



Asterisk 12 and PJSIP

Asterisk's PJSIP channel driver: a SIP architecture
for the future

The future is now!

- Why write a new SIP stack?
- RFC 3261 – SIP: Session Initiation Protocol
 - June 2002
- chan_sip:
 - r472 | markster | 2002-06-28 15:34:46 -0500 (Fri, 28 Jun 2002) | 2 lines
 - Version 0.1.12 from FTP
- That's 12 years ago!

No Facebook, Twitter, or even MySpace

- Social network: friendster
 - This is now a gaming site?



<http://mediafactory.org.au>

<http://www.friendster.com/>

Linux landscape was changing

- RedHat releases first version of RHEL
 - May 6 2002: RHEL 2.1 AS (Pensacola)



https://fedoraproject.org/wiki/History_of_Red_Hat_Linux
<http://redhat.com>

We still cared about Blackberry

- Blackberry 5810 (March 4, 2002)



http://www.techhive.com/article/172837/the_mobile_phone_a_history_in_pictures.html

- Two asterisk-users mailing list emails still exist
- One is unsubscribe

January 2002 Archives by thread

- **Messages sorted by:** [\[subject \]](#) [\[author \]](#) [\[date \]](#)
- [More info on this list...](#)

Starting: *Wed Jan 16 01:02:31 MST 2002*

Ending: *Wed Jan 16 01:02:31 MST 2002*

Messages: 1

- [\[Asterisk-Users\] unsubscribe](#) *jave*

Last message date: *Wed Jan 16 01:02:31 MST 2002*

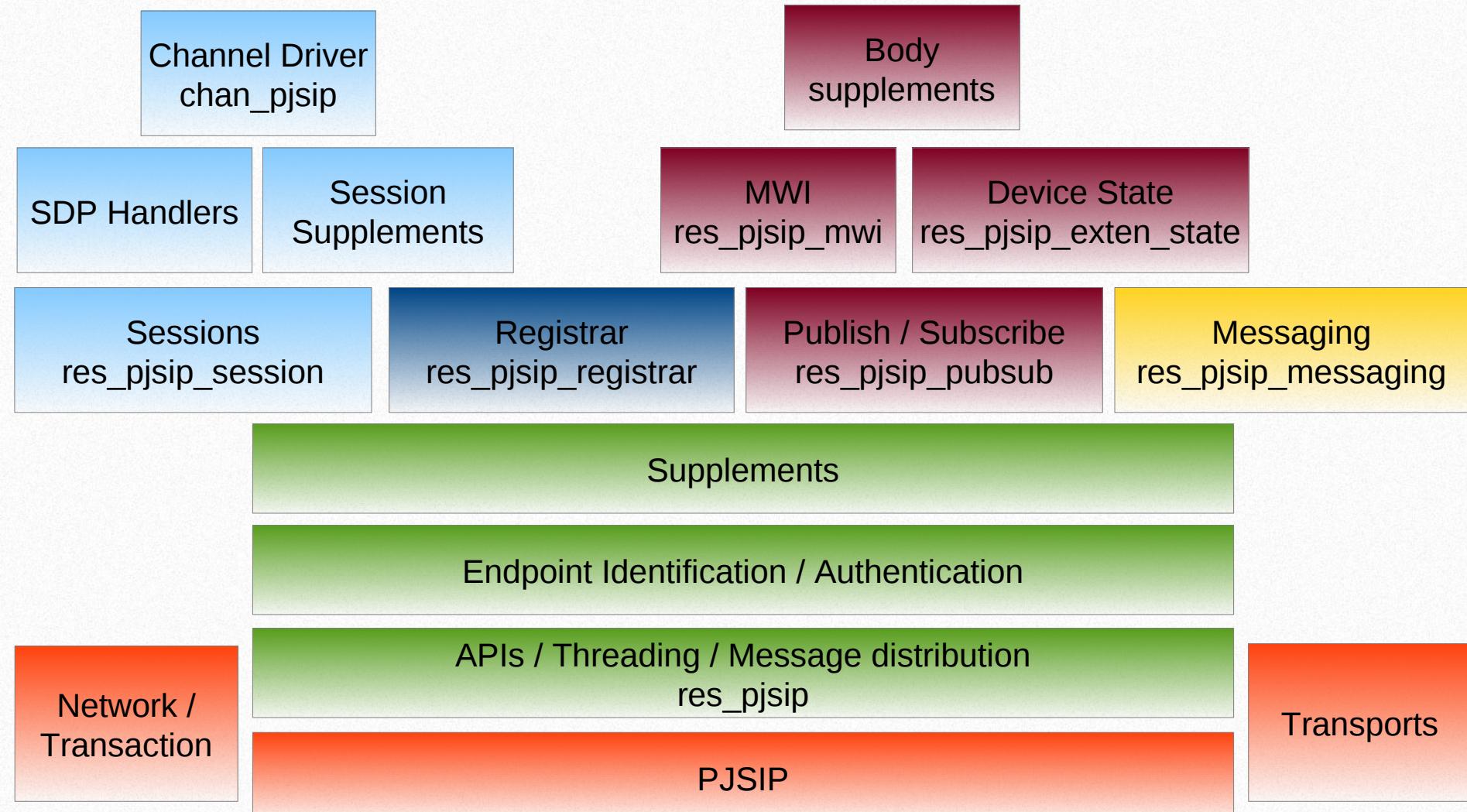
Archived on: *Tue Sep 5 15:25:22 MST 2006*

- **Messages sorted by:** [\[subject \]](#) [\[author \]](#) [\[date \]](#)
- [More info on this list...](#)

This archive was generated by Pipermail 0.09 (Mailman edition).

- Architecture was never designed for its current size
 - 0.1.12 – 1950 lines
 - Trunk – 34570 lines
- Current structure limits change
 - No stack
 - Large monolithic architecture
- Venerable, yet time to retire

