. The most representative results are in the interval that the table is based on

### Frameworks

The relationship between the frameworks is consistent for almost all the results (Table 10‑4). Socket.IO and polling is the only combination that deviates. Otherwise, each framework's "rank" using WebSockets remains the same across all transports:

* 1st: Socket.IO.
* 2nd: Lightstreamer.
* 3rd: SignalR.
* 4th: SockJS.
* 5th: Play.

|  |  |  |  |  |  |  |  |
| --- | --- | --- | --- | --- | --- | --- | --- |
|  | | Polling | Long  polling | Streaming | Server-Sent Events | WebSockets | WS-polling |
| SignalR | High:  Low:  Average: | - | 155,4  113,7  (134,8) | - | 27,1  14,1  (18,1) | 23,6  14,2  (18,4) | - |
| Socket.IO | High:  Low:  Average: | 666,8  553,5  (605,6) | 58,1  48,8  (52,6) | - | - | 7,4  5,3  (6,0) | - |
| SockJS | High:  Low:  Average: | - | 223,3  215,3  (198,6) | 172,9  86,5  (120,0) | - | 63,1  59,2  (61,2) | - |
| Play | High:  Low:  Average: | - | - | - | - | 85,4  69,9  (75,9) | - |
| Lightstreamer | High:  Low:  Average: | 247,9  98,1  (161,7) | - | 49,5  22,8  (33,1) | - | 29,0  12,9  (17,8) | 79,5  44,4  (61,8) |

Table ‑: Average latency.

Even though Socket.IO ran 10 seconds too long using polling, it is a little strange that the latency is so much more than long-polling. Lightstreamer's latency increases by a factor of 4,89 from streaming to polling[[38]](#footnote-38). Moving from long-polling to polling shows a factor of 11,51[[39]](#footnote-39) for Socket.IO. Inspecting the network traffic reveals a probable cause. Both frameworks bundle several messages into single responses. But the captures for Lightstreamer has almost the double amount of packages. This indicates that Socket.IO bundles more messages into single responses than what Lightstreamer does. Doing this probably cost time, resulting in the higher latency.

Socket.IO in general performs a lot better than even Lightstreamer and SignalR in the tests. With a higher load this would not be the case. A single threaded server will not be able to handle the same amount of load as one with many threads. On the other hand, one can run several Node instances for the same resource cost as a single IIS or Lightstreamer server. This case is outside the scope of this thesis, though.

Both Play and SockJS are more manual approaches. SockJS handle many real time features, but broadcasting is not something it supports out of the box. These frameworks relied on a loop to send messages to all clients, a solution that is obviously far from optimal. As a result, they are far behind the others. An observation that is very clear if you compare SockJS to Socket.IO. Both run on Node, but Socket.IO is a lot faster. Even long-polling beats SockJS's WebSockets.

### Transports effect on latency

The graph in figure (Figure 10‑1) shows the average values from table (Table 10‑4). Across all frameworks, WebSockets is the transport that performs best. The only exception is Server-Sent Events that perform just as well[[40]](#footnote-40). HTTP-streaming isn't far behind, but here there is a clear increase in latency. Long-polling and polling are, without doubt, a lot slower than any streaming technique–WebSockets or not.



Figure ‑: Average latency as bar chart.

The low latency of HTTP-streaming surprised me a little. It has an average latency of about twice as much as WebSockets. A large part of this difference is likely because it takes time to send a message from a client to the server. This uses a normal HTTP-request, which relies on setting up and tearing down a whole new connection. It is clear to me that HTTP-streaming is a reliable technique to build pure push applications upon.

Considering that Server-Sent Events also rely on HTTP-requests to get messages from a client, its performance is impressive. Higher loads may benefit WebSockets over both techniques. A blog post by William P. Riley-Land suggests that WebSockets perform better with higher loads [118]. This test used two different libraries for the different transports: connect-sse [119] for Server-Sent Events and Socket.IO for WebSockets.

Riley-Land claims that the results are "in the same order of magnitude" [118]. His results show that WebSockets are 31%[[41]](#footnote-41) faster. Therefore, I disagree with his conclusion, as it seems to me that WebSockets outperform Server-Sent Events by quite a lot. Another blog post [120] showed complete opposite results. This used a proxy with beta stage support for WebSockets, so I don't count these results as reliable.

Server-Sent Events and WebSockets are both HTML5 APIs. A question that came to my mind is: "Do we need both?". Server-Sent Events is simpler to implement than WebSockets: It has a more powerful API and server side it uses normal HTTP. With that in mind, I also support this opinion. But with a framework such as SignalR, this argument is invalid, since it handles the transport of messages. Still, for pure push applications, I can see the benefit of using Server-Sent Events.

The use of WebSockets to do polling is not meant for a push dominated application. I don’t see why it is part of the Lightstreamer stack, since the connections are kept open just as with streaming. The only difference is that the client has to send a poll message over the WebSocket connection to get data. The result is that the performance drops and streaming over HTTP becomes a better alternative. A browser that supports this mechanism, also supports streaming over WebSockets. For real time purposes, this is preferable anyways, which renders WebSockets polling little useful. To me though, it was nice to have, as it helps highlight differences between HTTP and WebSockets. Polling over WebSockets is almost three times faster than over HTTP.

