Janus: a general purpose WebRTC gateway

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Outline

1. A brief introduction
2. Some context
   - WebRTC and standardization activities
3. Writing a WebRTC gateway from scratch
   - Programmable Real-time Media Components
4. Janus: a general purpose WebRTC gateway
   - Modular architecture
   - What is it used for today, and by whom?
5. Next steps
Who am I?

- Someone not used to this weather!
  - From sunny Sorrento, Italy 😊

- Current activities
  - Just got my Ph.D @ UniNA
  - Co-founder @ Meetecho

- Worked on real-time applications for a long time
  - IETF participant
    - Several WGs
    - First time in IETF67 San Diego (2006)
  - Open source contributor
    - libbfcp, libmsrp, confiance, mediactrl, Asterisk, ...
    - Janus WebRTC gateway main author

- Getting older but, unlike whisky, not getting any better
  - https://twitter.com/elminiero
Christmas in Napoli!
Real-time media in browsers

- No standard solution!
  - No interoperability
  - Plugins need to be installed anyway

WebRTC = Joint standardization efforts

- Internet Engineering Task Force (IETF)
- World Wide Web Consortium (W3C)

- RTCWEB (IETF)
  - Real-Time Communication in WEB browsers WG
  - Defines protocols and formats to use

- WEBRTC (W3C)
  - Web Real-Time Communications WG
  - Defines UI and API to access devices
WebRTC reference architecture
Involving a gateway (and applications)
Involving different technologies as well
Do we really need a gateway?

• Several reasons for a YES, here
  • Relieve full-meshes (heavy on the client side)
  • Leveraging widespread technologies (e.g., SIP infrastructures)
  • Fixing things between implementations

• Reason for a NO?
  • You won’t go beyond interaction among few users
  • You don’t want an infrastructure
  • You don’t care about legacy stuff

“What is a WebRTC Gateway anyway?”

• http://webrtcchacks.com/webrtc-gw/
Real-time Media Components

- Writing a gateway from scratch is a heavy task
  - Implementation of the WebRTC protocol suite
- Bridge between “legacy” stuff (SIP, RTMP, etc.) and WebRTC
  - Needs to support both (WebRTC gateway)
  - What about statistics?
  - Reachability may be an issue
- Programmable interface
  - Different applications/technologies, different requirements
  - Dynamic management of media flows and users
  - Something à la MEDIACTRL?
The WebRTC protocol suite

- Signalling (well, sort of) and Negotiation
  - Javascript Session Establishment Protocol (JSEP)
  - Session Description Protocol (SDP) adaptation
- Connection Establishment and NAT Traversal
  - Session Traversal Utilities for NAT (STUN)
  - Traversal Using Relay NAT (TURN)
  - Interactive Connectivity Establishment (ICE)
- Media Transport and Control
  - Real-time Transport (and Control) Protocol (RTP/RTCP)
  - Secure Extensions to RTP (SRTP)
  - Datagram Transport Layer Security (DTLS)
- Media Transport and Control
  - Opus audio codec (MTI, Mandatory-to-implement)
  - VP8 video codec (MTI candidate)
- Generic Data
  - WebRTC Data Channels (SCTP)
Janus: a general purpose WebRTC gateway

“In ancient Roman religion and myth, Janus [..] is the god of beginnings and transitions, and thereby of gates, doors, passages, endings and time. He is usually depicted as having two faces, since he looks to the future and to the past.”

Janus: a general purpose WebRTC gateway

- A door between the communications past and future
  - Legacy technologies (the “past”)
  - WebRTC (the “future”)

Janus

General purpose, open source WebRTC gateway

- https://github.com/meetecho/janus-gateway
- Demos and documentation: https://janus.conf.meetecho.com
Modular architecture

- The core only implements the WebRTC stack
  - JSEP/SDP, ICE, DTLS-SRTP, Data Channels, ...
  - Modules for API over HTTP / WebSockets / RabbitMQ

- Application logic implemented in server side plugins
  - Users attach to plugins via the gateway core
  - The gateway handles the WebRTC stuff
  - Plugins route/manipulate the media/data

- Some proof of concept plugins implemented
  - Echo Test
  - Streaming (→ Live events!)
  - Conferencing (→ Meetecho!)
  - SIP Gateway (→ “Legacy” SIP!)
  - ...