Asterisk 12 SIP Stack



Example: Inbound INVITE request

Application

Dialog

UA/Proxy Layer

Transaction Layer

PJSIP

Example: Inbound INVITE request

Application

Dialog

UA/Proxy Layer

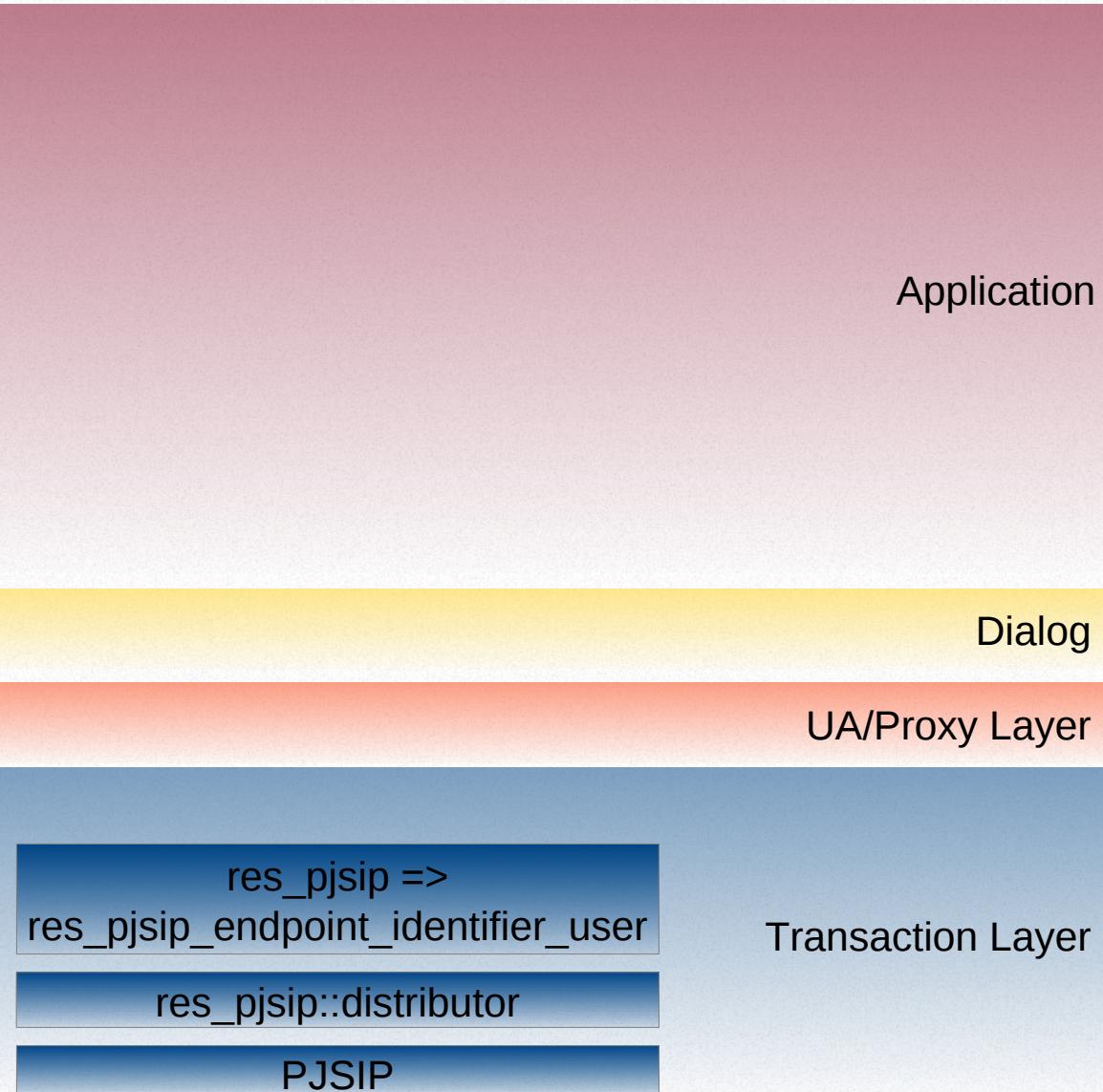
Transaction Layer

res_pjsip::distributor

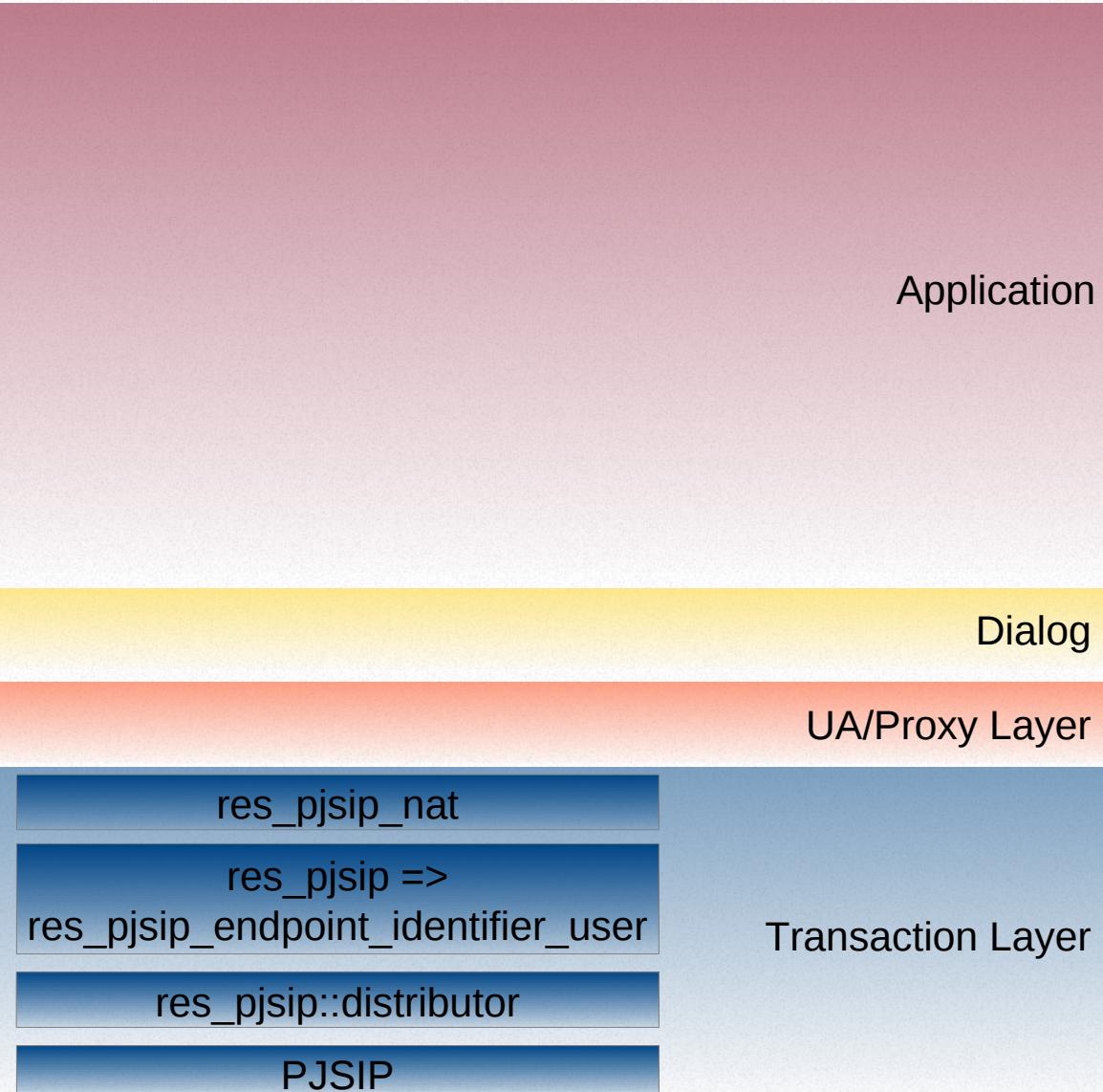
PJSIP

Put the request in a
threadpool for processing

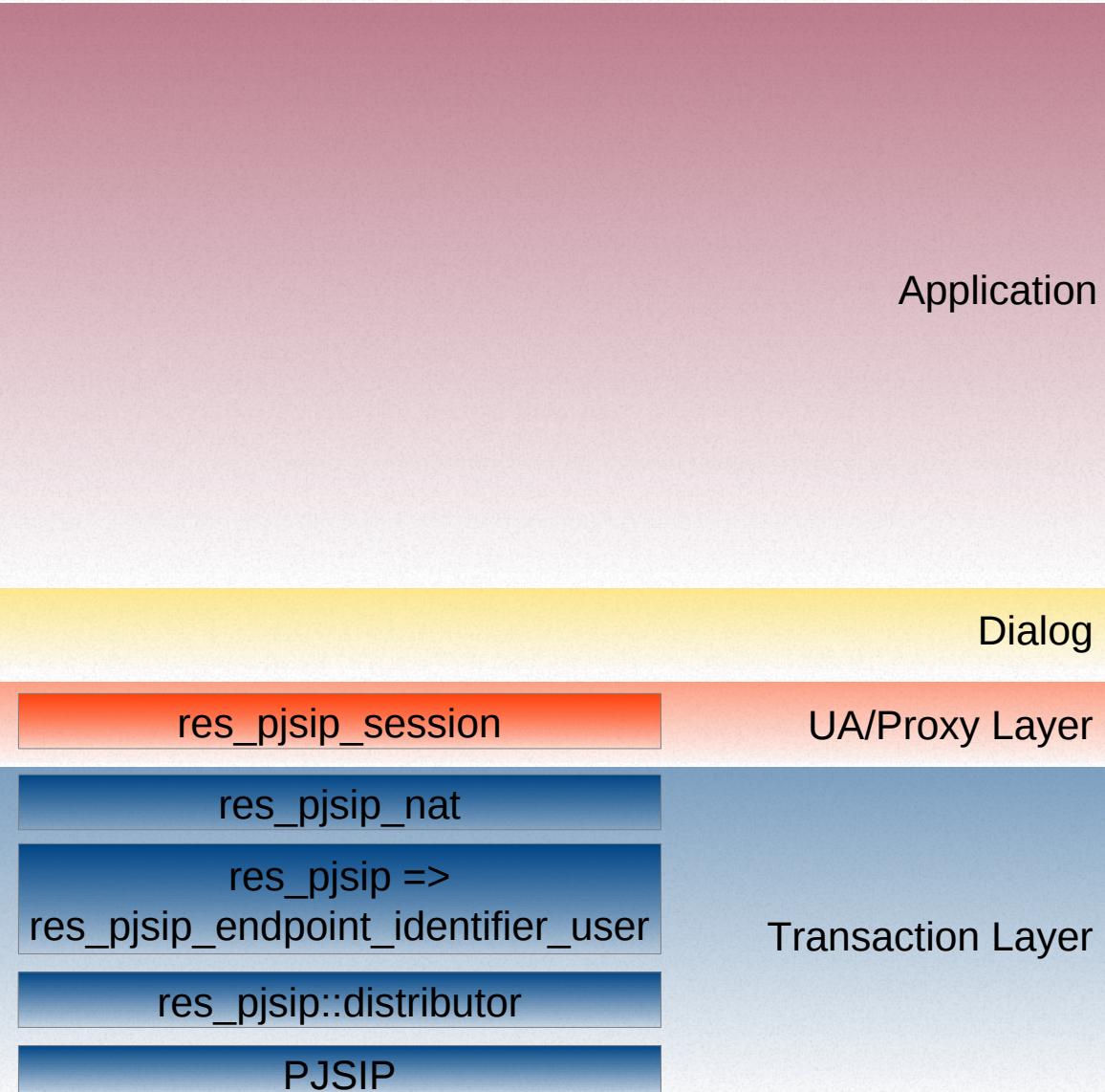
Example: Inbound INVITE request



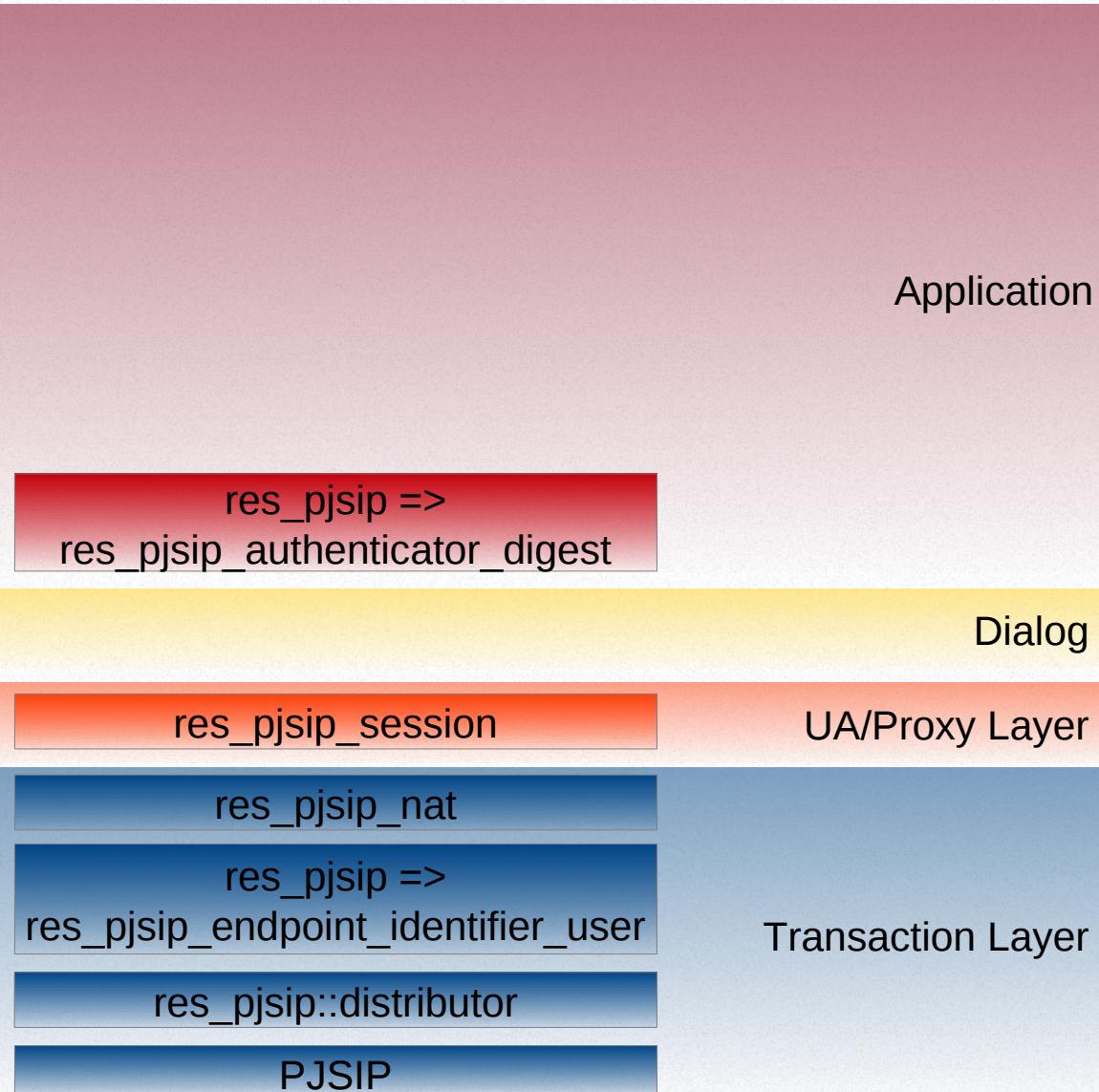
Example: Inbound INVITE request



Example: Inbound INVITE request



Example: Inbound INVITE request



Example: Inbound INVITE request

res_pjsip_session

res_pjsip =>

res_pjsip_authenticator_digest

Application

Make a new session

Dialog

res_pjsip_session

UA/Proxy Layer

res_pjsip_nat

res_pjsip =>

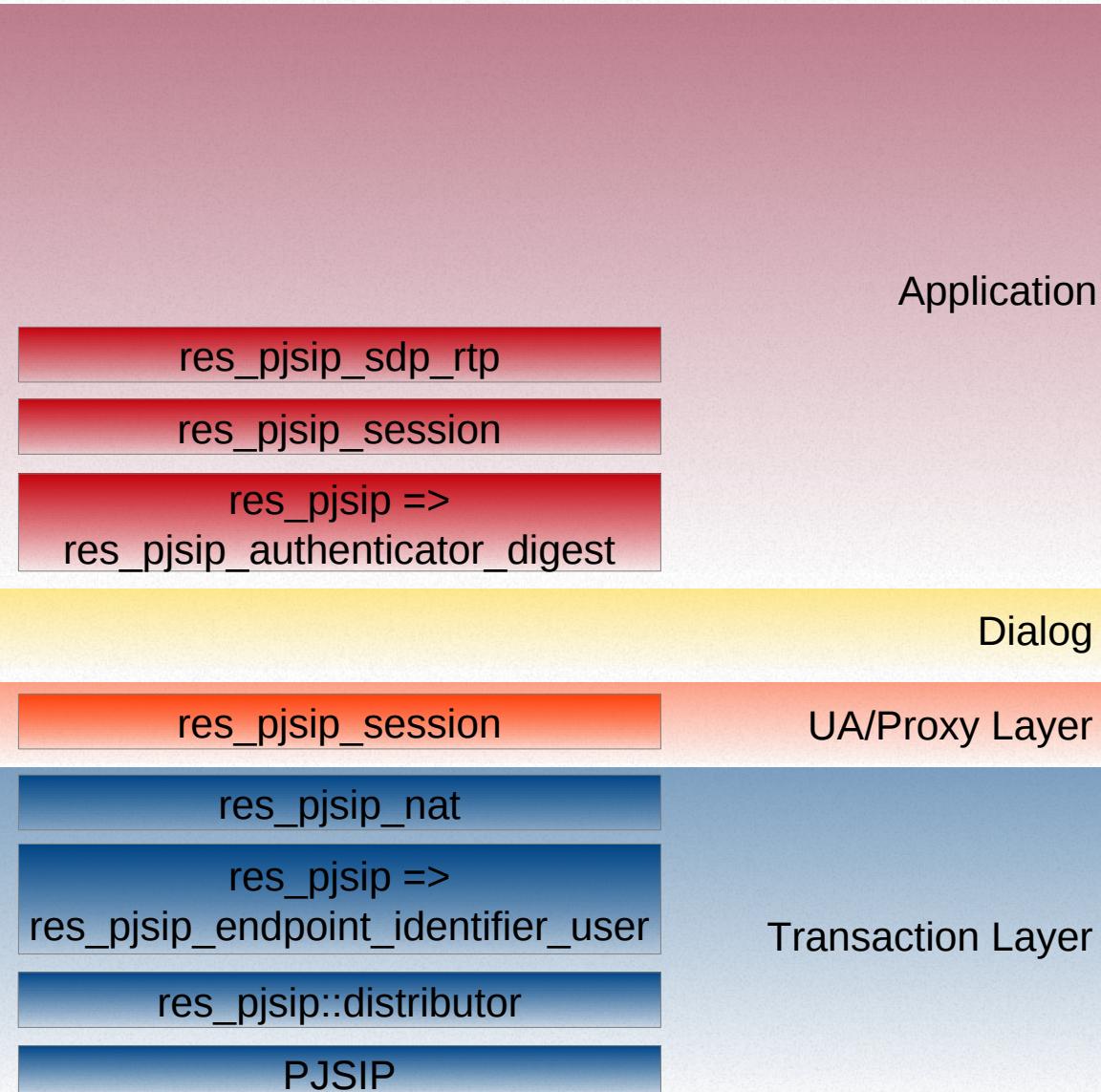
res_pjsip_endpoint_identifier_user

res_pjsip::distributor

Transaction Layer

PJSIP

Example: Inbound INVITE request



Example: Inbound INVITE request