## Machine resources

This section will focus on the resource usage of the different transport mechanisms in the context of their framework. Section 8.11 describes the differences between the various platforms. Implications of this is that it is hard to compare for instance SignalR to Socket.IO or even Lightstreamer for that matter. Still, I will provide a short discussion regarding this.

Table 10‑5 gives an overview of resource usage data collected from the graphs in Figure 9‑13 and Figure 9‑14. There are some interesting aspects in these results. SockJS uses more processor than Socket.IO with WebSockets, but less with long-polling. Except for this the general trend is the same as for latency: the "rank" with WebSockets remains for the other transports. SockJS use a library for its WebSocket support: Faye [121], whereas the rest is built from scratch. But since long-polling uses less than HTTP-streaming as well, I do not think this is the reason. It may be part of it though. It may be something about how Node.js handle this technique.

|  |  |  |  |  |  |  |  |
| --- | --- | --- | --- | --- | --- | --- | --- |
|  | | Polling | Long  polling | Streaming | Server-Sent Events | WebSockets | WS-polling |
| SignalR | Processor:  Memory: | - | 58,7%  171 KB | - | 37,6%  165 KB | 37,9%  144 KB | - |
| Socket.IO | Processor:  Memory: | 8,2%  32 KB | 14,8%  32 KB | - | - | 14,3%  35 KB | - |
| SockJS | Processor:  Memory: | - | 10,7 %  32 KB | 21,9%  31 KB | - | 17,1%  32 KB | - |
| Play | Processor:  Memory: | - | - | - | - | 32,1%  137 KB | - |
| Lightstreamer | Processor:  Memory: | 70,2%  888 KB | - | 26,2%  107 KB | - | 21,6%  110 KB | 47,8%  209 KB |

Table ‑: Resource usage of all frameworks and transports.

Another thing that stands out is that Play use more resources than Lightstreamer. As Play is the cleanest implementation, I did not expect this. It shows that you benefit from having a server that only handles real time. Play's server is an application server that also have to serve the MVC application.

It is a trend in the data that the use of machine resources increase as the transport moves farther from WebSockets[[42]](#footnote-42). Except from polling with Lightstreamer it nothing dramatic. Some transports even use a little less of either processor or memory than WebSockets. This difference, though, is so small that it can be counted as equal. This means that Server-Sent Events use just as much processor as WebSockets. HTTP-streaming with Lightstreamer also use just as much as WebSockets.

WS-polling use twice as the amount of resources as streaming with WebSockets, which is a little strange. That it uses more processing power isn't too unlikely, but the extra memory is. It uses the same amount of open connections as streaming with WebSockets. Handling polling may require some sort of mechanism that streaming doesn't need. This can explain the extra memory usage. But it does not explain why polling over HTTP takes up more than four times as much as over WebSockets.

Despite the presence of a garbage collector, memory leaks can occur in Java programs [122]. With a immature framework I could have believed this to be the cause. But with a 14 year old framework, it is hard to believe. No other combination of transport and framework shows the same increase in resource usage. In my opinion, this makes a memory leak plausible. It would explain it at least.

Unfortunately, the polling results of Socket.IO cannot be deemed accurate. The server used less resources, but as the test took 10 seconds longer, the results cannot be compared to Lightstreamer.

## Network performance of frameworks

The results, shows only small differences between most of the frameworks. SignalR, on the other hand, stands out both in the calculated and the captured results. Looking into the capture files, one can see that it is just a manner of how it handles messages. There are some excess data like cursors that the other frameworks don’t have in the same way. SignalR uses a cursor to help with reconnection. This may give SignalR an edge over the other frameworks in this matter. Investigating this is outside the scope of this thesis, which is why I haven’t looked into it.

## Transports effect on network traffic

In section 9.7 I showed figures displaying the captured network traffic and the calculated. The table below (Table 10‑6) compares the data from the different results.

|  |  |  |  |  |  |  |  |
| --- | --- | --- | --- | --- | --- | --- | --- |
|  | | Polling | Long  polling | Streaming | Server-Sent Events | WebSockets | WS-polling |
| SignalR | Calculated:  Captured: | - | 220M  49M | 95M  - | 94M  45M | 65M  45M | - |
| Socket.IO | Calculated:  Captured: | 105M  31M | 104M  30M | - | - | 27M  30M | - |
| SockJS | Calculated:  Captured: | - | 125M  27M | 29M  27M | 70M  - | 27M  31M | - |
| Play | Calculated:  Captured: | - | - | 36M  - | - | 28M  26M | - |
| Lightstreamer | Calculated:  Captured: | 188M  65M | - | 42M  32M | - | 29M  29M | 68M 36M |

Table ‑: Both captured and calculated network traffic.

As long-polling and polling don't have an open connection, the captures of these two transports should be correct. The case with SignalR shows that these cannot be trusted either. Still, the difference between the calculations and the captured results are substantial for most transports. WebSockets actually turned out to be closest to the theoretical throughput.

