#### Building a video conference (WebRTC) controller with Elixir

Or how Elixir was introduced into VoiSmart

#### Hello!

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#### **About us**

15 years into the VoIP industry, building PBX and voice termination turnkey solutions using open source softswitches, like Asterisk, FreeSwitch.

Mostly coded with Python and the Twisted Networking engine, doing multi process dialog & coordination with RabbitMQ.

In 2017 we jumped into the video conference bandwagon.



#### Telco is hard...

- Many concurrent tasks running
   Call control, logging, device states, ...
- Many identical tasks running
   Without shared data between them
- → Events everywhere

Each call is very chatty about what is doing



#### Telco is realtime...

You cannot wait to handle a call Everything served in few milliseconds

#### → Async, async, async!

A phone system is async, several things may happen when you're serving a call

Python is not the right tool here But using Twisted and RabbitMQ helped a lot in being async and distributing work

# A video conference system has the same requirements ?

#### Yes! It does :)

Let's see why and how we handled it, better.

## AN KY

#### Requirements

- → Web based conference Because our roots are in the www
- → Should carry video It's a \*video\* conference
- → Should carry audio And should connect the PSTN
- → Should carry chat messages For quick text snippets sharing

#### WebRTC

Simple APIs for rich Real Time multimedia communications. But signalling is out of scope for WebRTC.

(so just supporting webRTC means almost nothing...)

### Mesh

Pros

Lowest delay Best compatibility **Cons** Does not scale High CPU usage High bandwidth

#### Image courtesy of https://webrtchacks.com



#### Image courtesy of https://webrtchacks.com



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### Signalling

#### A way to communicate Session Descriptions to others

You have to build your own protocol and transports

### Signalling #2

Not only webRTC stuff, but also: Instant messages Audio events Call control (kick, mute, join, quit...)

### Signalling #3

Not only client 2 server, but also inside the server: Controlling the SFU (janus) Controlling the audio bridge (freeSwitch) Signalling between clients

#### this is Realtime stuff

Continuous flow of asynchronous events, from multiple sources, that must be handled concurrently

# Things may also fail

- Lost connection to the clients (common)
- Lost connection to the media servers
- webRTC is still in development, a client may crash us
- Programming error :(



#### Tools needed for the job

- → A tool to ease async message passing between functional units
- Every unit may do "something" while waiting for events

So the tool should allow me to create live "loops" easily

- Delays on some dialogs should not slow things down
- → Issues on some units should not crash the system

#### **Elixir to the rescue!**

Because is Erlang after all

#### **Our architecture**

Or what we learned so far

# Use umbrella apps

Split your services in separate apps to ease development

- One for messages
- One for video
- One for audio
- One for auth
- One for logging
- And web, and config APIs, etc

# Log everything

#### With a correlation ID

And a machine parsable format

#### Authorization & Authentication

JWT between apps, using **;joken** JWTs contains all **[A,M,F]** tuples for the logged user Zuul, a service to handle users/permission Each service registers/updates own permissions via Zuul on app start

### **Roles handling**

Group permission into roles Delegated to a subapp, depends on the others On system start, registers/updates roles via Zuul

# defguarded

A macro that checks JWT tokens Put your claims into the token and let the macro check if you can call the function

( a near miss with elixir 1.6 defguard/defguardp ) ( split entry point with real implementation or tests will suffer ) The token cache/opaquer A revoking mech for JWTs JW/Ts can also be long (Putting them into Authorization header may become an issue) Map them with an opaque string Born with in-memory storage, easily migrated to Ecto, because it is a GenServer and the public APIs did not change

### **Client communications**

:phoenix websockets for live events and room protocol One channel dispatching to several modules
GraphQL APIs for the rest, with :absinthe
( basically, only 2 endpoints for everything )

# Identify your processes

Create a process only if needed (they're async, you should handle state, messages and crashes, so do that only if really needed)

Most of the time you can pass some data to a module and keep it in the calling process.

### **Processes are useful**

For long running tasks (e.g. token expire) For persistent connections

To keep temporary state (which may be used later/shared) To create a controlled crash chain

To handle disconnections (client disconnects does not close the room and notifies others)

For out-of-band processing (Adaptive Bitrate Controller)

#### how a room looks like



# Going :global

#### As the old adage goes "Always wrap process lookup in a module"

Moved from Registry to :global in few hours, Distributed and fault tolerant Video Conference!

# **Testing this stuff**

"Mock as a noun" we chat a lot with external entities, so the need of mocking modules that mimic the behaviour in normal and error conditions

#### **Testing this stuff #2** Aka (partial) integration tests

Because services are not isolated when deployed An umbrella app for integration tests It depends on all the others, started only in :test env

# **Testing this stuff #3**

Continuous delivery to the instance we use everyday for ourself and for demos Using :edeliver in GitLab pipelines

# Deploying

We don't do services, only ship software packages Packaged as RPM with Koji (Fedora build server) Release built with :distillery, :conform and included runtime

Building in a disconnected env is problematic for Elixir Upload already fetched deps, rebar(3) and hex to build it

#### **Future?**

#### A SIP bridge to let SIP video calls interact with the SFU based webRTC (mixing video, on the fly layouts and so on...)

#### Session Recorder

Using :ffmpex, a nice frontend to ffmpeg in elixir

#### **OSS Contributions**

:ecto unsafe\_fragment/1 by @xadhoom
:swoosh 2 feature PRs and 2 fixes PRs by @davec82/@xadhoom
:event\_socket\_outbound new package by @davec82
:elixir\_mod\_event features and fixes by @davec82/@xadhoom
:ffmpex improvements by @xadhoom
dialyzer fixes :phoenix, :gen\_state\_machine by @flaviogrossi
:websocket\_client reconnection fixes by @flaviogrossi
Janus SSL for rabbitMQ connections by @flaviogrossi

# **Open Telecom Platform**

Everything into a Telco application fits into what Erlang provides. A feeling hard to explain, but strong. By extension also Elixir, with a nicer syntax and a lot of higher level libs that ease development.

#### **Demo time!**

#### Thank you.

Made with love at VoiSmart by Matteo Brancaleoni @xadhoom Davide Colombo @davec82 Flavio Grossi @flaviogrossi Maybe You ? We're also hiring :)

**Questions?**