`res_pjsip_caller_id`

Application

`res_pjsip_sdp_rtp`

Extract caller ID and store it

`res_pjsip_session`

`res_pjsip =>`
`res_pjsip_authenticator_digest`

Dialog

`res_pjsip_session`

UA/Proxy Layer

`res_pjsip_nat`

`res_pjsip =>`

`res_pjsip_endpoint_identifier_user`

`res_pjsip::distributor`

Transaction Layer

PJSIP

Example: Inbound INVITE request

chan_pjsip

res_pjsip_caller_id

res_pjsip_sdp_rtp

res_pjsip_session

res_pjsip =>
res_pjsip_authenticator_digest

Make the ast_channel object

Application

Dialog

res_pjsip_session

UA/Proxy Layer

res_pjsip_nat

res_pjsip =>
res_pjsip_endpoint_identifier_user

Transaction Layer

res_pjsip::distributor

PJSIP

Example: Inbound INVITE request

res_pjsip_t38

chan_pjsip

res_pjsip_caller_id

res_pjsip_sdp_rtp

res_pjsip_session

res_pjsip =>
res_pjsip_authenticator_digest

Application

Dialog

res_pjsip_session

UA/Proxy Layer

res_pjsip_nat

res_pjsip =>
res_pjsip_endpoint_identifier_user