Using a short run with only one client as basis for calculating, introduces possible differences with a full run. I have already described behavior where several messages have been put into single responses (see section 10.1). This is a common way of saving network usage, and it is likely that all frameworks do this across all transports.

So the calculated data does not take message compression into consideration. Nor does it consider any altered behavior in cursor messages such as SignalR sends. But the data is an accurate representation of the theoretical throughput. The only presumption is that communication remains the same. Keeping this in mind, comparing frameworks and transports on this basis is a valid method.

The background chapter highlighted overhead regarding header data as a drawback to HTTP. Looking at the result may cause you to believe that this is false for HTTP-Streaming and Server-Sent Events. Let us use SignalR as an example. The capture used to calculate throughput contained 14960 bytes for WebSockets and 45879 bytes for long-polling. In other words 3,1 times as many bytes. The calculated data show 64,98 and 219,69 million bytes for WebSockets and long-polling respectively. Long-polling has a theoretical throughput of 3,4 times as many bytes. This supports what I wrote initially, but if we look at Server-Sent Events, we see that the results are different.

29066 bytes was captured with Server-Sent Events. More than twice the amount of bytes as for WebSockets. But the calculations show 93,57 million bytes, only 1,4 times as much.

If we look at the behavior of the two cases, the capture and the full test, the reason becomes clear. The small test used for the capture sent the same amount of messages as it received. In the full test, as you can see from Figure 10‑2, this is not the case at all. Since messages going from the server use an already open connection, there is no header data. Only the messages from the clients has this. With non-streaming techniques, receiving from the server involves a GET or POST request first. Then you get the extra overhead also for the messages going from the server.



Figure ‑: Piechart that shows relationship between messages to and from the server.

This does not implicate that pure push HTTP-technologies match WebSockets's byte usage, though. Every framework "spends" more bytes sending a message over a streaming connection than over WebSockets. Most of the frameworks show that HTTP-streaming or Server-Sent Events use around 30-50% more on average. SockJS is an exception to this. Here HTTP-streaming use the same as WebSockets. As the WebSockets part of SockJS use another library, it is not possible to tell if this is representative.

## Idle clients with WebSockets

Memory usage does not increase with the same factor all the time as new clients connect. But for a worst case scenario, we can pretend it does. Recall the results from Table 9‑1 in section 9.8. SignalR was the framework that showed the largest increase in memory usage. It also used the most memory to begin with. So I will use Play as an example[[43]](#footnote-43). I don't see a future where the majority of servers are Node-based. A Java Framework will therefore be more representative.

The graph in Figure 10‑3 shows how the memory usage will develop if it increases with a factor of 1,88 while clients increase with a factor of 4,5. When it closes on two million clients, it approaches a memory usage of 2,9 gigabytes[[44]](#footnote-44). Two million concurrent clients is a lot for most web pages. But, if a static web page has this many clients at once, it will not affect the server at all if it uses traditional HTTP. If it uses WebSockets to serve content to all clients, the idle connections take up 2,9 unnecessary gigabytes.

Figure ‑: Theoretical development of memory usage of idle WebSockets clients with Play.

This thesis has established the fact that WebSockets is faster than request/response based HTTP. But seeing the extra memory usage, the tradeoff is not large enough. Furthermore, my tests focused on real time use cases. For a web page without real time content, the differences in latency would be minimal.

Another benefit of WebSockets we have seen is that it has less overhead. Using the protocol as basis for a new standard for all client/server communication, much of this benefit disappears. WebSockets does not have headers in the same sense as HTTP. To implement a similar request/response pattern as HTTP with WebSockets, this would have to be added.

These are the main reasons why WebSockets is a separate protocol designed for real time. The background gave a brief overview of the upcoming HTTP/2.0 specification (see section 2.3). While the specification proposes changes to HTTP including push behavior [11], it does not mention WebSockets.

In my opinion, all web pages should use bundling and minification of CSS and JavaScript. This means that all code are compressed into two single files: one for CSS and one for JavaScript. As a result, a browser can get the whole page using a minimum of three requests. Any images will increase this number. Doing this allows for better utilization of HTTP. This makes the benefits of WebSockets decrease even more.

While WebSockets may never become the basis for a new generation of HTTP, the results from the idle connections experiment has a side effect. Idle WebSockets connection take up resources. Some application may not need to send updates to the clients all the time. With a set, rather large, interval that you know, polling might actually be a better choice. No resources will be wasted serving idle connections, and users get their data when they need it. At such low loads, the extra network traffic becomes insignificant. An application with an interval so large that polling becomes better than WebSockets is not a common case. To keep the benefit, messages going from client to server should be in the same order of magnitude.

# Conclusion

# Frameworks

This chapter covers the problem statements regarding the different frameworks. The problem statements are reproduced in the paragraphs below. Before drawing up the final conclusion for each, I will summarize experiences made during the project. This includes both usability and performance.

**I: What framework do I consider to be the best for real time web applications?**

SignalR and Socket.IO stand out as the most enjoyable frameworks to work with. They both provide a simple, yet powerful programming model. Fallbacks to older transports as well as serialization of data happens behind the scenes no matter which you use.

