INTERRA

INTERRA SIP SERVER INSTRUCTION MANUAL







OVERVIEW

This guide includes required information for installation and making the settings of Sip Server product. Please read this manual carefully before using the program.

In case those information are ignored, please do not forget that Yonnet Bilisim shall not accept any reliability, and the device shall not be covered by the warranty.

Please keep your User Manual since it is an crucial source for the reliably and securely use of your device.



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1 USER LOGIN

It is enough to write server ip in order to login Interra Sip Server. When Interra Sip Server interface is appeared, a user name and password is requested from you.

Default values:

Username : admin

Password : yonnet



Figure 1 Interra Sip Server User Login



2 HOME PAGE- WELCOME SCREEN

You will be directed to System Status screen after login with username and password. You can see logged in users in sip server and their status in this screen.

(*) Asterisk Configuration	×		4			
	92.168.1.69:8088/asterisk/st	atic/config/index.htm	nl			◎馬☆ 🔽 ■ Ξ
	Tashing Tashi 💦 🗂 ayahasin 🔗 i	Calaaa 🍋 Vaaralaa I	 7 Faaliah - 🗁 Aadaai		UD & Marriel 🙃 CCU	
en oygulamalar Mar P90X	Turkiye Turki 🔄 mobolix 🂋	oelecegi Yazanlar 🕒	_ English Androi	ia 🦲 Obuntu 🔄 P	HP & Wysqi 🔄 35H Vs	» 🦲 Diger yer işaretleri
	INT	ERRA SIP SI	ERVER			Network Logout
III System Status	System Status 🏶					
Please click on a panel to						
manage related features	- Trunke th			[-1	Conference Booms	[-1
	Statuc Trunk Tyr		Port/U	lostnamo/IP		
		osemanie	POIN	iosulaine/ir	🖌 Parking Lot	[-]
	* Extensions			[-]	Caller ID Channel E	xtension Timeout
	All Analog Features IAX SIP	•	Free 💛 Ringing 🛡 B	usy 🌑 UnAvailable	No Parked Calls	
	Extension	Name/Label	Status	Туре)
:: Trunks	• 1000	1000	Messages : 0/0	SIP User	★ System Info	[-]
Outgoing Calling Buloc	01001	1001	Messages : 0/0	SIP User	General Network Memory Di	sk
	1002	1002	Messages : 0/0	SIP User	Hostname:	
B Dial Plans	0004	1004	Messages : 0/0	SIP User	13-c753	
## Users	005	1005	Messages : 0/0	SIP User		
# Ring Groups	1006	1006	Messages : 0/0	SIP User	OS Version:	
Time Intervals	• 1007	1007	Messages : 0/0	SIP User	Linux 13-c753 3.4.79-sertac+ #3	3 SMP PREEMPT Sun May 4
	000	1008	Messages : 0/0	SIP User	23:45:43 EESI 2014 armv/1 GNU/1	linux
## incoming Calling Rules	1010	1010	Messages : 0/0	SIP User	Asterisk Build:	
# Paging/Intercom	1011	1011	Messages : 0/0	SIP User	Asterisk/1.8.13.1~dfsg1-3+deb7u	13
ss Follow Me	0 1012	1012	Messages : 0/0	SIP User	Asterisk GUI-version : SVN-bran	nch-2.0-r5220
:: Call Features	1013	1013	Messages : 0/0	SIP User		
88 Sustam Info	1014	1014	Messages : 0/0	SIP User	Server Date & Timezone	
	• 10800	10800	Messages : 0/0	SIP User	Fri Dec 19 14:00:55 EET 2014	
## Backup		200		Ring Group Ring Group	Uptime:	
BB Options	*No Extension assigned	Check Voicemails		VoiceMailMain	14:00:55 up 24 days, 21:12, 2 u	isers,
## Asterisk Logs	- *No Extension assigned	Dial by Names		Directory	Load Average: 1.09, 1.09, 1.08	
	🜟 Queues 🗄			[-]		
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		G	or-version : Sym-pranch-2	0.040220		

Figure 2 Home Page



3 NETWORK

You will see NETWORK button enabling you to make network settings, on upper right corner of Home Page. You can change ip address, net mask, gateway and dns settings of sip server from network screen. Device should be restarted after modifications are performed.

192.168.1.69/network.php ×		
← → C ⋒ 🗅 192.168.1.69/network.	hp	ର୍ഹ୍ଥ 🔀 🔳 ≡
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Network	etatir 🔻	
Ip Address	192.168.1.69	
Netmask	255.255.255.0	
Gateway	192.168.1.1	
Dns	192.168.1.1	
	Update	
	Back	
4		•

Figure 3 Network Settings



4 USERS

When you enter Users from left menu, you will see previously created users. You can add new users, edit or delete existing users in this screen.

