

# Asynchronous SIP Routing

## SIP Express Router (SER) and Kamailio (OpenSER)

[SIP-Router.org](http://SIP-Router.org)

---

**Daniel-Constantin Mierla**

*Co-Founder Kamailio*

*asipto.com*



## Welcome to Kamailio (OpenSER) – the Open Source SIP Server

Kamailio (former OpenSER) is an Open Source SIP Server released under GPL, able to handle thousands of call setups per second. Among features: asynchronous TCP, UDP and SCTP, secure communication via TLS for VoIP (voice, video), SIMPLE instant messaging and presence, ENUM, least cost routing, load balancing, routing fail-over, accounting, authentication and authorization against MySQL, Postgres, Oracle, Radius, LDAP, XMLRPC control interface, SNMP monitoring. It can be used to build large VoIP servicing platforms or to scale up SIP-to-PSTN gateways, PBX systems or media servers like Asterisk™, FreeSWITCH™ or SEMS.

- Kamailio SIP Router at Google Summer of Code 2010
- SIP Router Devel Meeting, Berlin, June 8, 2010
- Listen VoIP User Conference – The SIP Router Project
- Remarks About v3.0.x Strong Stability
- January 11, 2010 – Kamailio (OpenSER) – New Major Version v3.0.0 Released
- September 01, 2009 – Kamailio awarded Best Open Source Networking Software 2009



### Recent News

- 2010-06-03: Kamailio Booth at LinuxTag 2010  
2010-06-02: Kamailio Presentation at LinuxTag 2010  
2010-06-01: VoIPToday Kamailio Interview  
2010-05-29: Kamailio and Freeswitch Integration, Jun 2, 2010  
2010-05-28: Kamailio at Amoocon 2010

- Download Latest Stable v3.0.2 -

### Pages

- [Home](#)  
[Features](#)  
[Download](#)  
[About](#)  
[Old Site](#)

### Documentation

- [Main Index](#)  
[Wiki Site](#)  
[Modules](#)  
[SIP Router Wiki](#)  
[Devel Guide](#)  
[Doxygen](#)

# History

2002      Jun 2005      Jul 2008      Aug 2008      Nov 2008      Oct 2009      Jan 2010

**SIP Express Router (SER)**

**SIP Router Project**

*Integration  
Completed*

v3.0.0

v1.5.0

**OpenSER**

**Kamailio**

**Other Forks...**

# Development Portals

- Open source, GPL SIP server – IETF RFC3261
  - Web: <http://www.kamailio.org>
  - SourceForge Project: <http://sourceforge.net/projects/openser/>
  - SIP Router Project: <http://sip-router.org>

The screenshot shows two related web pages. On the left is the SIP-ROUTER.ORG homepage, featuring a search bar and a sidebar with links for Menu, About, Answers, and Support. The main content area displays recent news items:

- Jan 18, 2010: [Siremis 1.0.0 Released](#)
- Jan 11, 2010: [Kamailio \(OpenSER\) 3.0](#)
- Dec 02, 2009: [Daily tarballs for 3.0 br](#)

On the right is the Kamailio (OpenSER) SIP server project page on SourceForge. It includes the SourceForge logo and navigation links for Find Software, Develop, Create Project, Community, Site Support, and About. The project summary page highlights the Kamailio (OpenSER) SIP server, developed by anomarme, henningw, juhe, klaus\_darillion, and miconda. It features a download button for "Download Now!" (kamailio-1.5.2-tls\_src.tar.gz, 3.6 MB) and a link to "View all files".



Awarded  
**Best Open Source  
Networking Software  
2009**  
By InfoWorld

# Usage in Germany

- Some of biggest VoIP deployments world wide
  - 1&1 (members of the management board)
    - Over 3 millions subscribers
    - Over 1.5 billion minutes per month
  - Sipgate
  - Freenet
    - 0.8 million subscribers
    - Hundreds of millions of minutes per month



<http://www.kamailio.org/w/references/>

# Features

**SIP proxy, redirect  
and registrar server  
user registration  
with  
digest authorization**

**Multi-domain support  
LDAP/H.350 support**

**Plug in module  
interface  
Perl programming  
interface**

**Customizable routing  
policy  
User location service  
IPv4-IPv6  
UDP/TCP/TLS/SCTP  
SIP translator**

**Instant Messaging  
Offline message  
service  
Presence server**

**ENUM lookup support  
Advanced routing  
(dispatching and LCR)  
Dialing support  
aliases and speeddial**

# Features

**OSPF support for peering  
Java SIP Servlet  
programming interface**

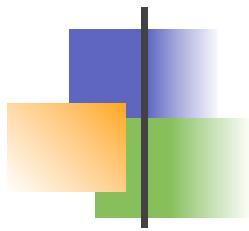
**Link any application to Kamailio using  
FIFO/UNIXSOCK/DATAGRAM/XMLRPC interfaces**