Socket.IO is built according to the mantra of Node.js (TODO: homepage). It is a lightweight library for real time. The communication interface uses events, which are easy to relate to an work with in JavaScript. Writing write automated tests for event driven applications can be tricky. This is no issue with Socket.IO as it provides a client library for Node.js that you can use.

The RPC model of SignalR is clever and intriguing. Calling a function from the server and vice versa, is even simpler to relate to than events. Some frameworks that has "magic" going on behind the scenes like SignalR, can get confusing. With SignalR though, this is not the case, especially because of the thorough documentation. Plenty of tutorials makes getting started with SignalR a breeze. And if you are an experienced .NET developer, you should have no trouble using it.

The performance tests showed Socket.IO as winner in the latency category. Lightstreamer and SignalR follows a little behind with a large gap down to SockJS and Play. Node is single threaded, which implicates that Socket.IO doesn't scale as well as Lightstreamer or SignalR. But since it is such a light weight platform, one can have several instances running for the same cost as one IIS or Lightstreamer server. This was the tendency of the results for use of machine resources as well. Both Node frameworks used a lot less than any of the others.

When it came to network performance, Socket.IO once again was best. However, it does not support HTTP-streaming as fallback. As a result, all frameworks first fallback[[45]](#footnote-45) is better than Socket.IO's. SignalR is behind all the others in this category, which is because of how it handles messages. Other data than just the message payload gets sent with each message.

Based on my experiences I must say that Socket.IO is the best all-round framework. But, as Node isn't all that widespread for enterprise web applications, this would require at least one new server. If the importance of performance outweighs usability, Lightstreamer can be just as good or even better. For .NET users I say that Lightstreamer should be considered if performance matters most. Else, there is no doubt that SignalR is best.

**II: Is it always useful to use a framework?**

Play Framework gave me insight into how it is to implement real time almost without support. This was a far more complicated process than with any framework. Lightstreamer was actually not far behind the complexity of the Play application. But when you look at the whole picture, Lightstreamer offers a lot more than Play. Play has no client handling, no serialization, no reconnect mechanism and no fallbacks out of the box. My application only offered two fallbacks. Introducing more, would introduce more complexity.

This leads me to the conclusion that you always want the support a framework offers. Even if fallbacks are not required, it is harder to do it manually. A framework also provides more flexibility. If fallbacks become a requirement after all, it is already there.

# WebSockets versus HTTP

In this chapter I will focus on the problem statements that pin WebSockets against HTTP. As with the previous chapter, these are reproduced in the paragraphs below.

**III: Does WebSockets outperform the old, established HTTP methods for real time?**

WebSockets outperform all the HTTP methods when it comes to latency. A surprising exception in my tests was Server-Sent Events. But, as I discussed in section (TODO), it is likely that WebSockets is better with higher loads. HTTP-streaming isn't that much slower, but it is noticeable. My tests had a strong focus on server to client communication. The other way around though, HTTP rely on normal requests. Long-polling and polling are a lot slower than WebSockets. Therefore, I have reason to believe that WebSockets is far better for applications with a higher percentage of client to server communication.

This also applies to network traffic. HTTP-streaming and Server-Sent Events almost keep up with WebSockets in push focused applications. The more client to server communication you get, though, the bigger the benefit of WebSockets.

For real time applications with high message frequency, WebSockets require less memory and processor than HTTP. As the frequency decreases, the more memory the server wastes serving idle connections. With a large enough interval between updates, that the developers know up front, polling is a good choice. There are not too many use cases that are like this, though.

Based on the observations I have made, WebSockets is better than HTTP in every aspect for real time applications. But for pure push applications, I will not say that WebSockets outperforms HTTP, even though it is better. With respect to client to server communication, there is no doubt that WebSockets is faster and use less resources. Some rare cases with a large push interval and little client to server communication, can cause polling to be a better choice than WebSockets.

**IV: Can WebSockets be the foundation for the next generation of HTTP?**

Idle WebSockets connections take up memory. With the request/response pattern of HTTP, idle connections don't exist. Using WebSockets as a foundation for a HTTP protocol would mean trading a significant increase in memory usage for faster connections. As header data would have to be added to WebSockets, savings on network performance will be minimal.

HTTP/2.0 incorporates an idea from real time, with the ability to push data to clients. But the protocol remains unidirectional. If developers take care to bundle CSS and JavaScript, the loading of web pages will become a lot faster with HTTP/2.0.

Meteor use real time for every aspect of client/server communication, but it serves files via normal HTTP. Still, it seems that the developers believe in WebSockets for all communication in the long run. With real time so central, though, Meteor is most suitable for applications with many real time components in my opinion.

HTTP/2.0 surfaced during the initial research for this thesis. At the time, I saw benefits of using WebSockets instead of HTTP for some aspects. But with the current draft of HTTP/2.0, I cannot see a future where WebSockets become an all purpose protocol. It was built as a protocol for real time, and that is what it will remain.

# Further work

This thesis has a lot of potential for further work. In this chapter, I describe possible projects that can build upon the work I have done.