Transaction Layer

res_pjsip::distributor

PJSIP

See if we need to do anything with T.38 fax state (nope!)

Example: Inbound INVITE request

chan_pjsip

Start the PBX!

res_pjsip_t38

chan_pjsip

res_pjsip_caller_id

Application

res_pjsip_sdp_rtp

res_pjsip_session

res_pjsip =>
res_pjsip_authenticator_digest

Dialog

res_pjsip_session

UA/Proxy Layer

res_pjsip_nat

res_pjsip =>

res_pjsip_endpoint_identifier_user

Transaction Layer

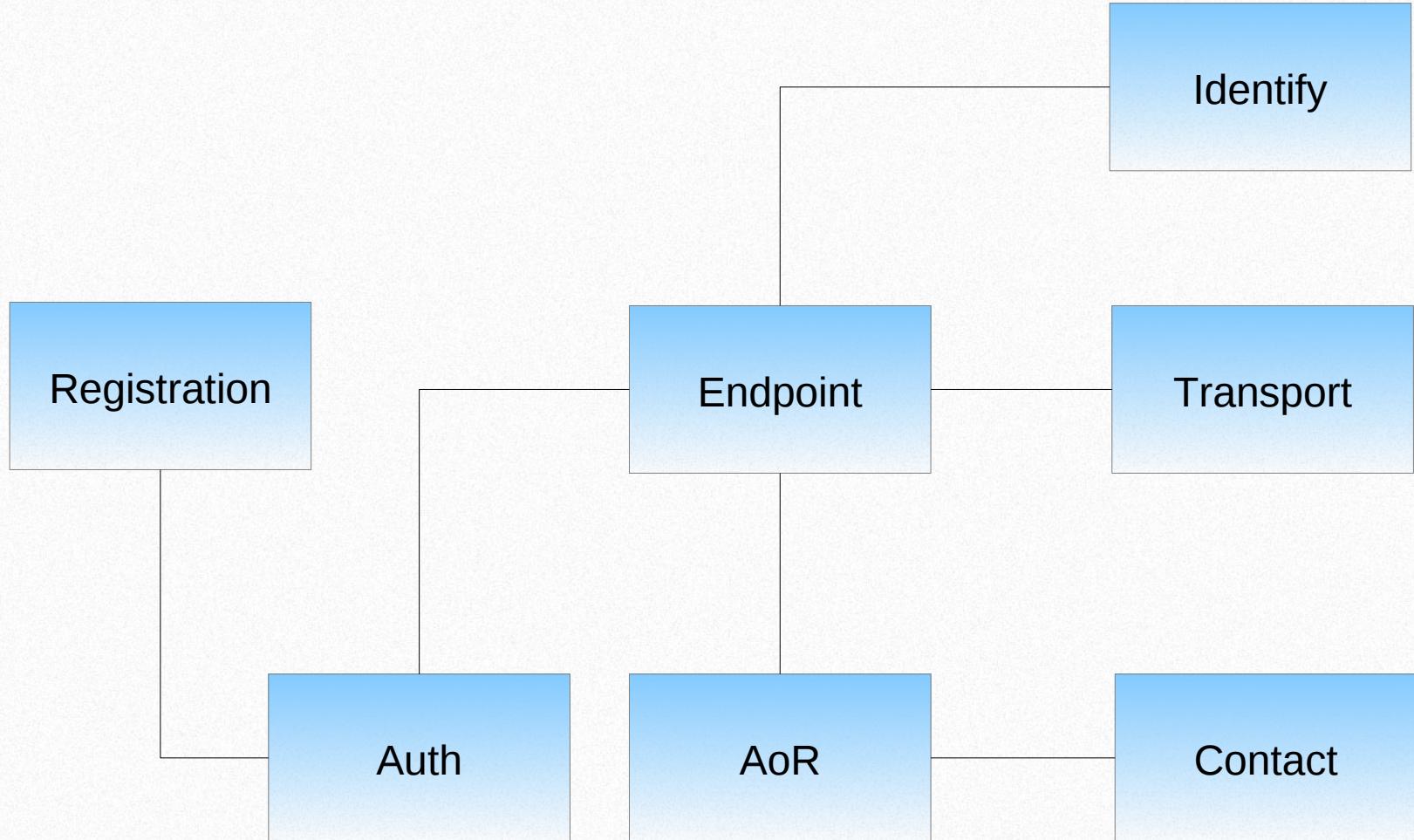
res_pjsip::distributor

PJSIP

Configuration

- Sorcery: Data Abstraction Layer
 - ORM (for some values of O and R)
 - Supports CRUD operations
 - Well defined lifetime, thread-safe, reload-safe
 - Prune realtime peers/reloads
- Smaller objects
 - Takes advantage of templating/databases
 - Can change storage location of each object
 - Example: Configuration data in database; contacts in AstDB
 - Simple mapping to in-memory representation
 - Defer higher level concepts to systems on top of Asterisk