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Extensible Architecture and API

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Modules and APIs
A few examples
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Extensible Architecture and API

Janus Gateway

Protocol messages

Plugin 1

Plugin 2

...
Extensible Architecture and API

User

Janus

PluginX

create session

sessionId

attach handle
(sessionId, Plugin X)

new user session

handleId

ok
Extensible Architecture and API

- User
- Janus
- PluginX

message (handleID, payload)

message from user

response

sync request
Extensible Architecture and API

User

Janus

PluginX

message
(handleID, payload)

message from user

async request

ack

event

ack

event

async event
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User
Janus
PluginX

negotiate WebRTC
(JSEP offer/answer)

notify WebRTC
(JSEP offer/answer)

ICE/DTLS/SRTP

RTP

Do whatever with plain RTP/RTCP
Plugins as “bricks”

- Each plugin is a feature, not an application
- Application can be composed out of different features
  - Features as “bricks” for a complex scenario

- A few examples...
  - Webinar with Q&A
    - Video Room (screen) + Video Room (speakers) + Audio Bridge (questions)
  - Video communication in social networks
    - SIP plugin (calls) + Echo Test (diagnostics) + Voice Mail (messaging)
  - Social TV
    - Streaming (TV channel) + Video Room (interaction)
Webinar with Q/A

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Social TV

TV live broadcast

User 1 video
User 2 video
User 3 video
User 4 video

Video MCU plugin
Streaming plugin
How do I...

- ... use Janus in my web app?
  - JavaScript library available *(janus.js)*
    - [https://janus.conf.meetecho.com/docs/JS](https://janus.conf.meetecho.com/docs/JS)
  - Several demos available to start from

- ... use Janus, but keeping my API?
  - Wrap the Janus API on the server side
    - [https://janus.conf.meetecho.com/docs/rest](https://janus.conf.meetecho.com/docs/rest)
    - [https://janus.conf.meetecho.com/docs/resources](https://janus.conf.meetecho.com/docs/resources)
  - Effective way to control what users can do
  - Also helps to orchestrate pool of Janus servers

- ... do this or that?
  - [https://janus.conf.meetecho.com/docs/FAQ](https://janus.conf.meetecho.com/docs/FAQ)
  - [https://groups.google.com/forum/#!forum/meetecho-janus](https://groups.google.com/forum/#!forum/meetecho-janus)
Anything wrong? Check the Admin API!

- Requests/response API to poll Janus
  - Query server capabilities
  - Control some aspects (e.g., enable/disable debugging)
  - Inspect handles and WebRTC “internals”

http://www.meeteecho.com/blog/understanding-the-janus-admin-api/
What is Janus used for today, and by whom?

- We use it ourselves for many things (obviously)
  - Web conferencing and Webinars
  - WebRTC-to-SIP gateway
  - Streaming of live events (e.g., IETF meetings)
- Many folks/companies also using it in creative ways!
  - E-learning
  - Coworking
  - TV broadcasting and Social TV
  - Home automation
  - Internet of Things
  - Mobile devices, Raspberry Pis, drones, etc.
- New third-party tools are starting to come out
  - [https://janus.conf.meetecho.com/docs/resources](https://janus.conf.meetecho.com/docs/resources)
    - New plugins for ad-hoc requirements
    - Server-side API wrappers (node.js, .NET, ...)
“Director” room @ IETF meetings

Completely WebRTC-based media streams
- Slides as a video feed from the beamer
- Static video feed from the room
- Dynamic video feeds for remote speakers
Meetecho: IETF meeting example

https://ietf.org/meeting/remote-participation.html
Meetecho: IETF recordings

https://www.youtube.com/user/ietf
A “silly” use case: The Jumping Janus!

https://www.youtube.com/watch?v=isGSnMlKcss
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Matrix demo at WebRTC Expo Miami 2015

Matrix wins Best of Show at WebRTC World!

https://www.youtube.com/watch?v=OMzDklvDS3c
“Matrix wins Best of Show at WebRTC World!”

https://www.youtube.com/watch?v=NpBStIlq6fM
Jangouts (for "Janus Hangouts" 😊)

https://github.com/jangouts/jangouts
SylkServer (more on that later, I guess!)

http://sylkserver.com/
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Lenovo’s AirClass

https://www.airclass.com
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https://www.sqwiggle.com
https://speak.io
Sqwigggle / Speak.io

https://www.sqwigggle.com
https://speak.io
Veeting rooms

https://www.veeting.com
What to do next?

- Finalize the WebRTC implementation
  - Better RTCP management
  - Implement multistream (Unified Plan)
  - Add octets (besides strings) to DataChannels
  - Keep up-to-date with newest stuff
- Keep on improving and fixing things
  - Implement admin API notifications (subscription)
  - Reference counters (currently in a PR)
  - Why not, some new transport modules
  - Maybe some changes to the pluggable architecture too?
- Help us improve this!
  - Play with it, more testing is important
  - Write your own plugins/applications!
Questions? Comments?