## Scalability

My work focused on usability as well as performance. In retrospect, it would be interesting to focus on performance alone. The tests I performed show clear trends between frameworks and transports. But an important aspect as scaling is not covered.

To be able to handle this, one have to change the approach. Using real browsers is not an alternative. The preferable client is a headless browser with full support for WebSockets. In a not too distant future, both Slimer and Phantom should be applicable. If not, the use of console applications is almost just as good. Both will allow for more clients on a single machine. Even more if you use more machines.

If I were to do this project, I would have changed the way I collected some of the data. This thesis used message frequency to see if and when messages were lost. For a larger test, one should only measure if. In other words, the server would just count the number of messages, not divide them into intervals.

## Separate message directions

The push nature of my tests had a large influence on the results. It seems that HTTP push technologies are not far behind WebSockets in performance. The other way, the results from long-polling and polling indicate that WebSockets is a lot better. But I did not perform any tests to find out for sure.

An interesting case would be to run tests with many more clients than I had, that focus on just one message direction. One test for server to client and one for client to server. The first will compare WebSockets to HTTP-streaming and Server-Sent Events. From client to server one needs to compare WebSockets to either POST or GET requests.

A project of this nature should not focus on different frameworks. To make it as even as possible, one should keep to one platform. The set up William P. Riley-Land uses in his blog post (TODO), is how I imagine it.

## Cluster of Node servers

I've paid a lot of attention to the lightweight nature of Node. In my tests, this was a huge benefit for both Socket.IO and SockJS. However, with higher loads, the single thread of Node should be overwhelmed. Node use little machine resources compared to both Java and C# counterparts. An interesting project would be to perform load tests where one compares a cluster of Node servers to for instance Lightstreamer and SignalR.

The ideal set up would have the same amount of Node servers as the other frameworks have available threads. All Node servers should run on the same machine. Further testing can involve the use of different load balancers for the Node servers.

SignalR does not depend on IIS. It can use OWIN for self hosting (TODO). This should make the server a lot more nimble. Testing this versus both Lightstreamer and a Node cluster could make this project even more interesting.

Depending on the time at hand, the work you can do with this is enormous. Further expansion can use a more advanced setup. You can introduce several servers for Lightstreamer and SignalR, running on several machines. The same machines can host one Node cluster each.

This would require many clients to be able to get viable data. Maybe as many as 100000 or more. Hence, you will need access to a great number of machines. The way you collect data will also have to be changed to fit the setup. A simple solution is to have the servers send a set number of messages as fast as possible, and then measure the time. You can also turn this around and have the clients send messages to the servers.

## Atmosphere

Over the last year and a half, Atmosphere has gained popularity. Lightstreamer was my Java alternative for real time. This framework has a different approach than Atmosphere. Atmosphere build on the same concepts as Socket.IO and SignalR.

Using those three frameworks as basis for a thesis, one could get a true showdown of platforms; Java, Node and ASP.NET/C#. One can use several approaches, but I think there should be a strong focus on performance. During my work, I learned just as much about the usability of the frameworks during the performance tests. A project with a load test of these three frameworks would require development with each. Experiences from this work can then be shared in the same manner as in part one of my project. (TODO: cross).

## HTTP/2.0

"Can HTTP/2.0 improve existing HTTP methods for real time?" This would be an interesting problem statement for a project based on this thesis. Doing the same, small scale tests, one can compare my data to data recorded for HTTP/2.o. To get a better picture, one can perform more large scale tests focusing on push.

A technique like long-polling should be able to benefit a lot from the push mechanism. Maybe it can be used almost like Server-Sent Events? If that is the case, one should see how it performs. A result showing that long-polling over HTTP/2.0 perform just as well as Server-Sent Events, would be intriguing.

# Sources

# Appendix

# Appendix A: E-mail communication with Weswit

The following is the e-mail communication between myself and Weswit regarding settings for the Lightstreamer Server.

-----------------------------------------------------------------------------------------------------

On Thu, Feb 13, 2014 at 5:50 PM, Simone Fabiano  
<[simone.fabiano@lightstreamer.com](mailto:simone.fabiano@lightstreamer.com)> wrote:

Hello Kristian,  
  
While I start looking at your project, could you please send me the server configuration file?  
Also, can you tell me the specs of the hardware running the tests?   
  
Thanks,  
Simone.  
Simone Fabiano  
Software Engineer

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On 2014-02-14 19:06, Alessandro Alinone wrote:

HI Kristian,  
  
We are reproducing the tests.  
We have not finished yet, and will continue on Monday, but there is  
already an important observation to mention.  
When testing Java-based servers, it is fundamental to warm up the  
JVM, before collecting the results. This is the foundation of all  
Java-based benchmarks. So, you should make the test last much longer  
and discard the first few minutes of results. This should lead to more  
significant stats.  
That being said, there are several other leverages available to tune  
performance, such as: GC algorithm, logging, delta delivery, JSON  
parser optimization, etc.  
On Monday we will focus on those other aspects. On your side, if you  
have a chance to re-execute the test with the JVM warm up, keeping all  
the rest fixed, this could help.  
  