**Database API**  
**MySQL**  
**PostgreSQL**  
**UNIXODBC**  
**BERKELEYDB**  
**ORACLE**  
**Text files**  
**RADIUS**

**Gateway**  
**SMS**  
**XMPP**

**Accounting through log file,  
database or Radius/DIAMETER  
servers**

**NAT traversal**  
**Security**  
**permissions**  
**anti-DOS attacks**  
**User call preferences**  
**Call Processing**  
**Language**

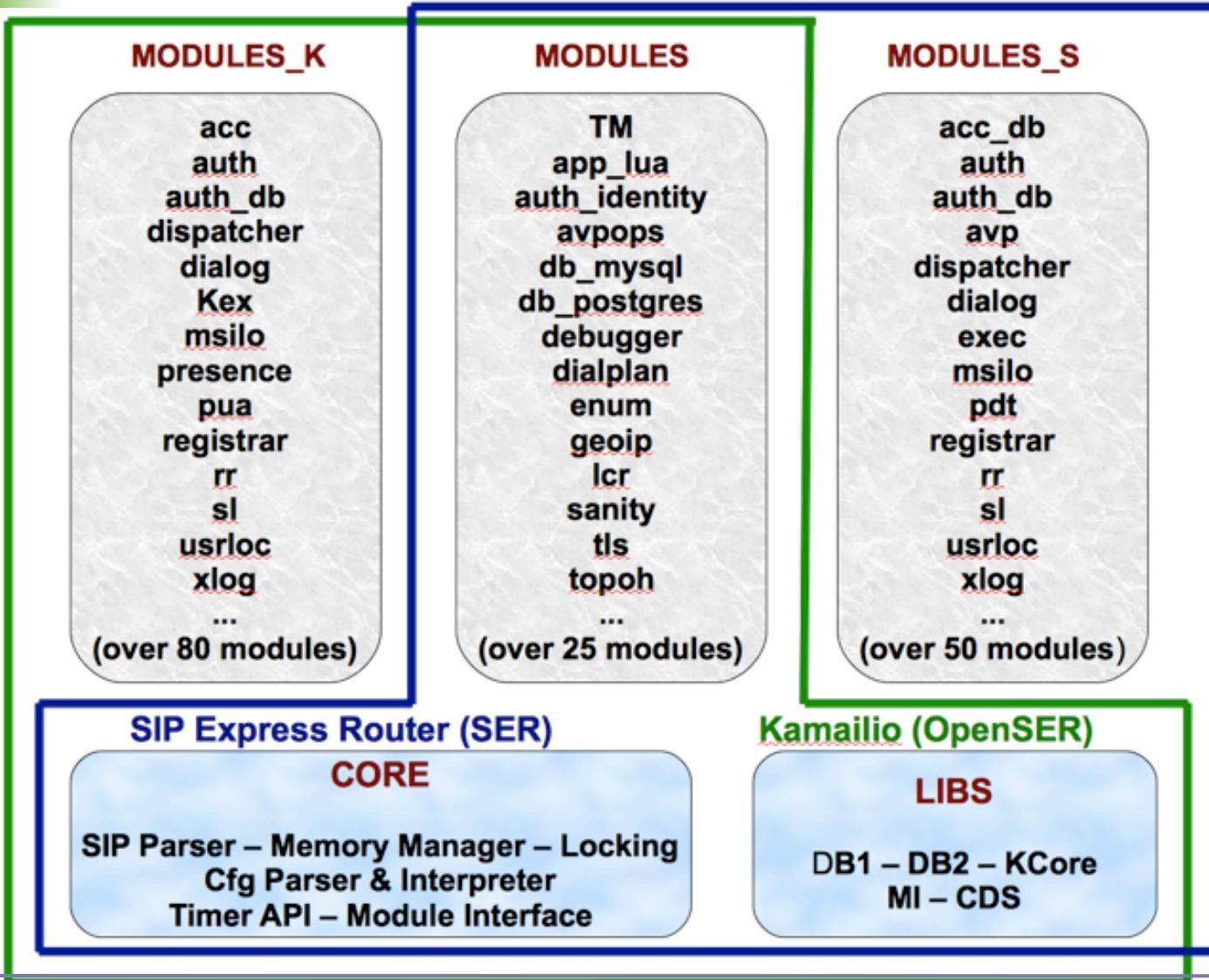


**New in 3.0.0**

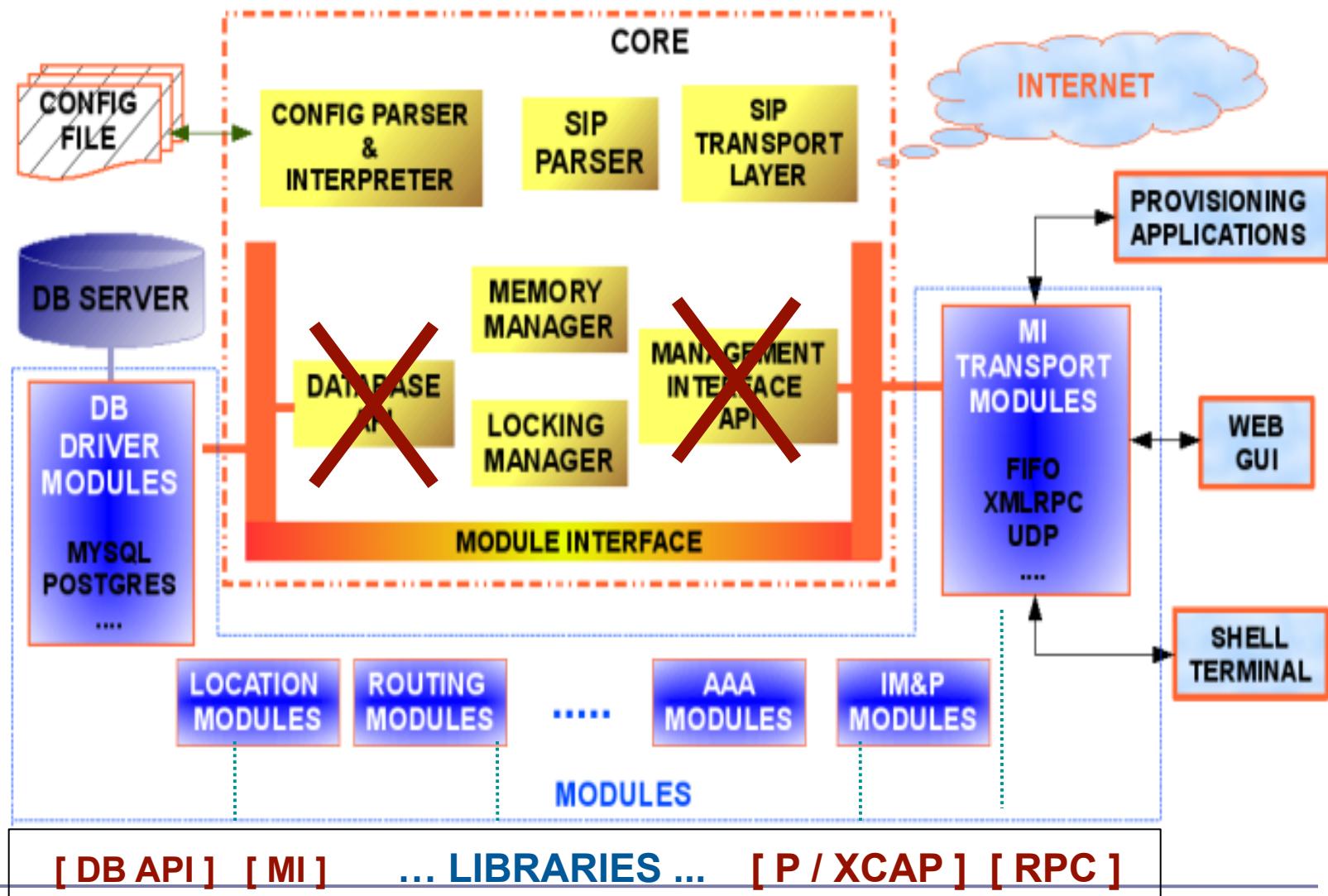
*Released January 11, 2010*



# 3.0 Releases



# Improving architecture



# New in 3.0.0

## Asynchronous TCP

several ten thousands  
of TCP connections

## Number Portability

customizable policies  
caching system

## SCTP

multi-homing  
multi-streaming  
statistics

