OFBiz CRM & & Asterisk Call Center Integration Youssef Khaye, Software Developper



A PACHECON E U ROPE

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Agenda

- Who am I?
- Asterisk
- Architecture
- Implementation process
- Outgoing calls
- Incoming calls
- Implementation details
- Demo



Resources

- Asterisk: the Future of Telephony, 2nd edition by Jim Van Meggelen, Leif Madsen, and Jared Smith
- http://www.voip-info.org/
- https://maven.reucon.com/projects/public/asterisk-java
- https://tomcat.apache.org/tomcat-7.0-doc/web-socket-howto.html



Who am I?

- 35 Years old, married, 4 children
- Born In Mauritania
- Bachlor degree from Hashemite University (Jordan)
- Professional Master Admin Systems & Networks from Reims University (France)
- Started working with OFBiz In 2008 (Néréide)
- Since October 2014 have been working for an OFBiz End User company

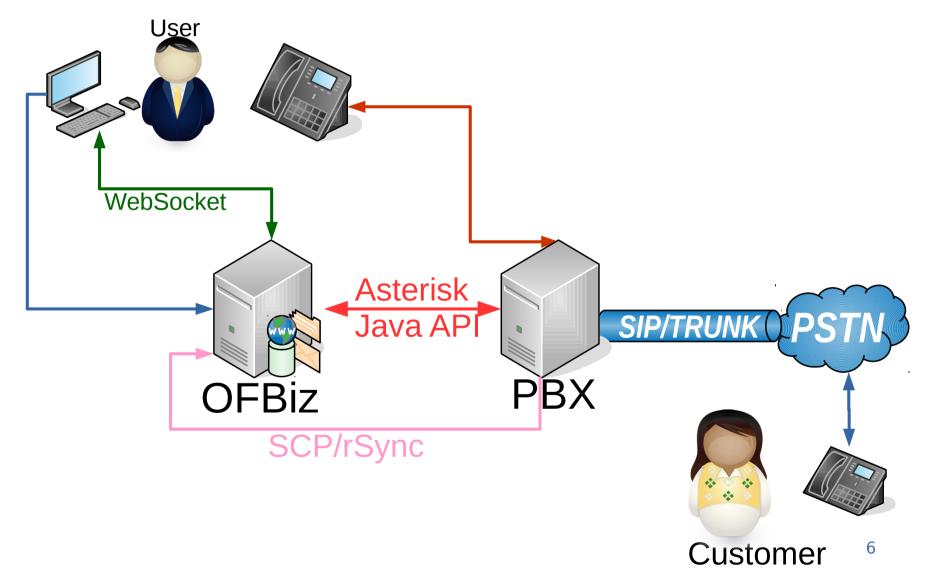


Asterisk

- Open source telephony platform framework(Software PBX)
- Very customizable
- Features
 - Voice-mail
 - Hosted Conferencing
 - Call Queuing
 - > MOH
 - Call Parking
 - > IM
 - > Fax
 - > SMS
 - >
- Too many configuration files
- Well documented but still complicated to master the framework.
- Elastix is good alternative (web interface for configuration)



Architecture

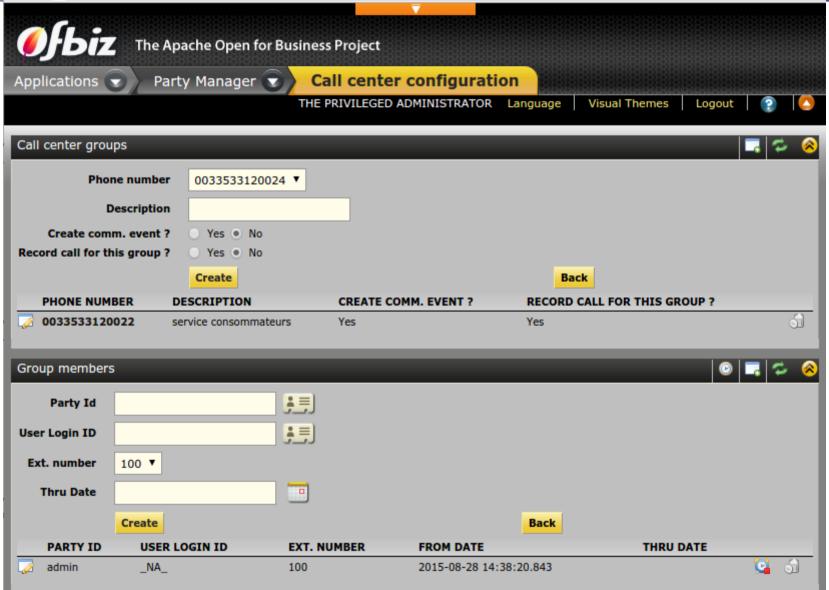




Implementation process

- Initialize Asterisk-java API
- Create SIP accounts, trunk and dialplan
- Assign phone numbers to OFBiz users
- Create call groups, associate theme to external phone lines
- Associate OFBiz users to call groups
- Create websocket on each OFBiz view with an autheticated user







