Kamailio as building block for Voice and Al platforms

Henning Westerholt VON:Builder New York October 2023

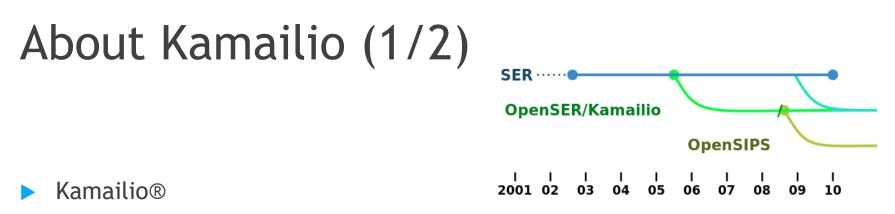
Agenda

- About
- Kamailio quick introduction
- Kamailio for Voice infrastructure
- Interacting with APIs
- Using machine learning and artificial intelligence tools
- Contact

About GILAWA



- We offer services for Real-Time Communication platforms
 - Consulting and Management
 - Administration/Developer trainings
 - Development and IT Operations
- Kamailio experience since 2007
- Independent and neutral service provider
 - No own end-user products
 - No vendor contracts
- Our customer are Internet Service Providers and Telephone Provider
- Germany, Europe, North-America, Asia and Middle-East



- Carrier grade SIP-Server published under GPL license, in development since 2001
- For the implementation of medium to large VoIP/Real-time Communications Plattforms
- Usual first use-case to scale existing PBX systems or to implement SBC functions
- Active and diverse developer team
 - Several core developers, different packager and frequent pull request
 - Regular maintenance releases
- Extensive core functions and over 200 extension modules

About Kamailio (2/2)

Kamailio was designed as a SIP proxy

- It is not PBX as for example Asterisk
- It is no "B2BUA", therefore no complete separation of incoming and outgoing traffic
- Learning Kamailio might be more difficult as learning other telephony software
 - Kamailio is not configured, but programmed
 - Solid SIP knowledge not only suggested but required
- Kamailio can be usually easily operated and maintained
 - Stable, good performance
 - Low dependencies from core and main modules

kamailio learning curve

it helps quite a bit: - imperative programming

- practicing

good knowledge of sip

Kamailio for voice infrastructure (1/2)



"SBC" functions

Load-balancing and security checks

Topology hiding

Network address translation for SIP and media

Transcoding

Application services

Device/Phone registration

Call routing

Authentication Authorization and Accounting



Presence services

Dialog-info Message-waiting Indication Other

Kamailio for voice infrastructure (2/2)





Translation functions

WebRTC gateway scaling

Adapting proprietary voice components

Interfacing with public commercial cloud services (like MS Teams or other)

IMS/NGN functions

Providing different IMS applications/functions for mobile networks Interfacing with different IMS protocols or commercial components

Deployment models

- Kamailio can be deployed in different ways, with Linux packages or container
- If you use docker for deployment, you usually end up doing your own docker packages as you need a specific configuration for your service
- Kamailio supports many pre-processing as command line parameter
- Recent Kamailio version also support "modparamx" for dynamic variable parameter evaluation
- With Kubernetes you usually will end up using environment variables for this information
- It's possible to to use ansible or saltstack to build the docker container, but usually a CI/CD infrastructure is used

Kamailio inside docker or kubernetes

CI/CD e.g. jenkins		
docker compose or k8s		
base docker image	Kamailio docker image	
e.g. Debian or alpine	Kamailio configuration	Environment variables
Data volumes, database sidecar, etc	Routing logic	IP addresses, defines etc

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Interacting with APIs

- Kamailio provides a wide variety of API interfaces
- Kamailio as API client
 - Recommendation to use a HTTP REST API, for example with JSON data fields
 - Both synchronous and asynchronous API calls are supported
 - Multiple API calls can be chained together in the configuration
 - Can be easily scaled depending on the requirements
- Kamailio as API server
 - Recommendation to use the JSONRPC for a REST HTTP API with JSON data fields
 - Kamailio provides embedded HTTP support, no external API server necessary
 - Can be easily scaled depending on the requirements

Using ML and AI tools

- Remember, Kamailio does not care about media (much), as usually this is the interesting information, we need to feed the data to external consumers
 - One option is to use the rtpengine-recording daemon, use storage for processing
 - Another option with less latency is the proc interface for one the fly processing
- Kamailio can of course provide all sorts of interesting events
 - Call data for getting information about caller or callee, duration, call quality etc..
 - Billing related data for detection of fraud or other malicious usage
 - IP addresses, User names, Password (hashes) etc..
- You could also interact a custom API by using the KEMI interface, for example with python scripts

Thank you

Thank you - any questions?

► Hope to see you at other conferences!

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Thank you

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