## Topology hiding

fast and secure  
no session dependency

## Web & CLI Tools

SIREMIS  
SERCMD

# New in 3.0.0

## DNS Caching System

**fast failover  
blacklisting  
avoid blocking**

## CFG Reload Framework

**update global parameters  
at runtime**

## XMLRPC

**scalable control  
interface  
secure**

## Memcached connector

**data distribution across  
many instances**

## CFG Optimizations

**operations - operators  
switch - break - while**

# New in 3.0.0

## CFG Directives

include  
define

## New routing blocks

onsend route  
event route

## Transport layer

UDP MTU fallback  
scalable TLS

## Auth Identity

RFC4474

## Async SIP Message Processing

suspend - process - resume

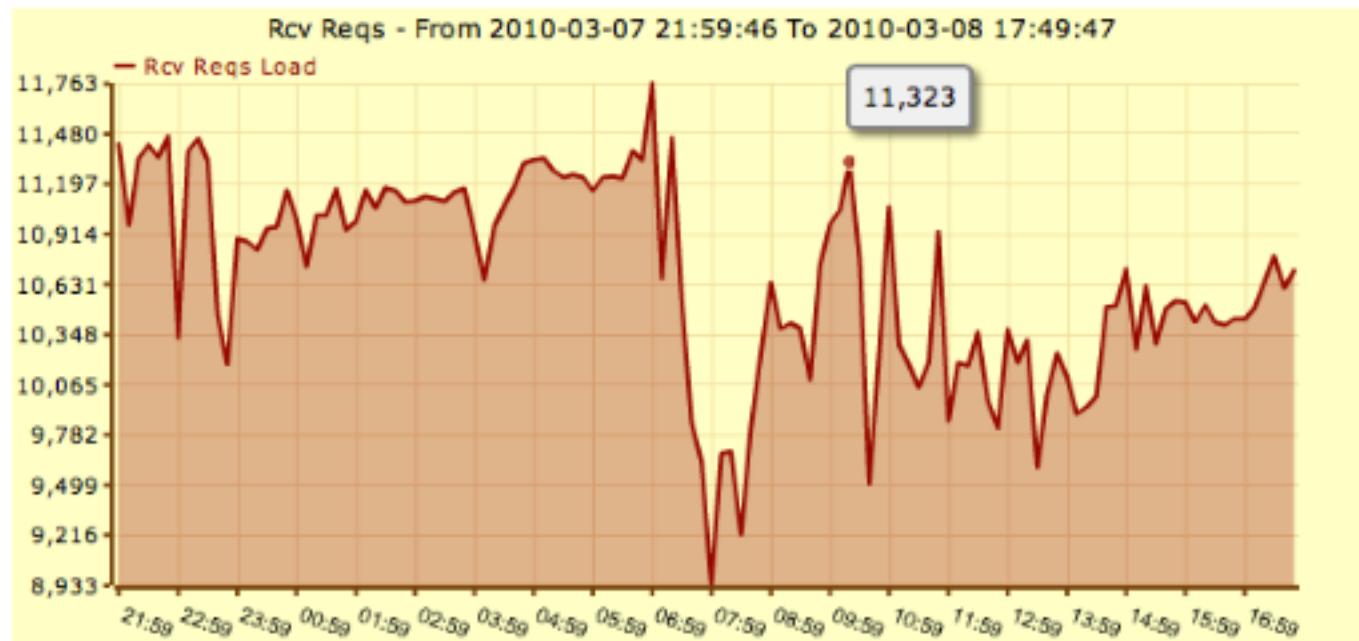
**Accounting Table**

Time	SIP Method	Src Username	Src Domain	Dst Username	Dst Domain	SIP Call-ID	SIP Reply Code
2009-01-21 14:47:36	INVITE	bob	asipto.com	alice	192.168.1.23	684598749@192.168.1.5	200
2009-01-21 14:55:41	INVITE	alice	asipto.com	bob	192.168.1.5	1142159929@192.168.1.23	200
2009-01-21 14:47:56	BYE	alice	asipto.com	bob	192.168.1.5	684598749@192.168.1.5	200
2009-01-21 14:55:41	ACK	alice	asipto.com	bob	192.168.1.5	1142159929@192.168.1.23	200
2009-01-21 14:56:03	BYE	alice	asipto.com	bob	192.168.1.5	1142159929@192.168.1.23	200
2009-01-21 15:29:38	INVITE	alice	asipto.com	bob	192.168.1.5	1553149770@192.168.1.23	200
2009-01-21 15:29:38	ACK	alice	asipto.com	bob	192.168.1.5	1553149770@192.168.1.23	200
2009-01-21 15:31:40	BYE	bob	asipto.com	alice	192.168.1.23	1553149770@192.168.1.23	200

