

Skype for Asterisk®



Digium is the innovative force behind Asterisk, the world's most widely used open source telephony platform. In addition, Digium develops the software, hardware, and other components needed to create a unified communications solution for a small or medium business.



Skype enables businesses and individuals to make free Skype-to-Skype video and voice calls, send instant messages and share files. And you can reach landlines and mobiles almost anywhere in the world at great rates too.

Finally, there's a cost-effective telephony solution for interoffice and worldwide communication. Two telephony market leaders, Digium® and Skype™, have joined forces to develop Skype for Asterisk®. With Skype for Asterisk, you can make, receive and transfer Skype calls from within your Asterisk system using your existing hardware. You'll also be able to use Asterisk applications like call routing and phone menus to manage incoming calls.

Skype for Asterisk is an add-on channel driver for Asterisk-based PBX systems. Adding Skype for Asterisk to any Asterisk server enables complete access to the Skype community, including low-cost PSTN access and free calling to more than 440 million registered Skype users.

Skype for Asterisk integrates with the Skype community. There is no secondary piece of hardware to manage as Skype for Asterisk will run directly from an Asterisk-based PBX.

Skype for Asterisk Key Benefits

- ▼ Save money:
 - Free calling to more than 440 million Skype registered users worldwide directly from your Asterisk server.
 - Great rates for worldwide inbound calling DIDs via online numbers (SkypeIn).
 - Call landlines and mobiles worldwide at great rates (SkypeOut).
- ▼ Add Skype to your call routing tables to optimize global calling costs.
- ▼ Add a click-to-call button to your website or email so customers can quickly contact you.
- ▼ Allows customers to call via a local online number.
- ▼ Perfect for the remote employee, as the office is one click away with free calling.
- ▼ Communicate securely with free, high-quality, encrypted Asterisk-to-Asterisk calls using Skype.



Key Features that are enabled by Skype for Asterisk

Please note that paid-for products are available to the extent that you select and purchase them through Skype's Business Control Panel.



In the News

"The mash-up of Skype with Asterisk, the leading open-source IP PBX, marks another advance in the evolution of the PBX from a premises-based box to a hosted and fully open-source telephony solution."

Richard Martin
Phone Plus Magazine



- Make Skype-to-Skype calls.

- Receive calls with online numbers (SkypeIn).



- Calls to landlines and mobile phones.

- Make and receive multiple concurrent Skype calls from the same Skype account.



- Make world-wide PSTN calls to landline and mobile phones (SkypeOut).

- DTMF support for incoming and outgoing calls.



- Set and retrieve online status.

- Read Skype profile fields from incoming calls.



- Set privacy settings.

- Handle incoming Skype calls using Asterisk applications such as voicemail, ACD, MeetMe conferencing, etc.



- Simultaneous access from both Asterisk and the Skype desktop client.

- Trunk calls between Asterisk servers over Skype.



- Supports G.711 and G.729 (included) codecs.

Skype for Asterisk Support

Skype For Asterisk comes with 90 days of installation support from the time of purchase. For subscribers of Open Source Asterisk and/or Asterisk Business Edition™ support, incidents can be used with Skype For Asterisk, as long as the subscription is current.

Skype for Asterisk is supported on Asterisk 1.4.x and 1.6.x, Asterisk Business Edition versions C.x, and AsteriskNOW.

Support can be requested at www.digium.com/support

For more information, go to www.digium.com/skype