Thanks and have a nice week-end,  
Alessandro

-----------------------------------------------------------------------------------------------------

On Mon, Feb 17, 2014 at 1:15 PM, Kristian Johannessen <[kjohann@student.matnat.uio.no](mailto:kjohann@student.matnat.uio.no)> wrote:

Hi!  
  
Just thought I should let you know that running the test over a longer timeperiod gave very improved results. After about 20 seconds, the latency stabilizes in the range of 15-25 ms. This is about the same performance as SignalR.

Kristian

-----------------------------------------------------------------------------------------------------

On 2014-02-17 18:43, Simone Fabiano wrote:

Hi Kristian,

I've run your tests with a few tweaks using two laptops attached to our local LAN via cable. I tested Socket.IO and Lightstreamer to have a reference on such different hardware.

I didn't use your chart server because I didn't have the chance to setup IIS on my machine, so I added a simple average calculation directly on the client code. Also I didn't limit the number of updates (I set it to 1000) and I made my code calculate the average latency every 100 updates (after 100 updates it resets itself). Then I gathered the average from each browser for the updates 301 to 400 and I came up with basically the same value for both Lightstreamer (136.586ms) and Socket.IO (136.795ms).

These are the tweaks I performed on the Lightstreamer installation

I changed VM configuration to use the G1 garbage collector and set a max pause of 5ms (don't do this unless you're using at least Java 7 as with previous VM the G1 gave some weird results). To do that edit the LS.bat file from the bin/windows folder of the lightstreamer installation and add -XX:+UseG1GC -XX:MaxGCPauseMillis=5 to the JAVA\_OPTS variable.

I disabled the delta delivery of Lightstreamer: using json updates this feature is completely useless. In the conf/lightstreamer\_conf.xml file search for the delta\_delivery option and set it N (<delta\_delivery>N</delta\_delivery>)

I disabled the log. Not sure if this had any impact at all but I decided to simplify the case

As I already told you I got the values after a few minutes in order to properly warm up the JVM

I set the max delay to 0 in the Lightstreamer configuration (<max\_delay\_millis>0</max\_delay\_millis>)

At that point I was thinking about upgrading the Java JSON parser version (there is a newer one, 2.2.3) but I preferred to skip that step.

Finally I have a couple of observation about the tests:

Limiting the test to 60 connections prevent the test to actually verify the scalability of the various servers. (obviously I'm pretty confident in the Lightreamer scalability but this is a general observation)

Using browsers to run load tests is probably not the right choice as you're adding a lot of overhead to each client. We usually prefer to use java clients when we have to run our load tests. (currently our own java library does not support websockets, and I'm not sure about [socket.io](http://socket.io/) java clients, anyway I wrote a very simple java client for a test once. It's not general purpose but it can easily be adapted I suppose. I also suppose you've not enough time for this :-) anyway see <https://github.com/Weswit/Lightstreamer-toolkit-socket.io-benchmark/tree/master/client/src/loadtestclient/client> )

My final note is about the messages you use to calculate the latency: I see that each client calculates the latency only considering the messages sent by itself. This may theoretically give you mixed results based on how a server implements its logic: imagine what happens if a server always send a message back to the client that originated it first and only after starts to send the same updates to all the other clients. (Noite that I didn't change that for my calculations)

HTH

Good Luck with your Thesis!

Simone.

--   
**Simone Fabiano**  
**Software Engineer**  
Weswit :: [www.weswit.com](http://www.weswit.com/)  
Lightstreamer :: [www.lightstreamer.com](http://www.lightstreamer.com/)  
  
E: [simone.fabiano@lightstreamer.com](mailto:simone.fabiano@lightstreamer.com)  
Skype: w.simone.fabiano  
  
*Lightstreamer, selected Cool Vendor by Gartner, is the most advanced server for real-time bi-directional data delivery over the Web. More Than Just WebSockets...*

-----------------------------------------------------------------------------------------------------

Hi!

I just tried to do a couple of long running test (4 minutes) with the tweaks you suggested. It does not seem to improve the results I got yesterday when I ran my tests longer (but with no other tweaks except from the max\_delay\_millis setting set to 0). There are minimums as low as 9ms and peaks as high as 30ms, but the average is a little under 20 (which is around the same as SignalR).

So, maybe the tweaks have no effect on my system (since it doesn't have a lot of performance), or the low load makes it take no effect. Anyways, I will bring your results into my discussion regarding the results. This discussion will also cover the observations you made about the tests (I have thought of the same things).

I can give some insight into why my tests are as they are:

60 connections was not the plan initially, but limitations with using browsers made that the maximum (uses almost 7 GB of my 8GB of RAM with 30 browsers). My initial idea was to use a headless browser, but I cannot use Phantom. It does not support WebSockets and it would probably have the same behavior as Chrome. With Chrome, most servers refuse any connections after 6 (HTTP), even from different browsers. There is Slimer, which is Gecko based, but this is not fully headless and not mature enough for me to find it reliable. Therefore I landed on Firefox driven by Selenium - not optimal, but it gives some results at least.