**CDRS**

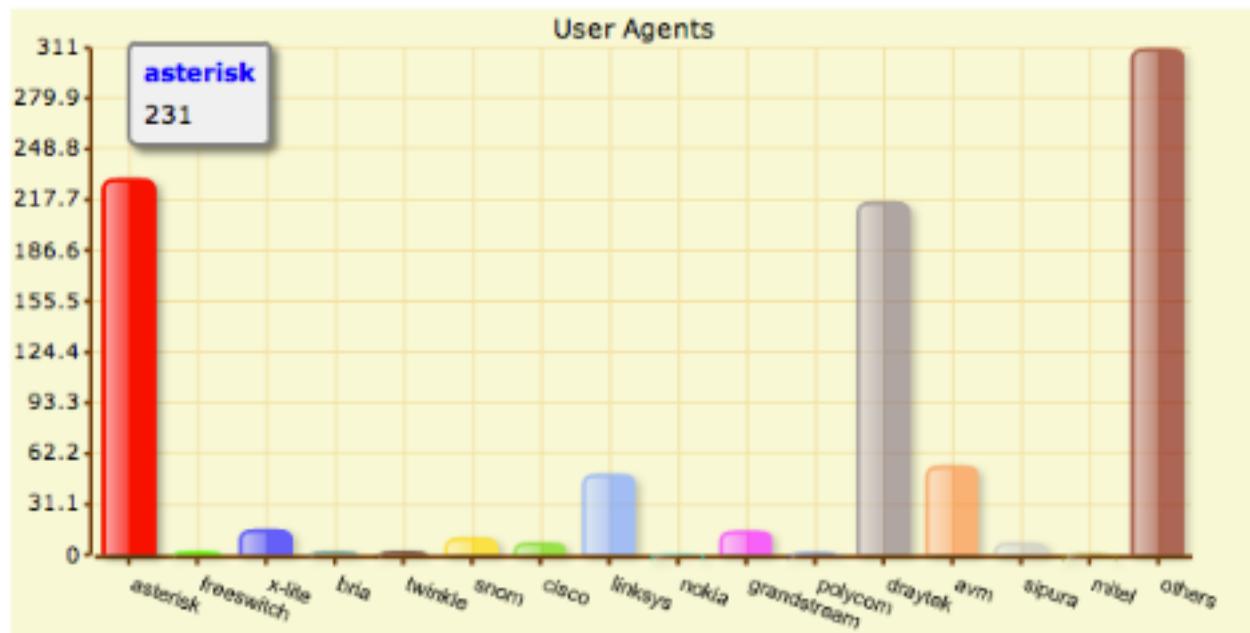
Id	Src Username	Src Domain	Dst Username	Dst Domain	Call Start Time	Duration
2	alice	asipto.com	bob	192.168.1.5	2009-01-21 14:55:41	22

RTS TM Charts Load Charts UL Stats



TM Charts Load Charts **UL Stats**

Processed 935 records.



# SERCMD

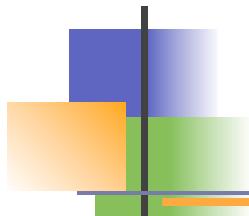
```
Terminal — ssh — 80x25

?
cfg.commit
cfg.diff
cfg.get
cfg.help
cfg.list
cfg.rollback
cfg.set_delayed_int
cfg.set_delayed_string
cfg.set_now_int
cfg.set_now_string
core.arg
core.echo
core.kill
core.printi
core.prints
core.ps
core.pwd
core.sctp_info
core.sctp_options
core.shmmem
core.tcp_info
core.tcp_options
core.uptime
--More--         

ctl.who
dns.add_a
dns.add_aaaa
dns.add_srv
dns.debug
dns.debug_all
dns.delete_a
dns.delete_aaaa
dns.delete_all
dns.delete_cname
dns.delete_ebl
dns.delete_naptr
dns.delete_ptr
dns.delete_srv
dns.delete_txt
dns.lookup
dns.mem_info
dns.view
dns_debug
dns_debug_all
dns_mem_info
dst_blacklist.add
dst_blacklist.debug
dst_blacklist.delete_all

dst_blacklist_mem_info
exit
help
license
list
listen
ls
mi
mi_dg
mi_fifo
mi_xmlrpc
ps
quit
server
serverversion
system.listMethods
system.methodHelp
system.methodSignature
tm.cancel
tm.hash_stats
tm.reply
tm.stats
tm.t_uac_start
tm.t_uac_wait
```

```
Terminal — ssh — 80x25
sercmd> core.tcp_options
{
    connect_timeout: 10
    send_timeout: 10
    connection_lifetime: 120
    max_connections(soft): 2048
    no_connect: 0
    fd_cache: 1
    async: 1
    connect_wait: 1
    conn_wq_max: 32768
    wq_max: 10485760
    defer_accept: 0
    delayed_ack: 1
    syncnt: 0
    linger2: 0
    keepalive: 1
    keepidle: 0
    keepintvl: 0
    keepcnt: 0
    crlf_ping: 1
    accept_aliases: 0
    alias_flags: 1
    new_conn_alias_flags: 2
}
```