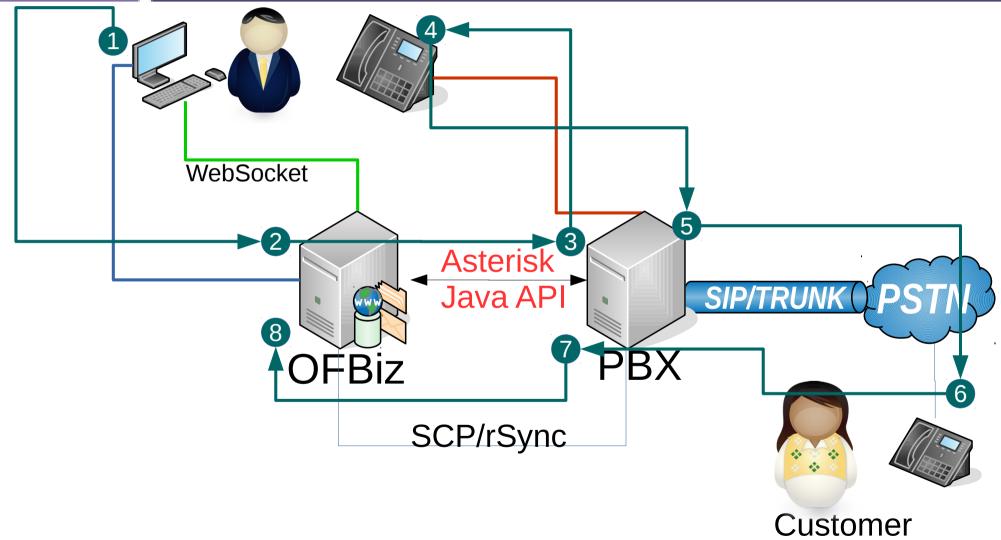
Outgoing call

Outgoing call

- 1) The OFBiz user calls a phone number using the OFBiz screen
- 2) An OFBiz service is called using a normal OFBiz request
- 3) The OFBiz service uses the ASP to send an OriginateRequest to Asterisk
- 4) Asterisk rings the OFBiz user's internal phone number
- 5) The OFBiz user picks up their phone
- 6) Asterisk then dials the requested outgoing phone number
- 7) Asterisk then connects both channels and notifies OFBiz
- 8) OFBiz creates a new CommEvent (and associates a file for recording the call to the CommEvent)
- 9) When the phone call is finished, Asterisk creates the recording file



