

OnDO PBX

datasheet

OUTLINE

OnDO PBX is an enhanced office telephony system that provides robust, high performance, and intelligent IP-PBX functionality. This IP-PBX software supports the industry standard VoIP (Voice over Internet Protocol), SIP (Session Initiation Protocol). Designed for scalability from smaller to larger offices, OnDO PBX delivers a highly interoperable product with flexible migration.

KEY BENEFITS

Advanced Administration System

The advanced design of the administration system will facilitate your understanding of SIP technology and minimize the time it takes you to learn the software. OnDO PBX's web-based administrative tool allows you to easily navigate through the configuration process. Ultimately, save time and money with your VoIP telephony deployment with the user-friendly administration system.

Lower Total Cost of Ownership

Owning a VoIP telephony system, like OnDO PBX, can lower your office expenses. Using our advanced administration system, you can save time and money on system maintenance, technical support, and training. Additionally, using VoIP technology can save long distance call charges for your offices with a minimal upfront investment.

Sophisticated and Customizable PBX Functionality

Despite its easy installation and maintenance, OnDO PBX offers the sophisticated functionality necessary in a mid-sized company's telephone systems. Moreover, all features of OnDO products offer maximum flexibility for customization to fit your needs and requirements.

Trusted and reliable

We have a proven record of stability that you can benefit from in your office. Our product has been used successfully in busy call centers where telephony is the central role of their business. Trusted by telephony businesses around the world, this superior technology is now available for your office at an affordable price.

Endless Options

Because of its exceptional product design, OnDO PBX is customizable to fit your company's unique needs and requirements. It offers intelligent call management, using ARS features or complex call forwarding setting, you can enhance the functionality and settings without purchasing additional systems or software. Additionally, you can easily integrate with new or existing applications.

FUNCTIONS

Auto Attendant	Call Transfer	Realtime Call Status View
Automatic Route Selection (ARS)	Attended Transfer	Ring Groups (Simultaneous Ringing)
Automatic Call Routing via VoIP Gateways	Unattended Transfer	Ringer Time Setting
Automatic Call Routing via ITSPs	Call Park	Schedule
ARS Failover *	Call Pickup	Supervising (Call Monitoring)
Call Forwarding	Call Pickup Group	Voicemail
Busy/No Answer Call Forwarding	Call Queue	Voicemail messages retrieval from Phone,
Unconditional Call Forwarding	Round Robin	email, or Web-Admintool
Round Robin Call Forwarding	Top-Down	Personal/Alternate/System Greetings
Top-Down Call Forwarding	Conference (Multi-Way Calling)	Name Recording
Call Interval Setting	Direct Inward Dialing	Voicemail Forwarding
Call Hold	Max Session Setting	Voicemail Notification (email)
Call Hunting	Music-On-Hold	Voicemail Waiting Indication
Call Recording	Mute	
*Available for Standard Edition Only		

OnDO PBX FAMILY PRODUCTS

	Edition	Registered User*	Voicemail Inbox	 Concurrent Sessions**
٦	Standard Ed.	Unlimited	Unlimited	Unlimited
	SmallOffice Ed.	40	40	12 sessions

^{*}The number of users enabled for all PBX features.

SPECIFICATIONS

◆ Call Control Protocol SIP (RFC3261 Standard)

Voice Decoding Method G.711 U-LAW, G-711 A-LAW, iLBC

◆ DTMF Generator / Decoder · · · · Inbound, RFC2833, INFO

OPERATING ENVIRONMENT

Operating System Supported Microsoft Windows XP/2000, Red Hat Linux v.8.x/9.x

 ◆ Java™
 JDK 1.4 or later

 ◆ Memory
 512 MB minimum

Bundled Software with OnDO PBX

OnDO SIPServer SIP Proxy Server, SIP Registrar Server

Multi-Platform

OnDO SIP Server is designed to run on any OS that can run Apache Tomcat (v.4.1.2 or later). Supported OS: Microsoft Windows, Linux, Mac OS X, and Solaris.

Complete NAT traversal

With OnDO SIP Server NAT Traversal solution, you don't need to worry about call failures caused by NAT traversal issues on VoIP calls.

Upper/Thru Registration

This unique feature of OnDO SIP Server allows easy configuration of parallel usage of preexisting or another SIP servers. With this feature, users can take advantage of SIP communication through ITSP lines or third party SIP Servers.

Authentication

By specifying authentication settings on REGISTER or INVITE requests, you can limit calls that go through OnDO SIP Server.

Session Management

You can check the real-time status or terminate the active calls.



Brekeke Software, Inc. 4 West 4th Avenue, Suite 501 San Mateo, CA 94402-1614 http://www.brekeke.com/ For more information: email:sales@brekeke.com tel: 650.401.6636

sip: info@sip.brekeke.com

^{**}The maximum number of concurrent sessions is determined by the specifications of the server and the installed environment.

^{*}Red Hat Linux users need to install Apache Tomcat 4.1.2 or later.