



DIGIUM AND GRANDSTREAM CREATE HIGH-PERFORMANCE, OPEN-SOURCE VOIP SOLUTION FOR THE SME MARKET

*Grandstream SIP Enterprise Voice Phones & ATAs officially interoperable
with Digium Asterisk Business Edition™ Platform*

ASTRICON - PHOENIX, AZ (September 24, 2007) — Grandstream Networks, Inc. (www.grandstream.com), a leading designer and manufacturer of next generation IP Voice and Video products for broadband networks and Digium (www.digium.com), the leading provider of open source telephony software platforms for business, announced today the completion of interoperability certification between Digium's Asterisk Business Edition™ platform and the Grandstream GXP 2020 and Multi-line SIP Enterprise Phone Series, and the HandyTone 50x Analog Telephone Adaptor Series.

The Grandstream product portfolio delivers excellent voice quality and an advanced feature set that is easy to use and easy to provision through an automated secure provisioning process. Digium's Asterisk Business Edition™ Platform provides traditional and advanced telecommunications applications based on open-source software. Together, the Grandstream and Digium IP telephony solution creates a very cost-effective, auto-configurable IP telephony solution for the world-wide SME market.

"Our SIP product portfolio offers a very compelling, competitive price-performance benchmark. The combination of Asterisk Business Edition™ and Grandstream SIP products creates an awesome opportunity for the small to medium sized business to create a world-class advanced IP communications system. Our official interoperability also means Asterisk™ system integrators can now deploy a high quality and cost-effective Voice-over-IP solution to a broad market," commented David Li, CEO of Grandstream Networks. "We believe this combined solution offers exceptional value for the SME market and we look forward to a prosperous relationship with Digium."

"Digium and Grandstream have worked together to ensure that Asterisk Business Edition™ and Grandstream products are officially interoperable, so now our customers have even more certified options for reliable and cost-effective VoIP solutions." said Jim Webster, Director of Technology Partnerships for Digium. "Small-medium enterprises world-wide are looking for high-performance converged VoIP solutions that deliver advanced features and are easy to manage and easy to deploy. The SME now has a new choice in advanced open-source IP communications systems."

Grandstream's GXP 2020 and GXP2010 SIP phones deliver superior performance and an advanced feature set. The GXP 2020 Series offers dual 10/100 auto-sensing Ethernet ports with integrated PoE, individual SIP accounts, advanced XML applications, MLS (multi-language support) in 5 languages and remote configuration for quick deployment over a large network. The GXP series offers a compelling package for both service providers and enterprise customers. The GXP Series retails for under \$169.

The HandyTone 50x Analog Telephone Adaptor series is the second generation of ATAs. The HT502 and HT503 offer either 2 FXS ports or programmable FXO/FXS port combination respectively, are based on SIP standard and feature an integrated high performance router, port status and message waiting LED. The HT50x series delivers more powerful feature functionality and significantly better data router performance than its predecessors and has a new sleek design. These ATAs are lightweight and compact and work with any phone or fax, support a broad range of codecs and T.38 fax. The HT50x Series ATAs are ideal for residential or 'road-warrior' users. The HT Series retails for under \$95.

About Grandstream Networks

Grandstream Networks, Inc. is an award-winning designer and manufacturer of next generation IP voice & video products for broadband networks. Grandstream's products deliver superb sound and picture quality,

rich telephony features, full compliance with industry standards, and broad interoperability with most service providers and 3rd party SIP based VoIP products. Grandstream is consistently recognized in the VoIP industry for their innovation, affordability and superior value in their products. Grandstream Networks is a private company headquartered in Brookline, MA with offices in Los Angeles, Dallas and China. For more information, please visit www.grandstream.com.

About Digium

Digium®, Inc., the Asterisk® Company, created, owns and is the innovative force behind Asterisk, the most widely used open source telephony software. Since its founding in 1999, Digium has grown to become the open source alternative to the traditional communication providers, with offerings that cost as much as 80 percent less than proprietary voice communication platforms. Digium makes Asterisk available to the open source community under the GNU General Public License (GPL) and uses business-class Asterisk to power a broad family of products for small, medium and large businesses. The company's product line also includes a wide range of hardware to enable resellers and customers to design their own voice over IP (VoIP) systems. This year alone, more than 12 billion minutes of calls will be made using Asterisk. More information is available at www.digium.com.

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