
FOURTH EDITION

Asterisk™: The Definitive Guide

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Table of Contents

Foreword.....	xix
Preface.....	xxv
1. A Telephony Revolution.....	1
Asterisk and VoIP: Bridging the Gap Between Traditional and Network	
Telephony	2
The Zapata Telephony Project	2
Massive Change Requires Flexible Technology	3
Asterisk: The Hacker's PBX	4
Asterisk: The Professional's PBX	5
The Asterisk Community	5
The Asterisk Mailing Lists	6
Asterisk Wiki Sites	7
The IRC Channels	7
Asterisk User Groups	7
The Asterisk Documentation Project	8
The Business Case	8
Conclusion	8
2. Asterisk Architecture.....	9
Modules	10
Applications	12
Bridging Modules	15
Call Detail Recording Modules	16
Channel Event Logging Modules	17
Channel Drivers	17
Codec Translators	18
Format Interpreters	19
Dialplan Functions	20

PBX Modules	22
Resource Modules	22
Add-on Modules	26
Test Modules	27
File Structure	27
Configuration Files	27
Modules	27
The Resource Library	28
The Spool	28
Logging	28
The Dialplan	28
Hardware	29
Asterisk Versioning	29
Previous Release Methodologies	30
The Current Release Methodology	31
Simplifying the Version Numbers	32
Conclusion	33
3. Installing Asterisk.....	35
Installation Cheat Sheet	37
Distribution Installation	41
RHEL Server	41
Ubuntu Server	45
Software Dependencies	49
Downloading What You Need	50
Getting the Source via Subversion	51
Getting the Source via wget	51
How to Install It	52
DAHDI	52
LibPRI	54
Asterisk	55
Setting File Permissions	56
Base Configuration	56
Initial Configuration	56
make menuselect	64
Updating Asterisk	70
Common Issues	72
-bash: wget: command not found	73
configure: error: no acceptable C compiler found in \$PATH	73
make: gcc: command not found	73
configure: error: C++ preprocessor “/lib/cpp” fails sanity check	73

configure: error: *** Please install GNU make. It is required to build Asterisk!	73
configure: *** XML documentation will not be available because the 'libxml2' development package is missing.	74
configure: error: *** termcap support not found	74
You do not appear to have the sources for the 2.6.18-164.6.1.el5 kernel installed.	74
E: Unable to lock the administration directory (/var/lib/dpkg/), are you root?	74
Upgrading Asterisk	74
Conclusion	76
4. Initial Configuration Tasks.....	77
asterisk.conf	77
The [directories] Section	77
The [options] Section	78
The [files] Section	82
The [compat] Section	82
modules.conf	83
The [modules] Section	84
indications.conf	84
musiconhold.conf	86
Converting Music to a Format That Works Best with Asterisk	87
Default musiconhold.conf file	89
Additional Configuration Files	91
Conclusion	91
5. User Device Configuration.....	93
Telephone Naming Concepts	94
Hardphones, Softphones, and ATAs	96
Configuring Asterisk	98
How Channel Configuration Files Work with the Dialplan	99
sip.conf	101
iax.conf	107
Modifying Your Channel Configuration Files for Your Environment	112
Digium Phones with Asterisk	113
Loading Your New Channel Configurations	113
The Asterisk CLI	113
Testing to Ensure Your Devices Have Registered	114
Analog Phones	115
A Basic Dialplan to Test Your Devices	118
Under the Hood: Your First Call	119

Conclusion	120
6. Dialplan Basics.....	121
Dialplan Syntax	121
Contexts	122
Extensions	124
Priorities	125
Applications	127
The Answer(), Playback(), and Hangup() Applications	128
A Simple Dialplan	129
Hello World	129
Building an Interactive Dialplan	131
The Goto(), Background(), and WaitExten() Applications	131
Handling Invalid Entries and Timeouts	133
Using the Dial() Application	134
Using Variables	137
Pattern Matching	140
Includes	145
Conclusion	145
7. Outside Connectivity.....	147
The Basics of Trunking	147
Fundamental Dialplan for Outside Connectivity	148
PSTN Circuits	150
Traditional PSTN Trunks	150
Installing PSTN Trunks	152
VoIP	161
Coping with Network Address Translation	161
PSTN Termination	170
PSTN Origination	171
VoIP to VoIP	173
Configuring VoIP Trunks	173
Emergency Dialing	181
Conclusion	183
8. Voicemail.....	185
Comedian Mail	185
The [general] Section	186
The [zonemessages] Section	195
The Contexts Section	196
An Initial voicemail.conf File	200
Standard Voicemail KeyMap	201