Outgoing call









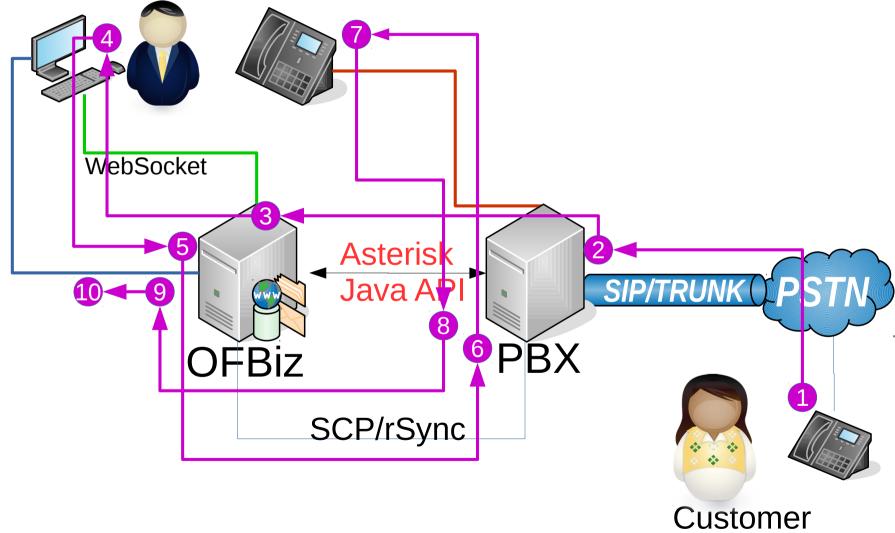
Incoming call

Incoming calls

- 1) A Customer dials the Support Service number
- 2) Asterisk recieves the call and creates a new channel for it
- 3) The ASP recieves a notification for the new event
- 4) OFBiz notifies the support group's connected members about the incoming call
- 5) One of the support group members clicks the Answer button on the OFBiz screen to tell OFBiz that they will accept the call
- 6) OFBiz asks Asterisk to redirect the call to this specific user (support group member)
- 7) Asterisk then rings the user's phone
- 8) The user picks up his phone and speaks to the Customer
- 9) Asterisk notifies OFBiz of the redirection success
- 10) OFBiz creates a new CommEvent with Associated content for the recording file

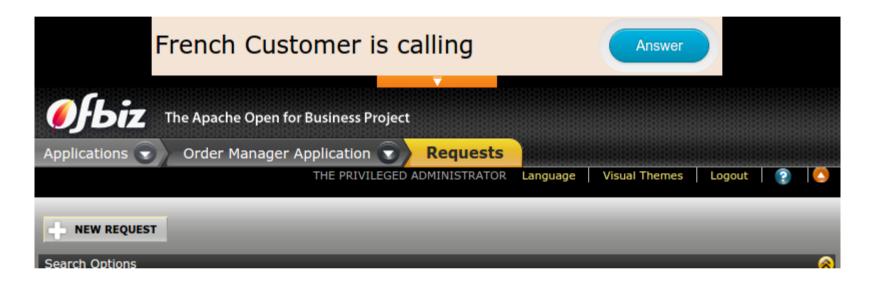


Incoming call



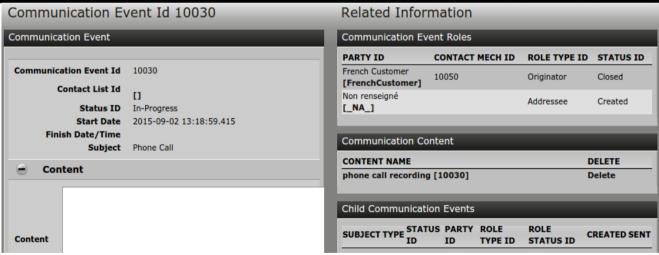


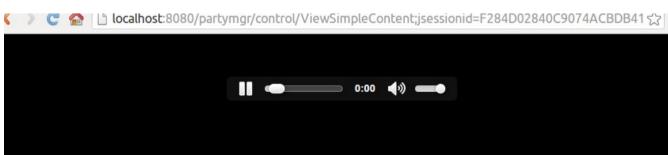
Incoming call notification





List Communications FrenchCustomer							
SUBJECT	TYPE	STATUS ID	PARTY ID	ROLE TYPE ID	ROLE STATUS ID	CREATED	SENT
Phone Call[10030]	Phone	In-Progress	French Customer [FrenchCustomer]	Originator	Closed	9/2/15	9/2/15
No Subject[10020]	Phone	In-Progress	French Customer [FrenchCustomer]	Originator	Closed	8/31/15	8/31/15
No Subject[10010]	Phone	In-Progress	French Customer [FrenchCustomer]	Originator	Closed	8/31/15	8/31/15
No Subject[10000]	Phone	In-Progress	French Customer [FrenchCustomer]	Originator	Closed	8/30/15	8/30/15







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- Apache OFBiz Release 13.07
- Elastix 2.4.0
- Add-on portlet-party from ofbizextra.org
- Tomcat WebSocket API
- Asterisk-Java API



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- AsteriskContainer
 - Read configuration file
 - Initialize connection
 - Bring up Asterisk service provider
- AsteriskServiceProvider
- AsteriskWebSocketServlet
 - externalLoginKey
- Add a small panel on each OFBiz screen (authenticated users)
- Create a new webApp for web sockets



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- Keep manager connection a live Thread
- Call recording Thread
- Asterisk phone numbers reading Thread
- New tables
 - AsteriskPeer
 - CallCenterGroup
 - GroupMembership
- A new Portal Page to configure the Call Center



- WebSocket messages:
 - Json format
 - Message Type
- Client → Server Messages :
 - Get user Profile URL (GURL)
 - Redirect (RD)
- Server → Client Messages:
 - Incoming call (NC)
 - Incoming call redirected (OC)
 - User profile URL (URL)



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Configuration parameters :

- CallRecordingFilesLocation
- RecorOutCAlls
- PartyProfileUrl
- UniqueUserSession
- autoFillExtensionAndTrunk



Q&A

Any Questions?

