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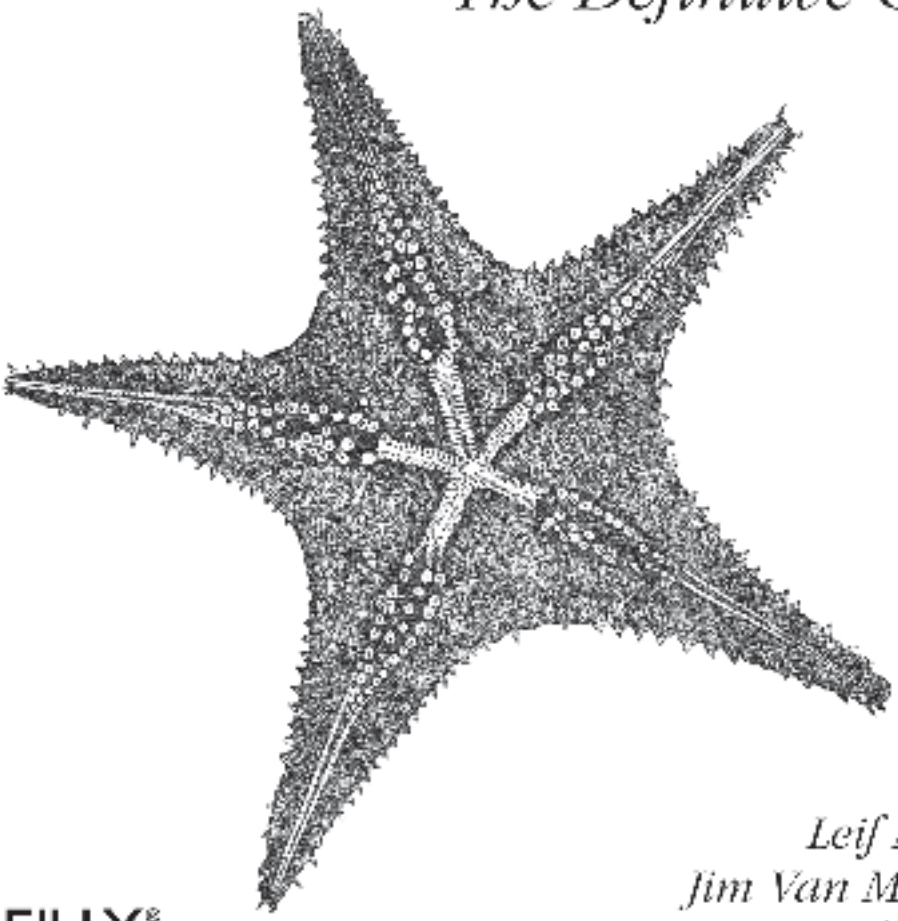
Asterisk

The Definitive Guide



Leif Madsen,
Jim Van Meggelen
& Russell Bryant

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Networking/Telephony

Asterisk: The Definitive Guide

Design a complete VoIP or analog PBX with Asterisk, even if you have no previous Asterisk experience and only basic telecommunications knowledge. This bestselling guide makes it easy, with a detailed roadmap to installing, configuring, and integrating this open source software into your existing phone system.

Great for Linux administrators, developers, and power users, this book shows you how to write a basic dialplan step by step, and quickly brings you up to speed on the latest Asterisk features in version 1.8.

- Integrate Asterisk with analog, VoIP, and digital telephony systems
- Build a simple interactive dialplan, and dive into advanced concepts
- Use Asterisk's voicemail options—including a standalone voicemail server
- Build a menuing system and add applications that act on caller input
- Incorporate a relational database with MySQL and PostgreSQL
- Connect to external services such as LDAP, calendars, XMPP, and Skype
- Use Automatic Call Distribution to build a call queuing system
- Learn how to use Asterisk's security, call routing, and faxing features

"If this is your first project with Asterisk, this book will take you to the point where you're able to stun everyone you know with the sophisticated tricks that your phone system can accomplish."

—John Todd
Asterisk and VoIP Evangelist

Leif Madsen is a consultant specializing in Asterisk clustering and call center integration. He is a co-founder of the Asterisk Documentation Project.

Jim Van Meggelen has more than 20 years of enterprise telecom experience. He is a partner in Core Telecom Innovations, Inc., and iConverged LLC.

Russell Bryant is the engineering manager for the Open Source software team at Digium, Inc. He has been a core member of the Asterisk development team since 2001.