Configuration Relationships



Configuration Example

```
[alice]
type=endpoint
context=internal
allow=!all,g722,alaw
auth=alice_auth
aors=alice_aors

[alice_auth]
type=auth
auth_type=userpass
username=alice
password=as8918hd!@8hs19a1m

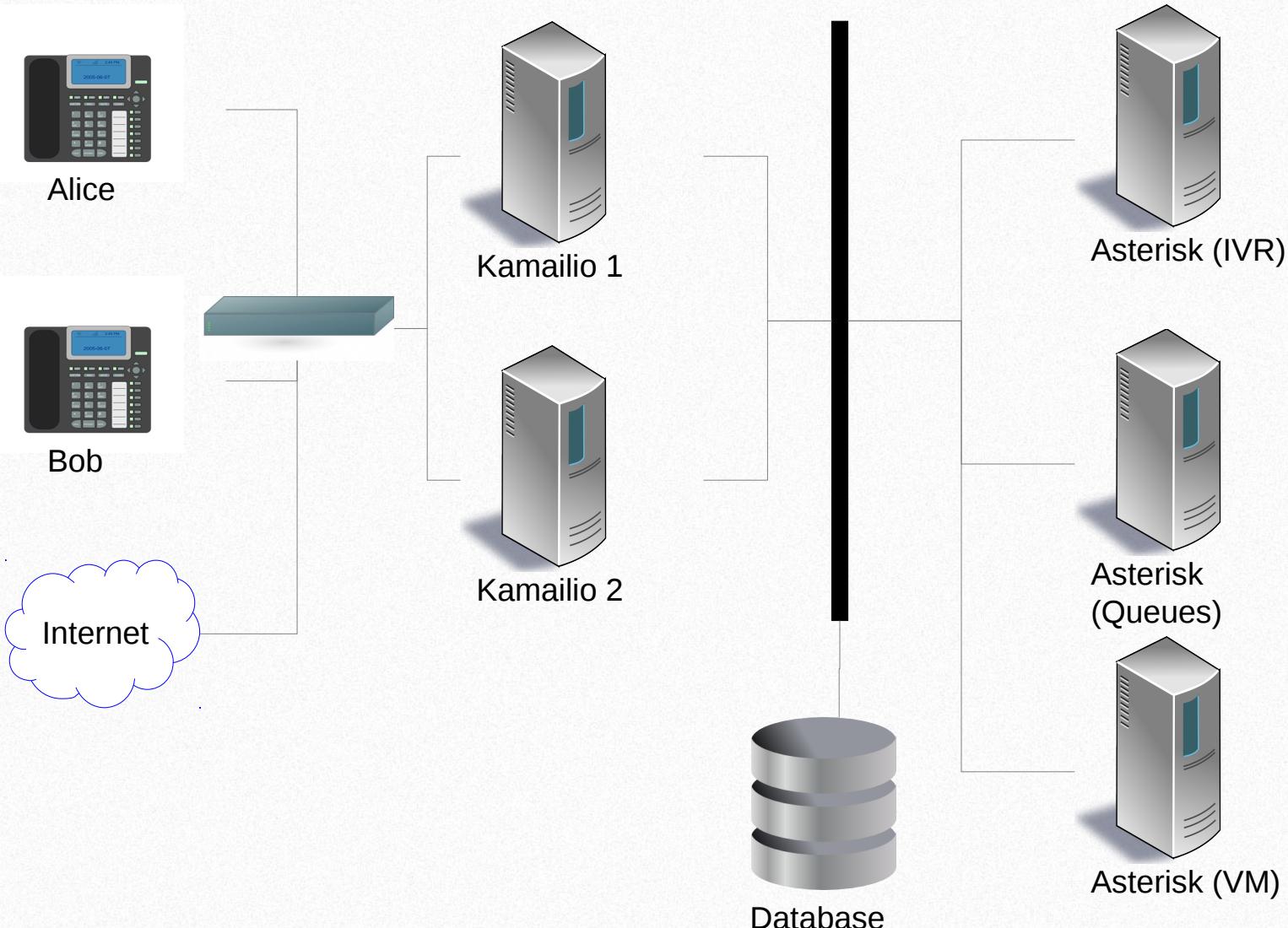
[alice_aors]
type=aor
max_contacts=10
```

Asterisk and Kamailio

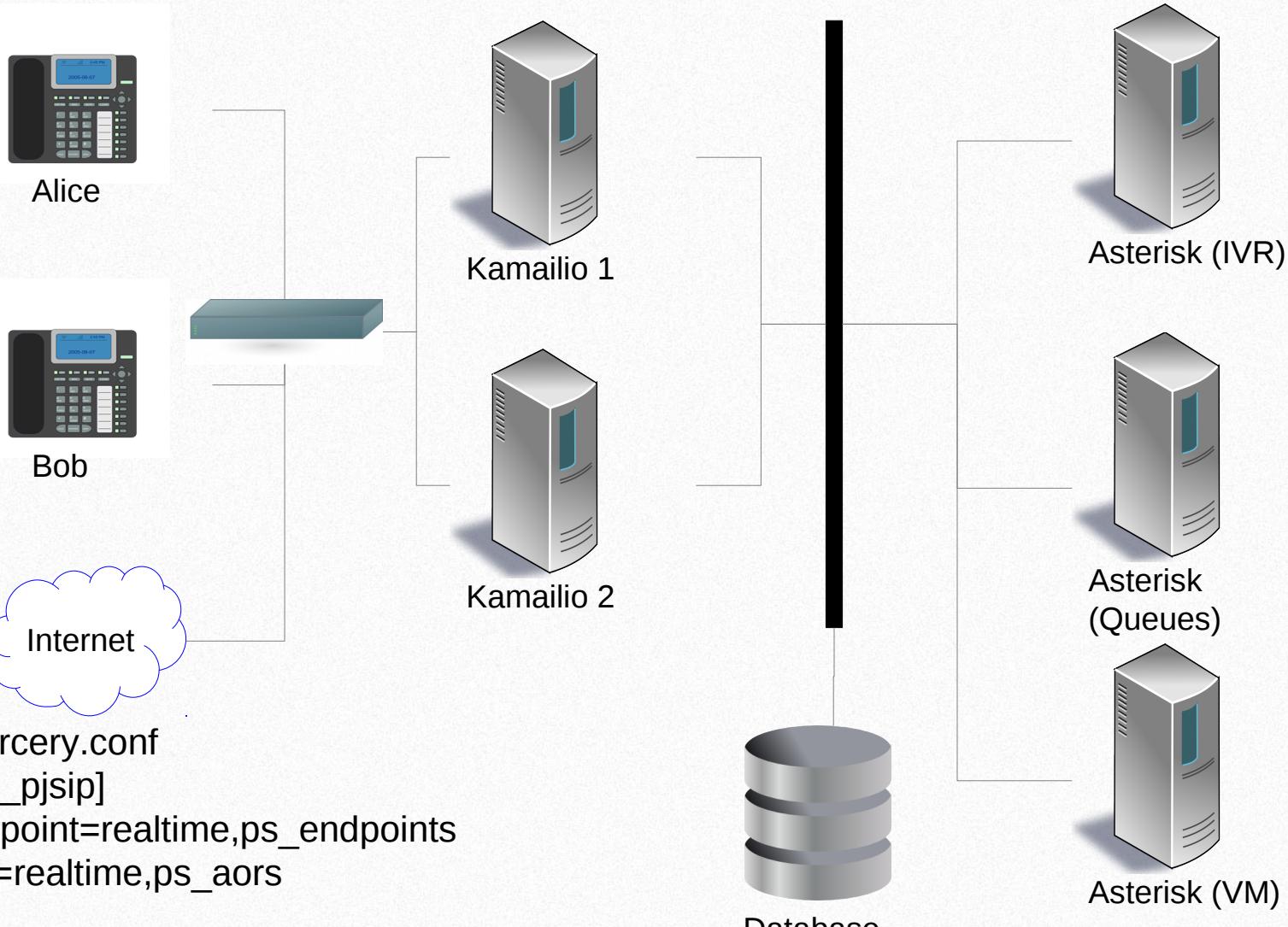


http://www.markthalle-in-hannover.de/p_73_bistro
<http://shop.brewforia.com/browse-by/style/german-pilsner>