Dialplan Integration	201
The VoiceMail() Dialplan Application	201
The VoiceMailMain() Dialplan Application	204
Creating a Dial-by-Name Directory	205
Using a Jitterbuffer	205
Storage Backends	206
Linux Filesystem	206
ODBC	207
IMAP	207
Using Asterisk as a Standalone Voicemail Server	208
Integrating Asterisk into a SIP Environment as a Standalone	
Voicemail Server	208
SMDI (Simplified Message Desk Interface)	212
Database Integration	213
Conclusion	213
9. Internationalization.....	215
Devices External to the Asterisk Server	217
PSTN Connectivity, DAHDI, Digium Cards, and Analog Phones	219
DAHDI Drivers	221
Asterisk	224
Caller ID	224
Language and/or Accent of Prompts	225
Time/Date Stamps and Pronunciation	226
Conclusion—Easy Reference Cheat Sheet	228
10. Deeper into the Dialplan.....	231
Expressions and Variable Manipulation	231
Basic Expressions	231
Operators	232
Dialplan Functions	234
Syntax	235
Examples of Dialplan Functions	235
Conditional Branching	236
The GotoIf() Application	236
Time-Based Conditional Branching with GotoIfTime()	240
Macros	242
Defining Macros	243
Calling Macros from the Dialplan	244
Using Arguments in Macros	245
GoSub	246
Defining Subroutines	246

Calling Subroutines from the Dialplan	247
Using Arguments in Subroutines	248
Returning from a Subroutine	249
Local Channels	250
Using the Asterisk Database (AstDB)	253
Storing Data in the AstDB	254
Retrieving Data from the AstDB	254
Deleting Data from the AstDB	254
Using the AstDB in the Dialplan	255
Creating a Hot-Desking Application with AstDB	256
Handy Asterisk Features	261
Zapateller()	261
Call Parking	261
Conferencing with MeetMe()	263
Conferencing with ConfBridge()	264
Conclusion	266
11. Parking, Paging, and Conferencing.....	267
features.conf	267
The [general] section	268
The [featuremap] Section	271
The [applicationmap] Section	272
Application Map Grouping	275
Parking Lots	276
Overhead and “Underchin” Paging (a.k.a. Public Address)	277
Places to Send Your Pages	278
Zone Paging	284
Advanced Conferencing	284
The [general] Section	285
Options for User Profiles	285
Options for Bridge Profiles	287
ConfBridge Menu Options	289
Enabling a PIN	290
Waiting for the Marked User to Join	292
Using ConfBridge() Menus	293
Enabling Videoconferencing	296
Conclusion	298
12. Internet Call Routing.....	299
DNS and SIP URIs	300
The SIP URI	300
SRV Records	301

Accepting Calls to Your System	302
Dialing SIP URIs from Asterisk	309
ENUM and E.164	310
E.164 and the ITU	310
ENUM	311
Asterisk and ENUM	312
ISN, ITAD, and freenum.org	313
Got ISN?	314
ITAD Subscriber Numbers (ISNs)	315
Management of Internet Numbering	315
IP Telephony Administrative Domains (ITADs)	316
Create a DNS Entry for Your ITAD	317
Testing Your ITAD	318
Using ISNs in Your Asterisk System	318
Security and Identity	320
Toll Fraud	321
Spam over Internet Telephony (SPIT)	322
Distributed Denial of Service Attacks	323
Phishing	323
Security Is an Ongoing Process	323
Conclusion	323
13. Automatic Call Distribution (ACD) Queues.....	325
Creating a Simple ACD Queue	326
Queue Members	331
Controlling Queue Members via the CLI	331
Defining Queue Members in the queues.conf File	333
Controlling Queue Members with Dialplan Logic	333
Automatically Logging Into and Out of Multiple Queues	335
An Introduction to Device State	339
The queues.conf File	341
The agents.conf File	349
Advanced Queues	351
Priority Queue (Queue Weighting)	351
Queue Member Priority	353
Changing Penalties Dynamically (queuerules.conf)	354
Announcement Control	356
Overflow	361
Using Local Channels	363
Queue Statistics: The queue_log File	367