Another reason to use browsers, is that I get a common testbed for all frameworks and all transports. I am not only comparing the performance of the different frameworks you see. The more interesting question in my thesis is whether or not WebSockets outperforms HTTP for various uses (real-time and "traditional" request/response sites like for instance Wikipedia). I'm basically trying to see if WebSockets can be the basis for the next generation of HTTP (sort of). Using Java clients for this purpose (or other languages) seemed to force me to use different implementations of the clients for both framework AND transport. With browsers, there are just one file different on the client for each framework.

Still, in retrospect, I probably would have done it with Java clients (or something else) if I could have done my work all over again. This load testing is just the second half om my thesis, the other was a more subjective look at how the frameworks were to work with. If I could go back, I probably would have focused solely on load testing.

Regarding the average calculation: Yes, I am aware of that the method has its flaws, but it felt like the best solution. It measures how long it takes for a client to get a response to its own message. If some servers send back to the originating client first and others last (isn't the first alternative most logical?), that is a possible source of error that will be discussed in my thesis.

Thanks a lot for your help! I've learned a lot from this - especially regarding JVM warmup :)

Regards

Kristian Johannessen

# Appendix B: All standard charts

This appendix contains all charts that can be derived directly from the raw data. Some of the graphs in the thesis are made from combining this standard data. These graphs are not included in this appendix.