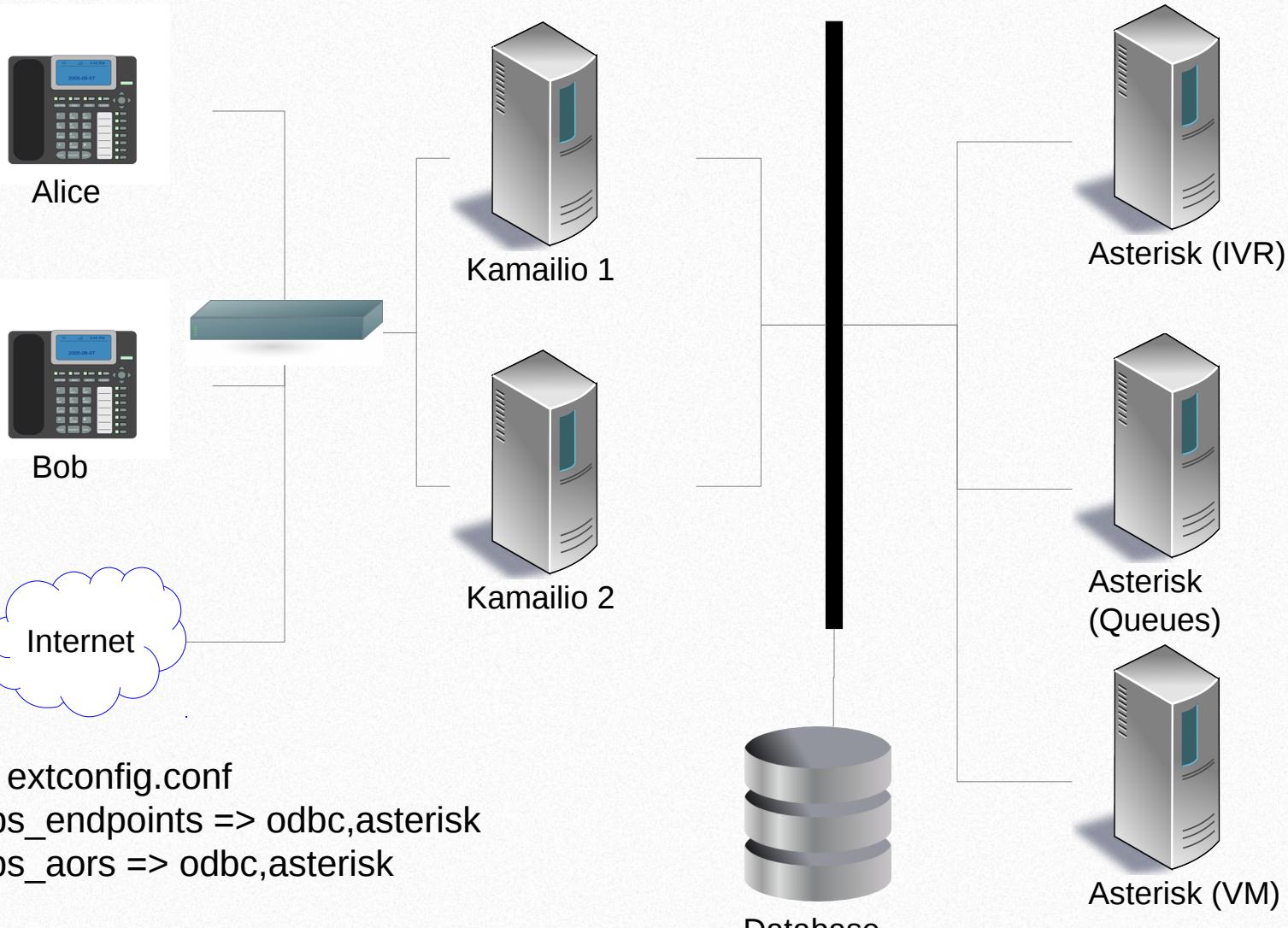
Asterisk 12 and Kamailio



Asterisk 12 and Kamailio

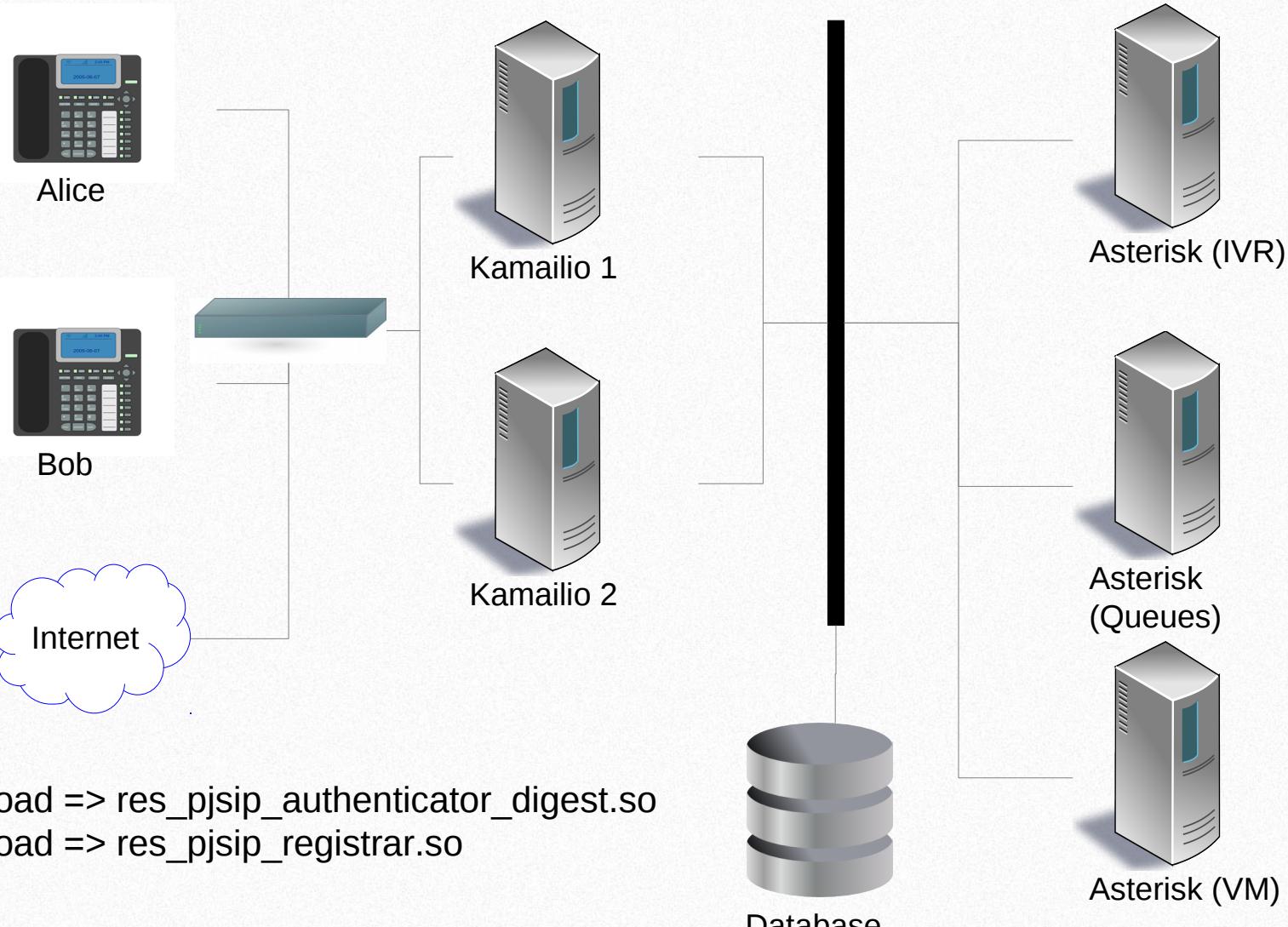


Asterisk 12 and Kamailio



```
; extconfig.conf
ps_endpoints => odbc,asterisk
ps_aors => odbc,asterisk
```

Asterisk 12 and Kamailio

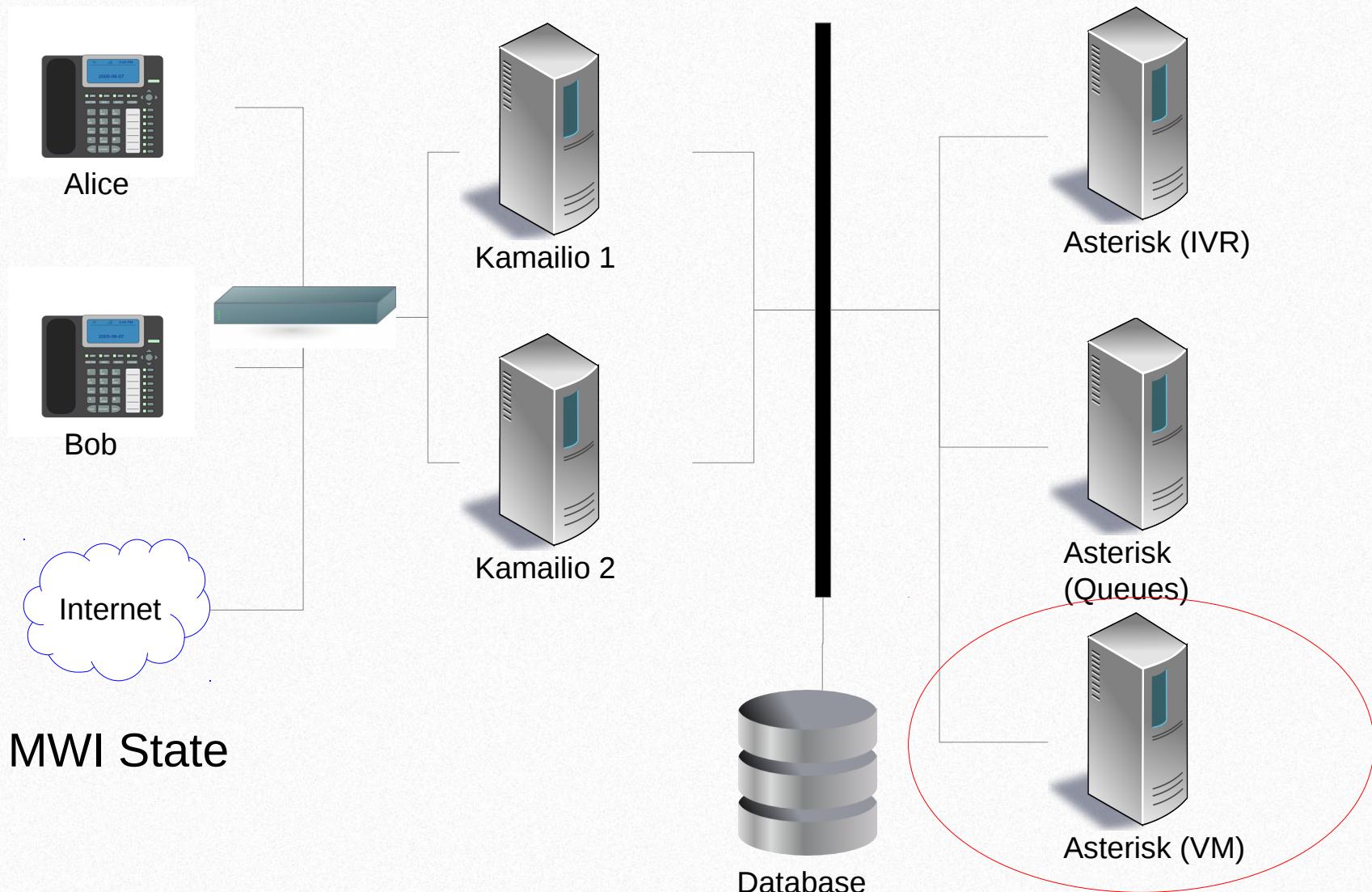


What's Next?