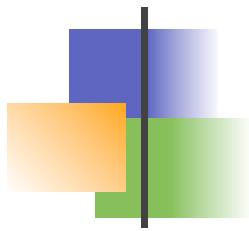
# New in 3.0.0

---

<http://www.kamailio.org/dokuwiki/doku.php/features:new-in-3.0.x>

<http://www.kamailio.org/w/kamailio-openser-v3.0.0-release-notes/>

<http://by-miconda.blogspot.com/2010/01/best-of-new-in-kamailio-300-toc.html>



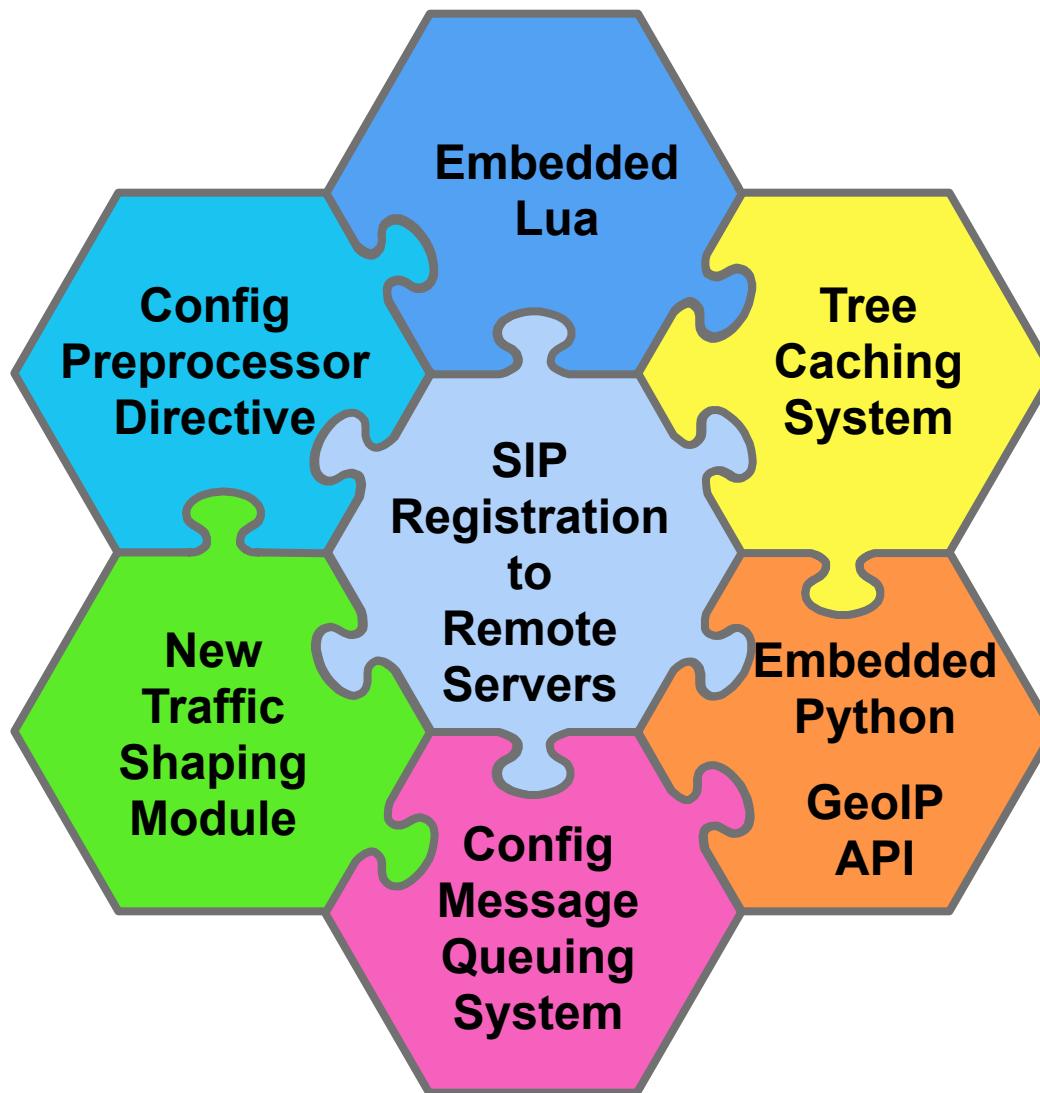
## New in Development

3.1.0

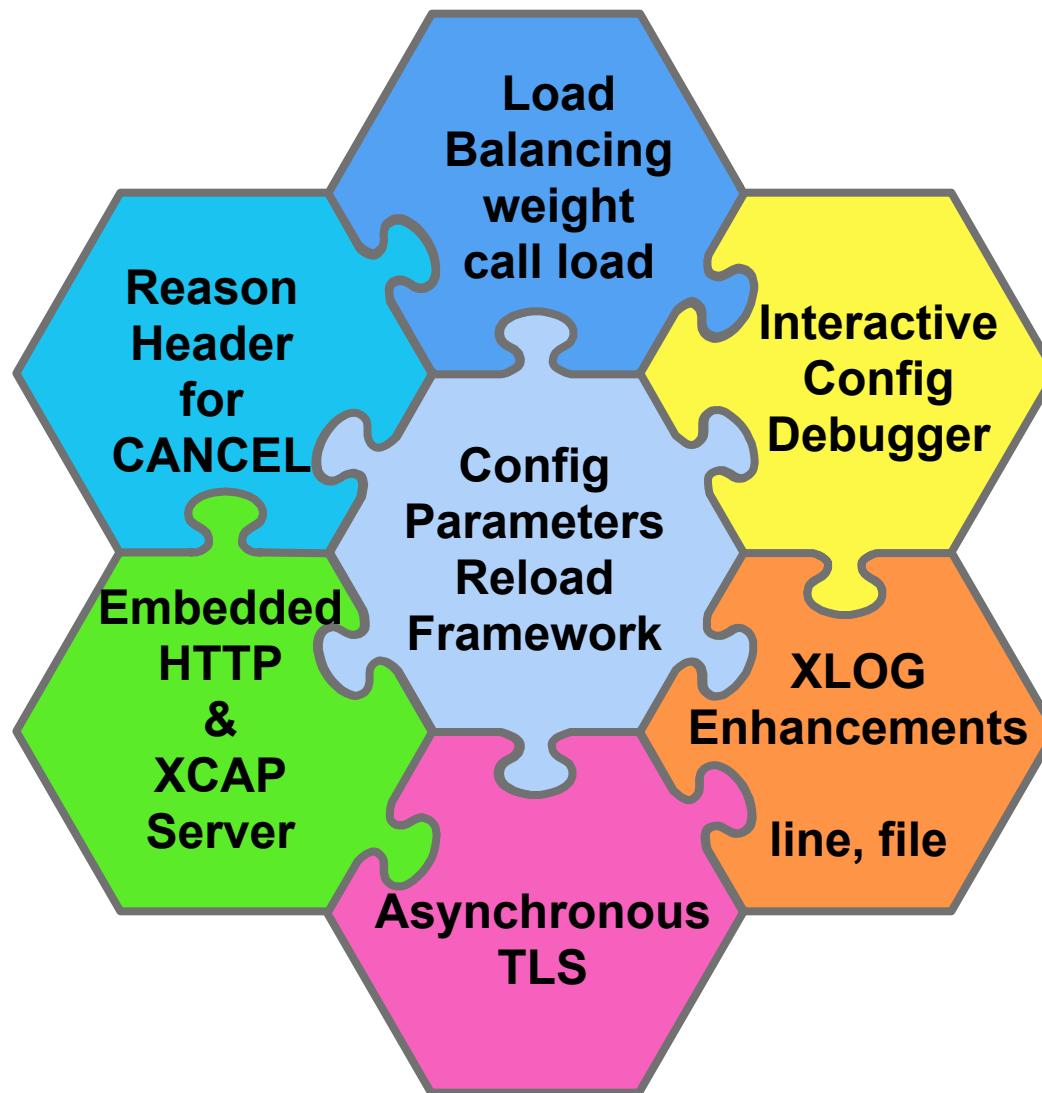
*Autumn 2010*

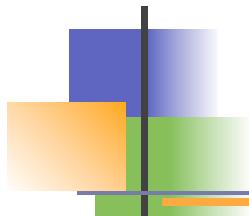
---

# New in devel (upcoming 3.1.0)



# New in devel (upcoming 3.1.0)



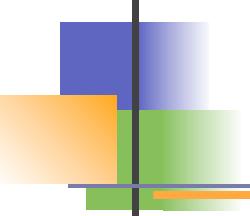


## New in devel (upcoming 3.1.0)

---

<http://sip-router.org/wiki/features/new-in-devel>

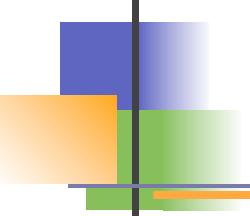
# Asynchronous SIP Routing



# Asynchronous TCP

---

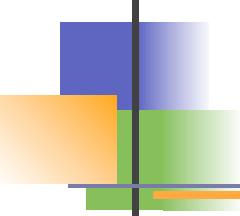
- present in 3.0.x (latest stable)
  - Issues solved
    - blocking the SIP worker process
    - opening a connection
    - sending data
  - Configuration as easy as:
    - `tcp_async = yes`
  - Benefits
    - increased processing capacity over TCP
    - better handling of traffic for IM and Presence
    - transparent for modules and config file functions
    - integrated with DNS blacklisting



# Asynchronous TLS

---

- Asynchronous TCP and TLS for SIP Routing
  - **worldwide - you can get them only with Kamailio and SER**
- in code tree GIT repository (ready for upcoming 3.1.0)
  - Same benefits as for asynchronous TCP
    - asynchronous TCP implies asynchronous TLS
    - same config parameter: `tcp_async=yes`
  - Extra
    - increased capacity of handling secure communication

