

Scala.js networking made easy

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Abstract

1 Introduction

- Relevance: importance of networking for Scala.js
- Motivation: Many JS APIs
 - Websocket
 - Comet
 - WebRTC
- Motivation: Many network programming models
 - Akka
 - RPC (type safe)
 - Steams (scalaz, akka-stream)
- Plan/Contributions

2 Transport

- This section, `scala-js-transport` library, main contribution

2.1 A Uniform Interface

In order to interchangeably use different means of communication, we begin by the definition of an interface for asynchronous transports. This interface aims at *transparently* modeling the different underlying technologies, meaning it does not add functionality but simply serves as a delegator.

```
trait Transport {  
  type Address  
  def listen(): Future[Promise[ConnectionListener]]  
  def connect(remote: Address): Future[ConnectionHandle]  
  def shutdown(): Future[Unit]  
}  
trait ConnectionHandle {  
  def handlerPromise: Promise[MessageListener]  
  def closedFuture: Future[Unit]  
  def write(message: String): Unit  
  def close(): Unit  
}  
type ConnectionListener = ConnectionHandle => Unit  
type MessageListener = String => Unit
```

Listing 1: Definition of the core networking interfaces.

A *Transport* can both *listen* for incoming connections and *connect* to remote *Transports*. Platforms limited to act either as client or server will return a failed future for either of these methods. In order to listen for incoming connections, the user of a *Transport* has to complete the promise returned by the `listen` method with a *ConnectionListener*. To keep the definition generic, *Address* is an abstract type. As we will see later, it varies greatly from one technology to another.

ConnectionHandle represents an opened connection. Thereby, it supports four type of interactions: writing a message, listening for incoming messages, closing the connection and being notified of connection closure. Similarly to *Transport*, listening for incoming messages is achieved by completing a promise of *MessageListener*.

The presented *Transport* and *ConnectionHandle* interfaces have several advantages compared to alternative found in other languages, such the WebSocket interface in JavaScript. For example, errors are not transmitted by throwing exceptions, but simply returned as a failed future. Also, some incorrect behaviors such as writing to a no yet opened connection, or receiving duplicate notifications for a closed connection, are made impossible by construction. Thanks to support of futures and promises in Scala.js, these interfaces can be cross compiles to both Java bytecode and JavaScript.

2.2 Implementations

The scala-js-transport library contains several implementations of *Transports* across three communication technologies: WebSocket, SockJS and WebRTC. This subsection briefly presents the different technologies and their respective advantages. Table 1 summarizes the available *Transports* by platform and technologies.

Table 1: Summary of the available *Transports* by platform.

Platform	WebSocket	SockJS	WebRTC
JavaScript	client	client	client
Play Framework	server	server	-
Netty	both	-	-
Tyrus	client	-	-

2.2.1 WebSocket

WebSocket provides full-duplex communication over a single TCP connection. Connections are established by emitting an HTTP request to initiate a handshake, the HTTP connection is then upgraded to become the WebSocket connection. This mechanism allows WebSocket to be widely supported over different network configurations.

WebSocket is also well supported across different platforms. Our library provides four WebSocket *Transports*, the native JavaScript client, the Play Framework server, the Netty client/server and the Tyrus client.

While having all three Play, Netty and Tyrus might seem redundant, each of them has its advantages. Play is a complete web framework is more suitable to build every component of a web application. On the other hand, Netty might be more

suitable to build a standalone WebSocket server. Since Play is itself based on Netty, it adds some overhead in term of performances and dependencies. Regarding client side, the Tyrus library offers a standalone WebSocket client which is lightweight compared to the equivalent Netty implementation. It would therefore be more suitable for a mobile JVM client.

2.2.2 SockJS

SockJS is a WebSocket emulation protocol which fallbacks to different protocols when WebSocket is not supported. It supports a large number of techniques to emulate the sending of messages from server to client, such as AJAX long polling, AJAX streaming, EventSource or streaming content by slowly loading an HTML file in an iframe. These techniques are based on the following idea: by issuing a regular HTTP request from client to server, and voluntarily delaying the response from the server, the server side can decide when to release information. This allows to emulate the sending of messages from server to client which is not supported in the traditional request-response communication model.

The `scala-js-transport` library provides a *Transport* build on the official SockJS JavaScript client, and a second one for the Play Framework via a community plugin. Netty developers have scheduled SockJS support to be included in the next major release.

2.2.3 WebRTC

WebRTC is an experimental API to allow peer to peer communication between web browsers. Initially targeted as audio and video communication, WebRTC also provides *Data Channels* to send arbitrary data. Contrary to WebSocket which only supports TCP, WebRTC can be configured to use either TCP, UDP or SCTP.

As opposed to WebSocket and SockJS which only need a URL to establish a connection, WebRTC requires a signaling channel between two peers in order to open the peer to peer connection. The signaling channel is not tied to a particular technology, its only requirement is to allow a back and forth communication between its peers. This is commonly achieved by connecting both peers to a WebSocket server which then serves as a relay for the WebRTC connection establishment.

To simplify the process of relaying messages from one peer to another, our library implements picklers for *ConnectionHandle*. Concretely, when a *ConnectionHandle* object connecting node *A* and *B* is sent by *B* over an already established connection with *C*, the *ConnectionHandle* received by *C* will act as direct connection

between A and C , hiding the fact that B is a proxy between the two nodes.

At the time of writing, WebRTC is implemented in Chrome, Firefox and Opera, and lacks support in Safari and Internet Explorer. The only non browser implementations are available on the node.js platform.

2.3 Wrappers

- Works fine with the raw api
- Akka
- Autowire (RPC)

2.4 Going further

- Testing infrastructure
- Two configurable browsers

3 Example: A Cross-platform Multiplayer Game

- Goal: Cross platform JS/JVM realtime multiplayer game
- History: Scala.js port of a JS port of a Commodore 64 game

3.1 Architecture

- Purely functional multiplayer game engine
- Clock synked, same game simulated on both platforms
- Requires: initialState, nextState, render, transport
- Result: Immutability everywhere
- Result: everything but input handler & UI is shared

3.2 Compensate Network Latency

- Traditional solutions (actual lag, fixed delay with animation)
- Solution: go back in time (Figure)
- Scala List and Ref quality and fixed size buffer solution

3.3 Implementation

- React UI (& hack for the JVM version)
- Simple Server for matchmaking
- WebRTC with SockJS fallback
- Results: 60FPS on both platforms, lag free gameplay
- Results: Lag Compensation in action (Screenshots)

4 Related Work

- Js/NodeJs, relies on duck typing
- Closure
- Steam Engine/AoE/Sc2/Google docs

5 Conclusion and Future Work

- Web workers
- scalaz-stream/akka-stream wrappers
- More utilities on top of Transport

[1]

References

- [1] J. Y. Gil. $\text{\LaTeX}2_{\epsilon}$ for graduate students. manuscript, Haifa, Israel, 2002.