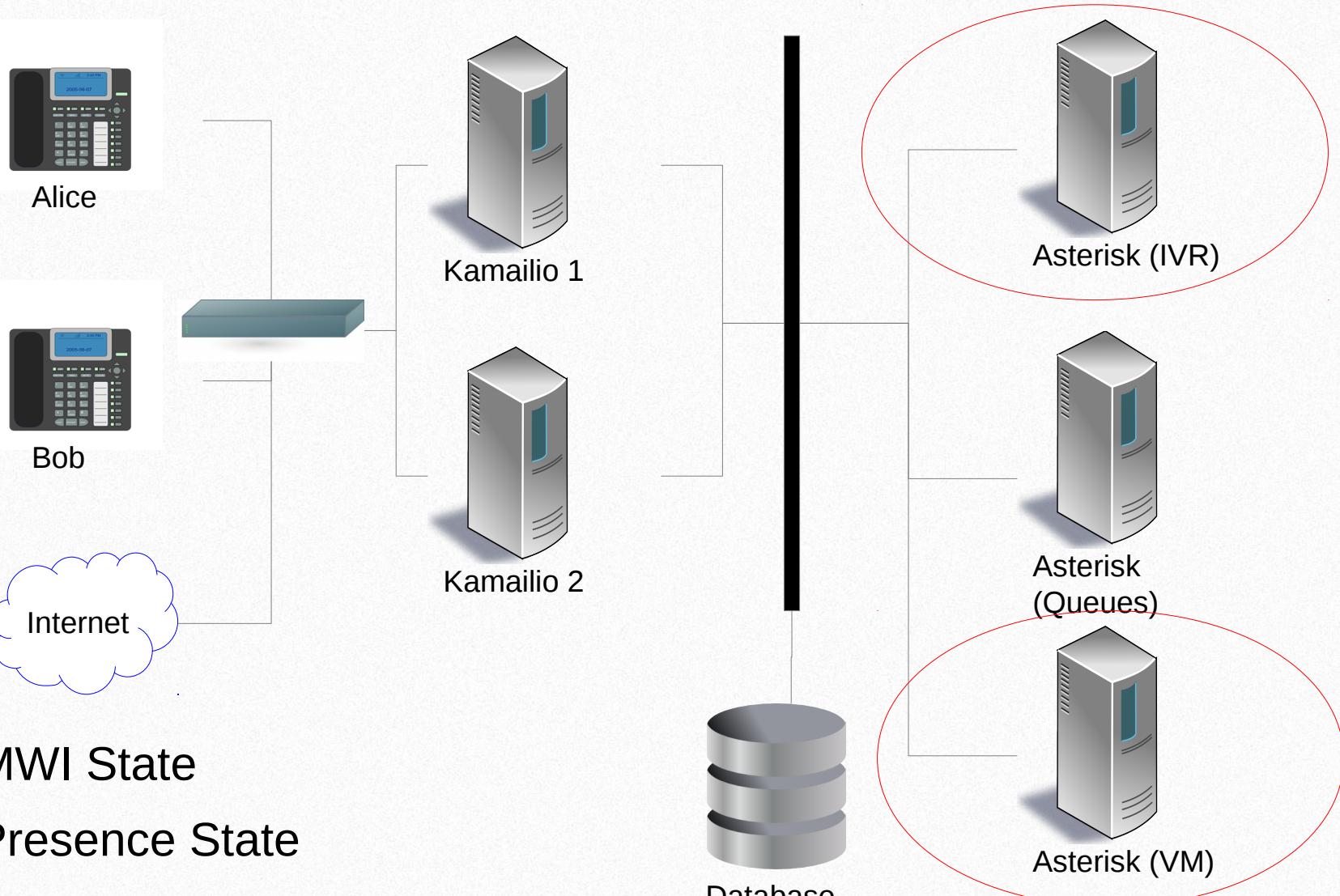


- Distributing Presence

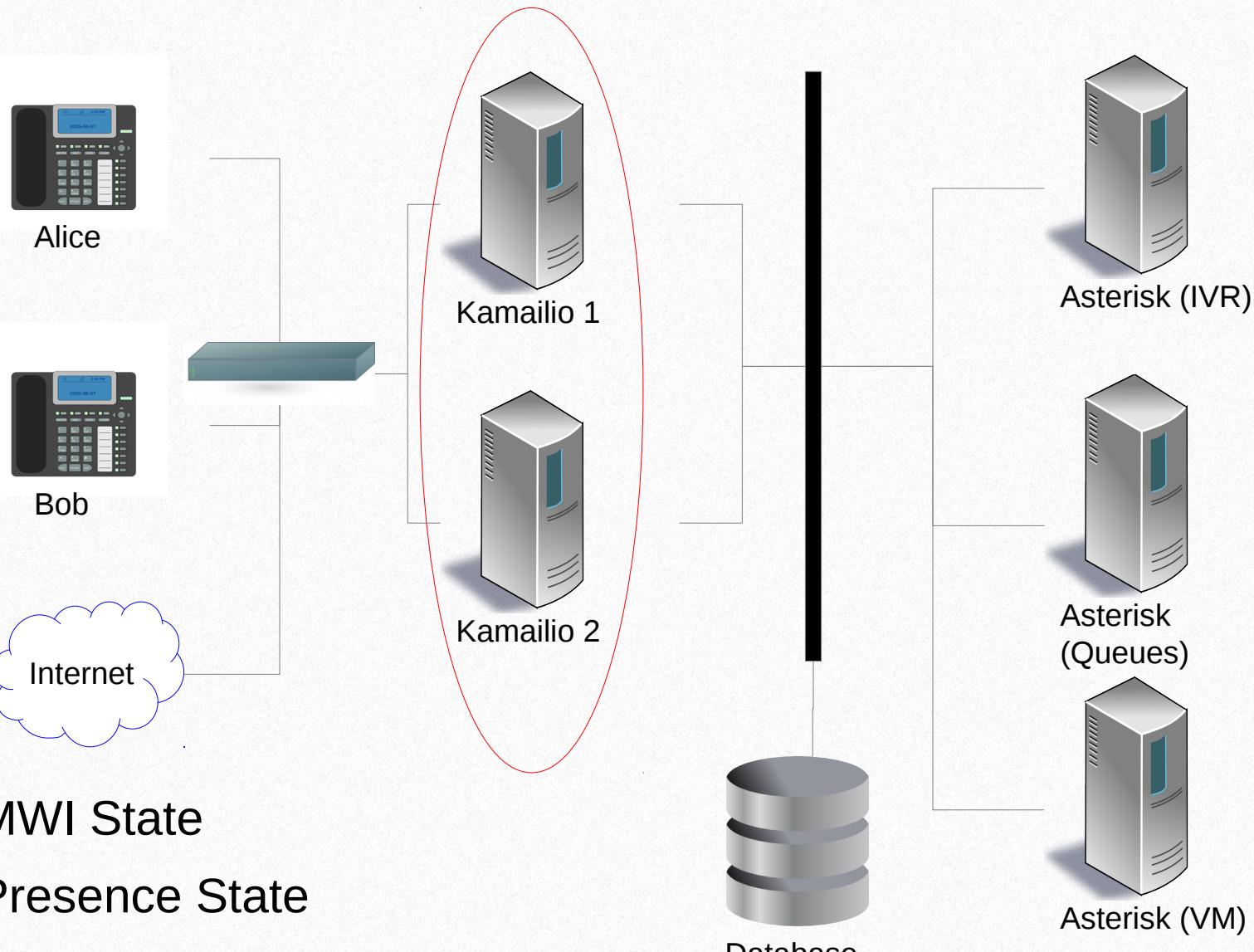
Asterisk 12 and Kamailio: Next Steps



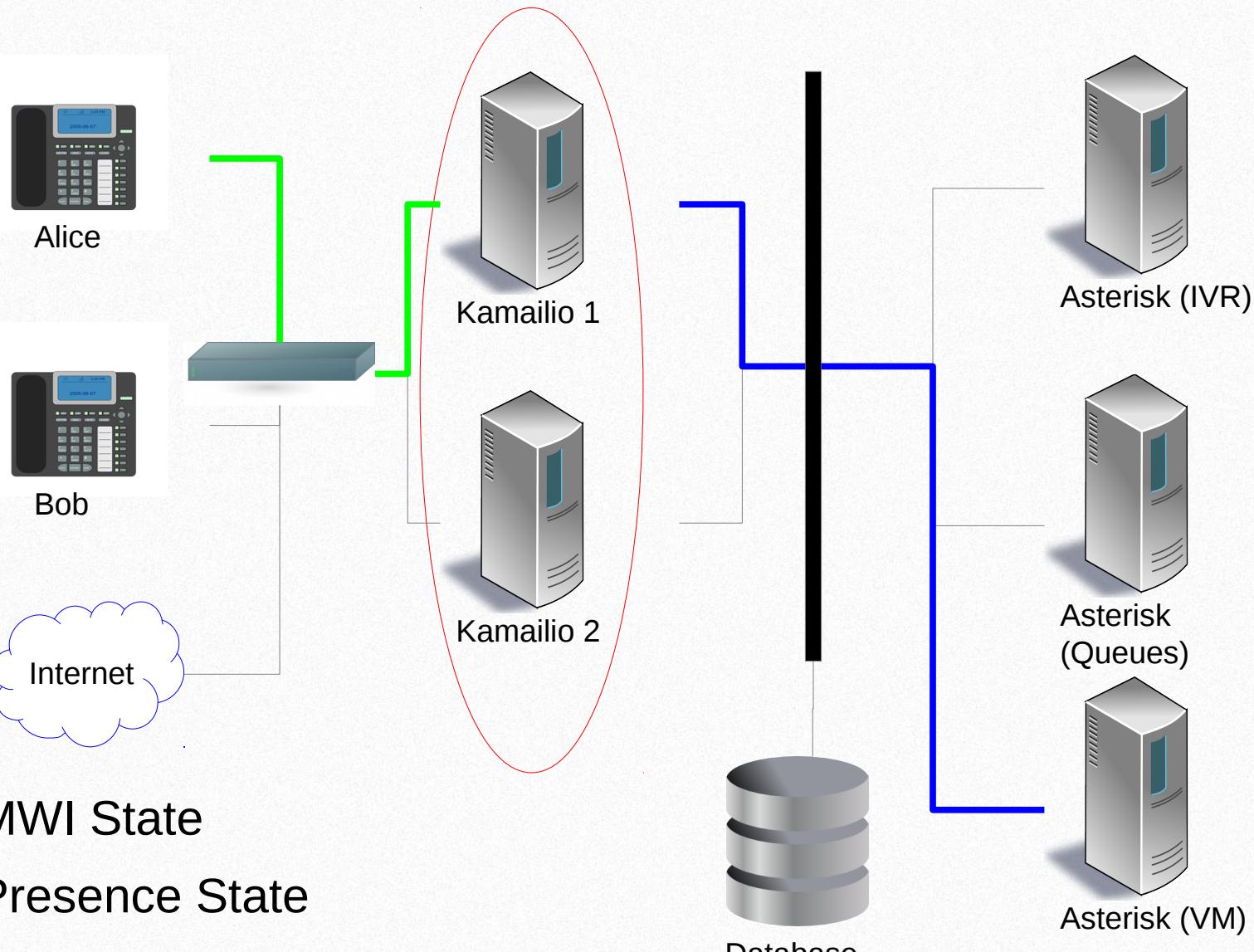
Asterisk 12 and Kamailio: Next Steps



Asterisk 12 and Kamailio: Next Steps

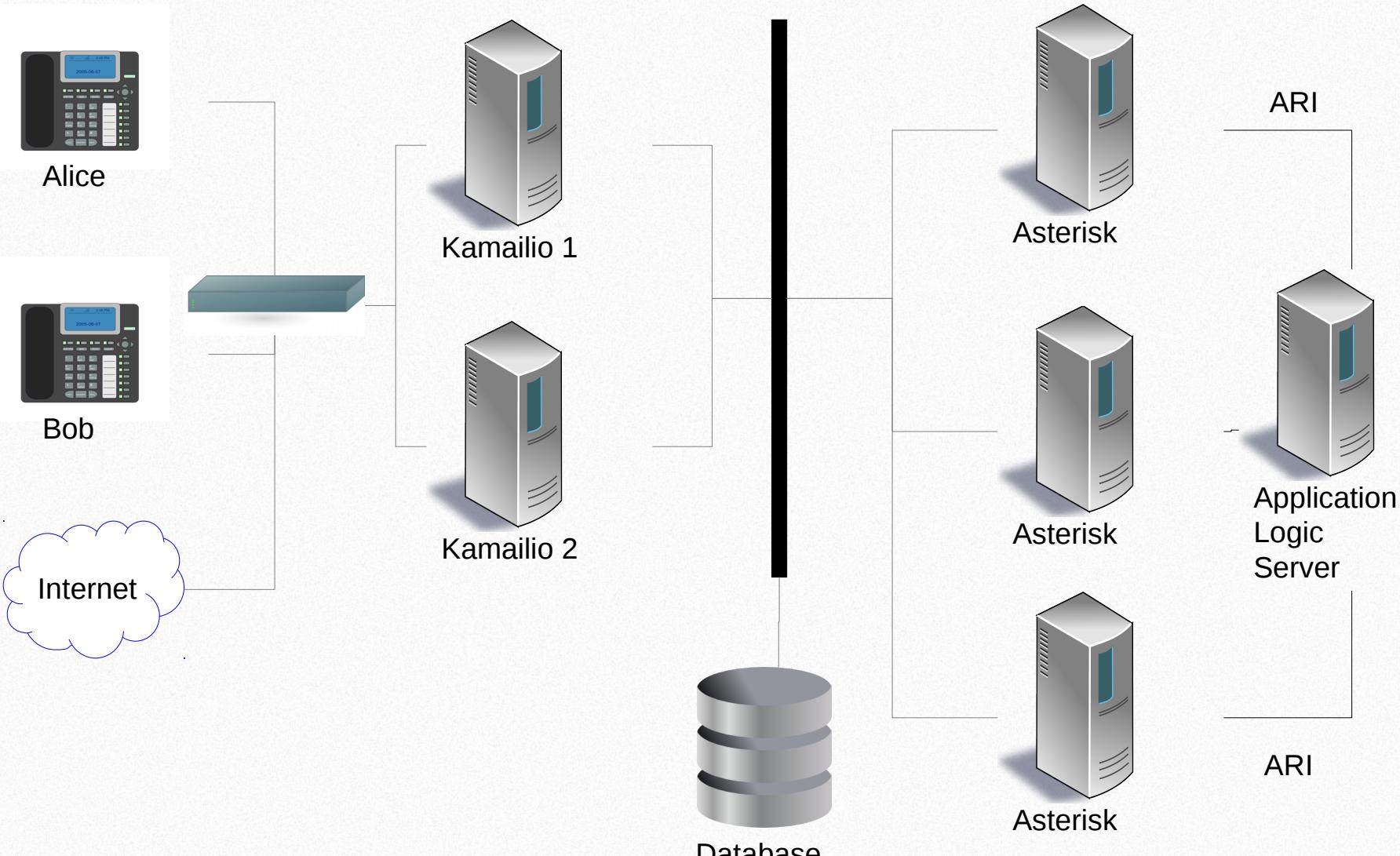


Asterisk 12 and Kamailio: Next Steps



- What happens if we need more of some resource?
 - Not always easy to change purpose of an Asterisk server
 - Major constraining point: dialplan
- Alternative: ARI
 - Treat Asterisk as an application engine
 - Push all dialplan logic out of the Asterisk instances

Asterisk 12 and Kamailio: Next Steps



- Testing
 - 390 Unit Tests
 - 433 Functional Tests
 - 2.1x more tests than Asterisk 11
 - Lots more to go
- PJSIP: Enhance and Extend
- Publish/Subscribe
 - Highly desired
 - RLS
- ARI: enable application logic outside of Asterisk

Questions

?