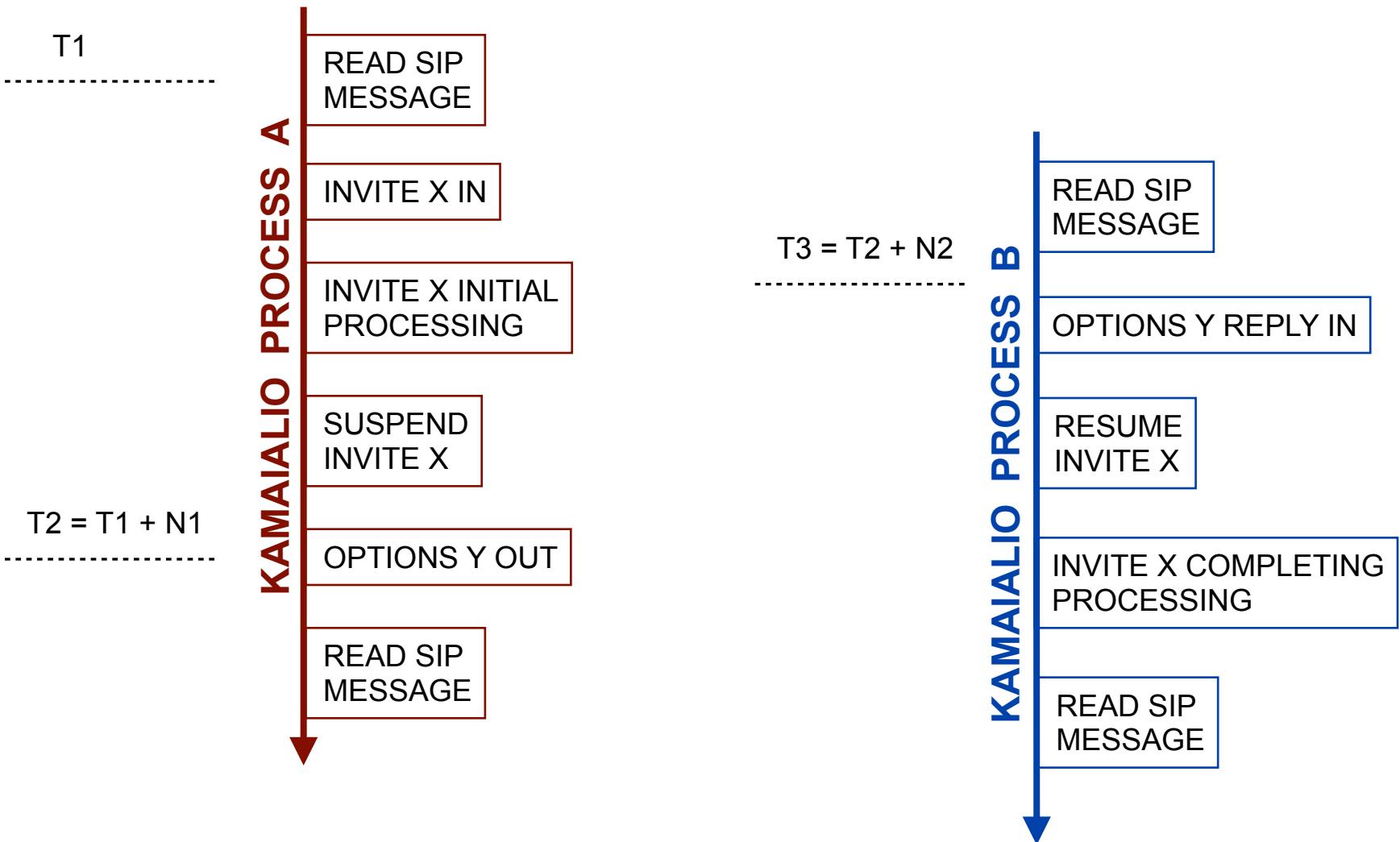


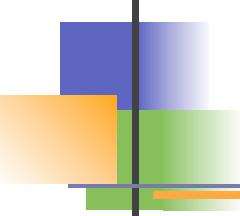
# Asynchronous SIP Message Processing

---

- present in 3.0.x (latest stable)
  - default message processing
    - receive - process - forward/reply
  - new possibility of message processing
    - **receive - process - suspend - do something else - resume - forward/reply**
  - requirements
    - transaction management module (tm)
  - example
    - call comes in, authenticate user, then send a request to billing engine to authorize the call
    - suspend handling the invite (park it), process other SIP requests
    - when billing engine replies, resume INVITE processing and allow or deny the call

# Asynchronous SIP Message Processing





# Config Message Queuing System

---

- in code tree GIT repository (ready for upcoming 3.1.0)
  - New module: mqueue
    - multiple message queues
    - message: (key, body)
  - Features
    - inter-process communication
    - SIP worker process can send commands to timer worker
  - Examples
    - **SIPWEET**
      - send notifications to Twitter (see my other talk **@18:00**)
      - email on various events



Questions?

**Contact:**

[daniel@asipto.com](mailto:daniel@asipto.com)

<http://www.asipto.com>  
<http://www.kamailio.org>  
<http://sip-router.org>