**Messages sent from clients**

**WebSockets**

**Server-Sent Events**

****

**HTTP-Streaming**

****

**Long-polling**

****

**Polling**

****

**Messages received by server**

**WebSockets**

****

**Server-Sent Events**

****

**HTTP-Streaming**

****

**Long-polling**

****

**Polling**

****

**Messages sent from server**

**WebSockets**

**Server-Sent Events**

****

**HTTP-Streaming**

****

**Long-polling**

****

**Polling**

****

**Average latency**

**WebSockets**

**Server-Sent Events**

****

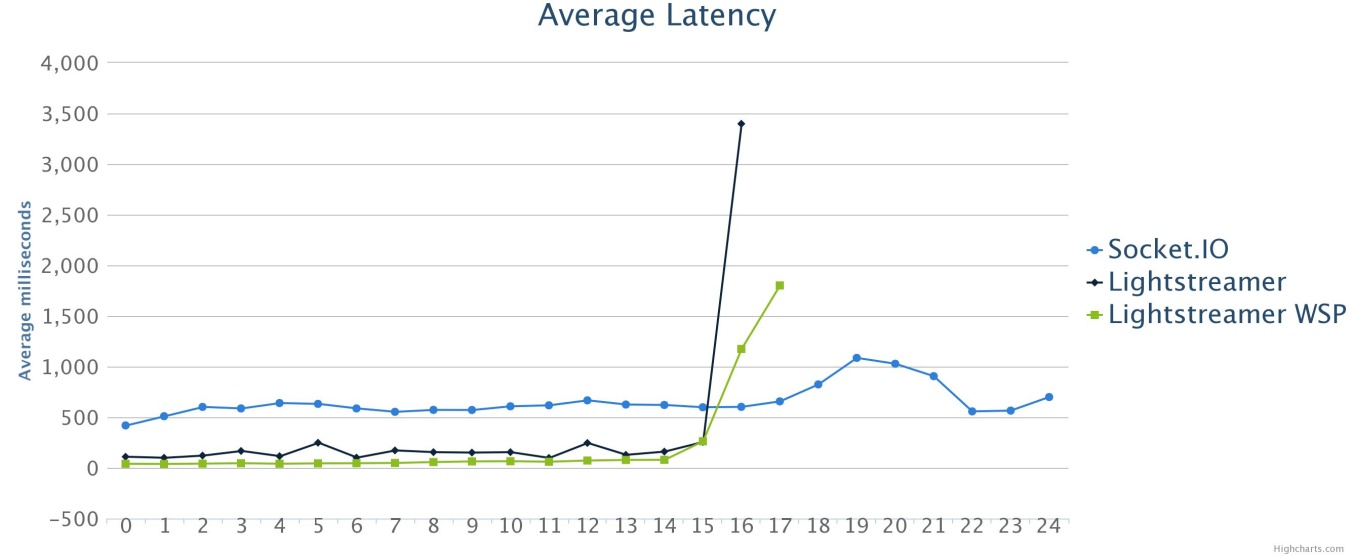
**HTTP-Streaming**

****

**Long-polling**

****

**Polling**

****

**Median Processor usage**

****

**Max memory consumption**

****

**Bytes sent/received (calculated)**

****

**Bytes sent/received (captured)**

****

**Total average latency**

****

1. The data access layer in an application refers to all code that access an external storage unit, normally a database. [↑](#footnote-ref-1)
2. Embedded objects consisted mostly of images, but also some early forms of style sheets. [↑](#footnote-ref-2)
3. 17902 vs. 57915 words. [↑](#footnote-ref-3)
4. Internet Engineering Task Force - Request for Comment series: see http://www.rfc-editor.org/ [↑](#footnote-ref-4)
5. Since this was written, NBIM has gotten new homepages. The counter is no longer there. [↑](#footnote-ref-5)
6. 87 100 000 bytes \* 8 = 696 800 000 bits / 10242 = 665 Mbits [↑](#footnote-ref-6)
7. 200 000 bytes \* 8 = 1 600 000 bits / 10242 = 1.526 Mbits [↑](#footnote-ref-7)
8. In order to write C# code on Android and iPhone, one has to use the cross platform framework Xamarin (https://xamarin.com/). [↑](#footnote-ref-8)
9. A remote procedure call is when a client calls a method situated on a server directly. For more information see rfc5531 [123]. [↑](#footnote-ref-9)
10. An Actor is a construct used to form hierarchies of entities that handle concurrency through messaging [124]. [↑](#footnote-ref-10)
11. An IDE is best explained as an advanced text editor designed to write code. It usually offers various support mechanisms like code completion, refactoring, debugging and keyword highlighting. [↑](#footnote-ref-11)
12. The operating system itself is not free. [↑](#footnote-ref-12)
13. Visual Studio Express is one of these products. [↑](#footnote-ref-13)
14. This can be observed when the Play server throws an Exception. It is often of a type within the org.jboss package. [↑](#footnote-ref-14)
15. [www.knockoutjs.com](http://www.knockoutjs.com) [↑](#footnote-ref-15)
16. WebSockets is the highest level possible, whereas polling is the lowest. [↑](#footnote-ref-16)
17. Examples of use: Trello, an online cooperation and project planning tool (<https://trello.com>). Sails.js, a web application framework with real-time at its core (see section 4.4) [↑](#footnote-ref-17)
18. See Play’s homepage at the bottom [38] [↑](#footnote-ref-18)
19. First commit 2011: <https://github.com/SignalR/SignalR/commits/master> [↑](#footnote-ref-19)
20. SignalR on Nuget: <https://www.nuget.org/packages/Microsoft.AspNet.SignalR/2.0.2> [↑](#footnote-ref-20)
21. Forking a project means that you use that project as a base for your own. See <https://help.github.com/articles/fork-a-repo> for more information. [↑](#footnote-ref-21)
22. According to Node.js’s homepage [54], its event-driven model makes it lightweight and efficient. [↑](#footnote-ref-22)
23. [www.ign.com/?setccpref=US](http://www.ign.com/?setccpref=US) (American IGN). [↑](#footnote-ref-23)
24. The subscription modes are RAW, MERGE, COMMAND and DISTINCT. [↑](#footnote-ref-24)
25. MVVM (Model View ViewModel), MVC (Model View Controller) and MVP (Model View Pattern) are referred to as MV\*. [↑](#footnote-ref-25)
26. IntelliSense is a support feature in IDEs that provide automatic code completion. [↑](#footnote-ref-26)
27. An example of an absolute path to a function is: window.someGlobalObject.someGlobalFunction(). [↑](#footnote-ref-27)
28. Jackson is a library for JSON handling in Java [125]. [↑](#footnote-ref-28)
29. There is one documentation for the 1.x versions and another for 2.x. Play has more separation, with documentation for both 2.0, 2.1.1 and 2.2.0 for example. [↑](#footnote-ref-29)
30. Class references in C# are similar to Javadocs for Java. [↑](#footnote-ref-30)
31. Following conventions you would ”call” a function called “someFunc” on the client. The server has named this “SomeFunc” though in order to follow C# naming convention. [↑](#footnote-ref-31)
32. If you declare a variable without the keyword “var” in front of it in JavaScript, you declare it globally, even from within a scope. Outside a scope, it is global even with “var” in front [126]. [↑](#footnote-ref-32)
33. A quick Google search reveals a lot of articles and blogs about Meteor. This indicates that Meteor has the attention of many people in the web development community. [↑](#footnote-ref-33)
34. A headless browser is a browser without the graphical user interface (TODO: what is a headless browser - blog). [↑](#footnote-ref-34)
35. I implemented the setup with SignalR first. I therefore chose to use its terminology. [↑](#footnote-ref-35)
36. If one program is running, it is the only program using the processor. Thus it uses 100% of the processor’s time. [↑](#footnote-ref-36)
37. A variation of 15 / 120 = 12,5%. [↑](#footnote-ref-37)
38. 161,7 / 33,1 = 4,89. [↑](#footnote-ref-38)
39. 605,6 / 52,6 = 11,51 [↑](#footnote-ref-39)
40. Server Sent Events actually beats WebSockets by 0,3 milliseconds. As this is close to 0, I deem them just as good. [↑](#footnote-ref-40)
41. WebSockets averaged 374,64 milliseconds. Server Sent Events averaged 490,44 milliseconds. 374,64 + 31% = 490,44. [↑](#footnote-ref-41)
42. Polling is as far away from WebSockets as you can get. [↑](#footnote-ref-42)
43. Using the worst case to show a worst case scenario don’t seem too representative to me. [↑](#footnote-ref-43)
44. 3.000.000 kilobytes /10242 = 2,9 gigabytes. [↑](#footnote-ref-44)
45. The ”first” fallback is the first transport a framework will try to fall back to if WebSockets is not supported. [↑](#footnote-ref-45)