Conclusion	370
14. Device States.....	371
Device States	371
Checking Device States	372
Extension States	373
Hints	374
Checking Extension States	375
SIP Presence	376
Asterisk Configuration	376
Using Custom Device States	378
An Example	378
Distributed Device States	380
Using Corosync	380
Using XMPP	385
Shared Line Appearances	389
Installing the SLA Applications	389
Configuration Overview	390
Key System Example with Analog Trunks	390
Key System Example with SIP Trunks	394
Alternate Key System Example with SIP Trunks	396
Shared Extension Example	399
Additional Configuration	401
Limitations	402
Creating a Callback Service	403
Conclusion	405
15. The Automated Attendant.....	407
An Auto Attendant Is Not an IVR	407
Designing Your Auto Attendant	408
The Greeting	409
The Main Menu	410
Timeout	411
Invalid	411
Dial by Extension	412
Building Your Auto Attendant	412
Recording Prompts	412
The Dialplan	414
Delivering Incoming Calls to the Auto Attendant	416
IVR	416

Conclusion	416
16. Relational Database Integration.....	417
Installing and Configuring PostgreSQL and MySQL	418
Installing PostgreSQL for RHEL	418
Installing PostgreSQL for Ubuntu	418
Installing MySQL for RHEL	419
Installing MySQL for Ubuntu	419
Configuring PostgreSQL	419
Configuring MySQL	421
Installing and Configuring ODBC	422
Configuring ODBC for PostgreSQL	424
Configuring ODBC for MySQL	425
Configuring ODBC for Microsoft SQL	426
Validating the ODBC Connector	427
Compiling the ODBC Modules for Asterisk	428
Configuring res_odbc to Allow Asterisk to Connect Through ODBC	428
Managing Databases	429
Troubleshooting Database Issues	430
SQL Injection	430
Powering Your Dialplan with func_odbc	431
A Gentle Introduction to func_odbc	432
Getting Funky with func_odbc: Hot-Desking	433
Using Realtime	449
Static Realtime	450
Dynamic Realtime	453
Storing Call Detail Records (CDR)	458
ODBC Voicemail Message Storage	463
Compiling the app_voicemail Module to Support ODBC Storage	464
Creating the Large Object Type for PostgreSQL	465
ODBC Voicemail Storage Table Layout	467
Configuring voicemail.conf for ODBC Storage	468
Testing ODBC Voice Message Storage	469
Database Integration of ACD Queues	473
Storing queues.conf in a Database	473
Storing Dialplan Parameters for a Queue in a Database	476
Writing queue_log to Database	477
Conclusion	478
17. Interactive Voice Response.....	479
What Is IVR?	479
Components of an IVR	480

IVR Design Considerations	482
Asterisk Modules for Building IVRs	483
CURL	483
func_odbc	483
AGI	483
AMI	483
A Simple IVR Using CURL	483
Installing the cURL Module	484
The Dialplan	484
A Prompt-Recording Application	484
Speech Recognition and Text-to-Speech	486
Text-to-Speech	486
Speech Recognition	486
Conclusion	487
18. External Services.....	489
Calendar Integration	490
Compiling Calendaring Support into Asterisk	490
Configuring Calendar Support for Asterisk	492
Triggering Calendar Reminders to Your Phone	494
Controlling Calls Based on Calendar Information	499
Writing Call Information to a Calendar	501
Additional Features	503
Voicemail IMAP Integration	503
Compiling IMAP Voicemail Support into Asterisk	504
Using XMPP (Jabber) with Asterisk	511
Compiling XMPP Support into Asterisk	512
Jabber Dialplan Commands	512
chan_motif	518
Out-of-Band Messaging (Messaging Infrastructure)	522
xmpp.conf Configuration	522
sip.conf Configuration	523
Dialplan Configuration	524
LDAP Integration	525
Configuring OpenLDAP	526
Compiling LDAP Support into Asterisk	528
Configuring Asterisk for LDAP Support	529
Text-to-Speech Utilities	531
Festival	531
Cepstral	534

