

Daniel-Constantin Mierla

Co-Founder Kamailio Project @miconda

VUC588 - April 8, 2016

History



(c) asipto.com



Kamailio overview









SIP Signaling Server

- SIP registrar
 - handle registrations from devices
- SIP location server
 - Iocate targets Icr, Ioad balancer, user location service
- SIP proxy server (router)
 - routing and re-routing huge volume of SIP traffic
- SIP application server
 - rich telephony services
 - instant messaging and presence
 - integration with third party applications
 - integration with web 2.0 and social networking

□ It is not

- SIP phone
- SIP metia server
- SIP back to-back user agent

It handles only signaling



Kamailio typical use cases

TWO BASIC ARCHITECTURES

main signalling server

edge signalling server



LOAD BALANCER

dispatcher module

- list of balancing nodes from file or database
- monitoring of nodes (activate/inactivate automatically)
- re-route in case of failure
- various algorithms: hashing, weight distribution, round robin, call load distribution, priority routing
- reload list of nodes without restart

```
# Dispatch requests
route[DISPATCH] {
          # round robin dispatching on gateways group 'l'
          if(!ds select dst("1","4")) {
                    send_reply("404", "No destination");
                    exit;
          xdbg("--- SCRIPT: going to <$ru> via <$du>\n");
          t_on_failure("RTF_DISPATCH");
          route(RELAY);
          exit;
# Re-route in case of failure
failure_route[RTF_DISPATCH] {
          if (t is canceled()) {
                    exit:
          # next node - only for 500 or local timeout
          if (t check status("500") || (t branch timeout() && !t branch replied())) {
                    if(ds next dst()) {
                              t on failure("RTF DISPATCH");
                              route(RELAY);
                              exit:
}
```



LEAST COST OR DID ROUTING



SIP FIREWALL

block DOS or dictionary attacks

part of default kamailio.cfg

- htable module
- pike module
- detect high volume of traffic from same source and block it for a specific interval of itme

```
# ip ban htable with autoexpire after 5 minutes
modparam("htable", "htable", "ipban=>size=8;autoexpire=300;")
....
route[SAFEGUARD] {
          # flood detection from same IP and traffic ban for a specific interval
          # be sure you exclude checking trusted peers, such as pstn gateways
          # - local host excluded (e.g., loop to self)
          if(src ip!=myself) {
                    if($sht(ipban=>$si)!=$null) {
                              # ip is already blocked
                              xdbg("request from blocked IP - $rm from $fu (IP:$si:$sp)\n");
                              exit;
                    if (!pike check req()) {
                              xlog("ALERT: pike blocking $rm from $fu (IP:$si:$sp)\n");
                              sht(ipban=>si) = 1;
                              exit;
          if($ua =~ "friendly-scanner") {
                    sl send reply("200", "OK");
                    exit;
}
```



TRANSPORT LAYER GATEWAY



core and tm module

- set transport in R-URI or outbound proxy address
- force transport via dedicated function

bridging networks

- public to private networks and back
- IPv4 to IPv6 and back
- any transport layer to another one and back

Force UDP route[TOUDP] { # round robin dispatching on gateways group '|' record_route(); \$du = "sip:nexthop.com:5060;transport=udp"; force_send_socket("udp:10.1.1.10:5080"); route(RELAY); exit; }

OFFLOAD SIGNALLING PROCESSING

edge proxy

- authentication
- nat traversal
- parallel forking
- serial forking
- call forwarding
- call blocking
- white/black listing
- DNS handling



DESIGNED FOR SCALABILITY



EXTRA FEATURES

plenty at signalling layer

- instant messaging
- presence
- gaming
- notifications

additional functions

- embedded XCAP server
- embedded MSRP relay
- IMS

extensibility

- define your new request types
- flexibility in handling unknown requests, headers, etc.





News Recent Years



rock solid SIP server excellence in SIP since 2001

□ security

- strong encryption algorithms for tls
- flexibility to work with many certificates at the same time
- dnssec, config file crypto tools
- performance, scalability and redundancy
 - full asynchronous processing layer
 - event driven api interface (json)
 - nosql: redis, memcached, mongodb, cassandra
 - distributed message queue between kamailio nodes
 - new memory managers
- flexibility
 - embedded http client for API interaction
 - dynamic memory manager selection
 - update of routing parameters without restart
 - more event routes
 - ison sip routing format

enhancements for typical use cases

- webrtc secure websockets with support for fragmentation
- classic sip to webrtc gateway, including RTP to SRTP
- volte ims: dozens of extensions
- sms routing handling
- more load balancing algorithms
 - call load distribution, relative weight, ...
- topology hiding and topology stripping
- user location partitioning and replication
- generic caching replication
- prepaid and call stateful control
- call rating limits
- detecting attacks and blocking them
- IPv6 cleanup



The Future

Kamailio 5.0

native configuration with embedded interpreters

- Iua, perl, python, .net, ...
- ability to reload routing logic
- highly optimized alternative of config with our own routing language
- continuous integration
 - unit tests
 - automatic builds
 - source code restructuring
 - review options for a new build system
- api driven processing
 - integration with external systems
 - keep kamailio as bare sip layer router
- your suggestions are welcome

• • • •



The 4th Kamailio World Conference May 18-20, 2016 Berlin, Germany

celebrating 15 years of Kamailio development



Celebrating 15 Years of Kamailio Development!

May 18-20, 2016 - in Berlin, Germany

www.kamailioworld.com

(c) asipto.com

highlights

- **D** Berlin, same nice location in the city center
 - half day technical workshops
 - two days of conference
- workshops
 - IMS tutorial
 - continuous integration
 - troubleshooting
 - alternative: visit to Fraunhofer FOKUS testbeds (to be confirmed very soon)
- conference
 - SIP from classic telephony to webrtc and volte
 - Kamailio and other open source projects well represented
 - Asterisk, FreeSwitch, Sems
 - security and scalability
 - typical use cases
 - experiences from operating large telephony platforms



Celebrating 15 Years of Kamailio Development!

May 18-20, 2016 - in Berlin, Germany

www.kamailioworld.com

(c) asipto.com



Questions?



Daniel-Constantin Mierla

Co-Founder Kamailio Project @miconda