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THIRD EDITION

Asterisk™: The Definitive Guide

Leif Madsen, Jim Van Meggelen, and Russell Bryant

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Asterisk™: The Definitive Guide, Third Edition
by Leif Madsen, Jim Van Meggelen, and Russell Bryant

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Foreword

"There's more than one way to do it." I've been working with Asterisk for nine years, and this motto becomes more true with each release, each added feature, and each clever person who attacks a telecommunications problem with this incredibly flexible toolkit. I had the fantastic opportunity to work as the community manager for the Asterisk project at Digium for two years, which gave me one of the best vantage points for seeing the scope and imagination of the worldwide development effort pushing Asterisk forward. The depth and breadth of Asterisk is staggering—installations with hundreds of thousands of users are now commonplace. I see Asterisk making deep inroads into the financial, military, hospital, Fortune 100 enterprise, service provider, calling card, and mobile environments. In fact, there really aren't any areas that I can think of where Asterisk isn't now entrenched as the default choice when there is a need for a generalized voice tool to do "stuff."

Asterisk has been emblematic of the way that open source software has changed business—and changed the world. My favorite part of any Asterisk project overview or conference talk is answering questions from someone new to Asterisk. As I continue to answer "Yes, it can do that," I watch as the person's eyes grow wide. The person starts to smile when he *really* starts to think about new things to do that his old phone or communication system couldn't possibly have done. Radio integration? Sure. Streaming MP3s into or out of phone calls? OK. Emailing recorded conference calls to the participants? No problem. Integration of voice services into existing Java apps? Easy. Fax? Instant messages? IVRs? Video? Yes, yes, yes, yes.

The affirmative answers just keep flowing, and at that point, the best thing to do is to sit the person down and start showing him quick demonstrations of how Asterisk can be quickly deployed and developed. Then, I typically point the person toward the first edition of this book, *Asterisk: The Future of Telephony*, and set him loose. In just a few hours of development (or longer, of course), companies can change the way they deliver products to customers, nonprofits can overhaul how their users interact with the services they offer, and individuals can learn to build a perfectly customized call-handling system for their mobile and home phones. Asterisk scales up and down from individual lines to vast multiserver installations across multiple continents, but the way to start is

to install the package, open up some of the configuration files, and start looking at examples.

From the basic beginnings of a PBX that Mark Spencer coded in 1999, the Asterisk project, with the help of thousands of developers, has moved from simply connecting phone calls and has matured into a platform that can handle voice, video, and text across dozens of virtual and physical interface types. The creation and growth of Asterisk were the inescapable results of the convergence of the four horsemen of the proprietary hardware apocalypse: open source development ideas, the Internet, Moore's Law, and the plummeting costs of telecommunications. Even hardware vendors who may be frightened of Asterisk from a competitive standpoint are using it in their labs and core networks: almost all devices in the Voice-over-IP world are tested with Asterisk, making it the most compatible system across vendors.

At a recent communications conference I attended, the question "Who uses Asterisk?" was posed to the 1000-plus crowd. Nearly 75 percent raised their hands. Asterisk is a mature, robust software platform that permeates nearly every area of the telecommunications industry and has firmly cemented itself as one of the basic elements in any open source service delivery system. I tell people that it's reasonable for anyone delivering services both via phone and web to want to add an "A" for Asterisk to the LAMP (Linux, Apache, MySQL, [Perl/Python/PHP]) acronym, making it LAAMP. (LAMA-P was another option, but for some reason nobody seems to like that version...I don't know why.)

The expansion of this book to include more examples is something I've been looking forward to for some time. Asterisk is accessible because of the ease with which a novice can understand basic concepts. Then it continues to succeed as the novice becomes a pro and starts tapping the "other ways to do it" with more sophisticated implementations, using AGI with Java, Perl, or Python (or one of the other dozen or so supported languages), or even writing her own custom apps that work as compile-time options in Asterisk. But the first step for anyone, no matter what his or her skill level, is to look at examples of basic apps others have written. Leif, Jim, and Russell have not only put together a fantastic compendium of Asterisk methods, but they have also provided an excellent list of examples that will let the novice or expert quickly learn new techniques and "more than one way to do it."

Asterisk 1.*x* is fantastically powerful and can solve nearly any voice problem you might have. For those of you building the most complex installations, there is even more interesting work—which will be realized quite soon—in development. The currently-in-development Asterisk SCF (Scalable Communications Framework) is being built as an adjunct open source project to allow Asterisk 1.*x* systems to scale in even more powerful ways—stay tuned, or better yet, get involved with the project as a developer.

If you're an experienced Asterisk developer or integrator, I'm sure this book will have a few "Hey, that's a neat way to do it!" moments for you, which is one of the joys of Asterisk. If this is your first project with Asterisk, I'd like to welcome you to the huge community of users and developers dedicated to making Asterisk better. This book will take you from a vague idea of doing something with computers and voice communication to the point where you're able to stun everyone you know with your phone system's sophisticated tricks.

You're encouraged to participate in the online mailing lists, IRC chatrooms, and yearly AstriCon conference that provide up-to-the-second news and discussion surrounding the project. Without your interest, input, and code, Asterisk wouldn't exist. Open source projects are hungry for new ideas and excellent contributions: I encourage you to be a participant in the Asterisk community, and I look forward to seeing your questions and examples in the next edition of this book.

—John Todd

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