Conclusion	534
19. Fax.....	535
What Is a Fax?	535
Ways to Handle Faxes in Asterisk	535
spandsp	536
Obtaining spandsp	536
Compiling and Installing spandsp	537
Adding the spandsp Library to Your libpath	537
Recompiling Asterisk with spandsp Support	537
Disabling spandsp (Should You Want to Test Digium Fax)	538
Digium Fax For Asterisk	538
Obtaining Digium FFA	538
Disabling Digium FFA (Should You Want to Test spandsp)	539
Incoming Fax Handling	539
Fax to TIFF	540
Fax to Email	541
Fax Detection	542
Using T.38	543
Outgoing Fax Handling	544
Transmitting a Fax from Asterisk	544
File Format for Faxing	545
An Experiment in Email to Fax	546
Fax Pass-Through	549
Using Fax Buffers in chan_dahdi.conf	549
T.38 Gateway	550
Conclusion	551
20. Asterisk Manager Interface (AMI).....	553
Quick Start	553
AMI over TCP	554
AMI over HTTP	555
Configuration	556
manager.conf	556
http.conf	561
Protocol Overview	562
Message Encoding	563
AMI over HTTP	564
Call Files	569
Example Usage	571
Originating a Call	571
Redirecting a Call	573

Originating a Call Using Python and StarPy	574
Development Frameworks	579
CSTA	580
Interesting Applications	580
Flash Operator Panel	580
Conclusion	581
21. Asterisk Gateway Interface (AGI).....	583
Quick Start	583
AGI Variants	585
Process-Based AGI	585
DeadAGI Is Dead	586
FastAGI—AGI over TCP	586
Async AGI—AMI-Controlled AGI	587
AGI Communication Overview	588
Setting Up an AGI Session	588
Commands and Responses	590
Ending an AGI Session	594
Example: Account Database Access	596
Development Frameworks	599
Conclusion	599
22. Clustering.....	601
Traditional PBXs	602
Hybrid Systems	603
Pure Asterisk, Nondistributed	605
Asterisk and Database Integration	606
Single Database	606
Replicated Databases	608
Asterisk and Distributed Device States	609
Distributing Device States over a LAN	610
Distributing Device States over a WAN	611
Multiple Queues, Multiple Sites	613
Conclusion	614
23. Distributed Universal Number Discovery (DUNDi).....	617
How Does DUNDi Work?	617
The dundi.conf File	619
Configuring Asterisk for Use with DUNDi	621
General Configuration	622
Initial DUNDi Peer Definition	623
Creating Mapping Contexts	625

Using Mapping Contexts with Peers	627
Allowing Remote Connections	629
Controlling Responses	630
Performing Lookups from the Dialplan	634
Conclusion	637
24. System Monitoring and Logging.....	639
logger.conf	639
Reviewing Asterisk Logs	641
Logging to the Linux syslog Daemon	642
Verifying Logging	643
Log Rotation	643
Call Detail Records	643
CDR Contents	644
Dialplan Applications	645
cdr.conf	645
Backends	646
Example Call Detail Records	653
Caveats	653
CEL (Channel Event Logging)	654
Channel Event Types	654
Channel Event Contents	655
Dialplan Applications	656
cel.conf	657
Backends	657
Example Channel Events	663
SNMP	668
Installing the SNMP Module for Asterisk	668
Configuring SNMP for Asterisk Using OpenNMS	669
Monitoring Asterisk with OpenNMS	672
Conclusion	673
25. Web Interfaces.....	675
Flash Operator Panel	676
Queue Status and Reporting	676
Queue Status Display	677
Queue Reporting	677
Call Detail Records	677
A2Billing	678
Conclusion	678
26. Security.....	679

Scanning for Valid Accounts	679
Authentication Weaknesses	681
Fail2ban	681
Installation	682
Configuration	683
Encrypted Media	687
Dialplan Vulnerabilities	687
Securing Asterisk Network APIs	689
IAX2 Denial of Service	689
Other Risk Mitigation	691
CLI Permissions	693
Resources	694
Conclusion—A Better Idiot	694
27. Asterisk: A Future for Telephony.....	695
The Problems with Traditional Telephony	695
Closed Thinking	696
Limited Standards Compliance	697
Slow Release Cycles	697
Refusing to Let Go of the Past and Embrace the Future	698
Paradigm Shift	698
The Promise of Open Source Telephony	698
The Itch that Asterisk Scratches	698
Open Architecture	699
Standards Compliance	700
Lightning-Fast Response to New Technologies	700
Passionate Community	700
Some Things that Are Now Possible	701
The Future of Asterisk	705
Speech Processing	705
High-Fidelity Voice	706
Video	707
WebRTC	709
Wireless	709
Unified Messaging	710
Peering	710
Challenges	711
Opportunities	714
A. Understanding Telephony.....	717
B. Protocols for VoIP.....	737

C. Preparing a System for Asterisk.....	761
Index.....	787