A Manage Lines Extensions											x
Manage Oser Extensions											
← → C ff 🗋 19	92.168.1.69:8	3088/asteri	sk/static/config/inde	x.html					Θ	l 🖪 😭 🔀 🔳	Ξ
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										Network Law	
		I	NTERRA SIF	' SE	RVE	R				Network Log	Jul
# System Status	User Extensions	on PBX 🌵									L.
# Trunks	+ Create Ne	w User Modif	/ Selected Users X Delete	Selected	Users	L	ist of User I	Extensions		Where to Buy	
Bio Outgoing Calling Rules		Extension	Full Name	Port	SIP	IAX	DialPlan	OutBound CID			
	0	1000	1000		Yes		DialPlan1	1000	Edit	🗴 Delete	
ss Users		1001	1001		Yes		DialPlan1	1001	Edit	🗴 Delete	
Users is a shortcut for quickly adding and	0	1002	1002		Yes		DialPlan1	1002	Edit	🗴 Delete	
removing all the necessary		1003	1003		Yes		DialPlan1	1003	Edit	🗴 Delete	
configuration components	0	1004	1004		Yes		DialPlan1	1004	Edit	🗴 Delete	
for any new phone.		1005	1005		Yes		DialPlan1	1005	Edit	🗴 Delete	
	0	1006	1006		Yes		DialPlan1	1006	Edit	🗴 Delete	
		1007	1007	-	Yes		DialPlan1	1007	Edit	🗴 Delete	
an Dine Orauna	0	1008	1008		Yes		DialPlan1	1008	Edit	🗴 Delete	
## Ring Groups		1009	1009		Yes		DialPlan1	1009	Edit	🗴 Delete	
## Time Intervals		1010	1010		Yes		DialPlan1	1010	Edit	🗴 Delete	
# Incoming Calling Rules		1011	1011		Yes		DialPlan1	1011	Edit	🗴 Delete	
## Paging/Intercom		1012	1012		Yes		DialPlan1	1012	Edit	🗴 Delete	
# Follow Me		1013	1013		Yes		DialPlan1	1013	Edit	🗶 Delete	
# Call Features		1014	1014		Yes		DialPlan1	1014	Edit	🗶 Delete	
# System Info		10800	10800		Yes		DialPlan1	10800	Edit	🗴 Delete	
# Backup											
# Options											
# Asterisk Logs											
											-
		Copyright 200	6-2011 Digium, Inc. Digium an	d Asterisk GUI-	k are regis version : S	tered <u>trademar</u> VN-branch-2.0	of Digium, Inc. -r5220	All Rights Reserved. Legal Information			

Figure 4 User Screen

Add New User page is displayed with Create New User button. You will have to fill the fields specified below.

Extension: User ID

CallerId Name : User name

CallerId No: User no

SIP : Active/Inactive status of SIP Protocol Make it active.

IAX : Active/Inactive status of IAX Protocol Make it inactive.

Codec Preference : Displays active codecs.

MAC Address : Mac adress of user. User no could be entered.

SIP/IAX Password : User's password.

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Manage User Extensions ×			
← → C ㎡ 🗋 192.16	8.1.69:8088/asterisk/static/config/index.html		Q 55 ☆ ■ ■
🔢 Uygulamalar 🔛 P90X Türkiye	Türki 🗀 mobotix 🤣 Geleceği Yazanlar 🗀 English 🦳 Android 🦳 Ubuntu 📄 PHP & Mysql 🧰 SSH vs 🧧 3:	S Surveillance IP Ca 📋 indir	» 🧰 Diğer yer işaretleri
	INTERRA SIP SERVER		Network
# System Status User Ex	xtensions on PBX 🗄		
# Trunks	Create Net Create NeurIlleer	v .	Where to Buy
Se Outgoing Calling Rules	Create new User		
# Dial Plans	General:		Edit V Delete
and Users	Extension: 6000 U CallerID Name: UalPlan: DialPlan: DialPlant V ()	001	Edit X Delete
Users is a shortcut for	Internal CallerID:6000 ① CallerID Number: ①	002	Edit X Delete
quickly adding and		003	Edit X Delete
configuration components	Enable Voicemail for this User	004	Edit 🗴 Delete
for any new phone.	VoiceMail Access PIN code: Email Address:	005	Edit 🗴 Delete
	- Technology	D06	Edit 🗴 Delete
	T CID D BIAY D Apples Challen: U D Back D: 770 and ach D: 4000	007	Edit 🗴 Delete
# Ring Groups	Codec Preference : First July X Second Com X Third Mass X Fourth Mass X Fifth Mass X	008	Edit 🗴 Delete
•• Time Intervals		009	Edit X Delete
as Incoming Calling Bulos	VoIP Settings	010	Edit 🗴 Delete
	MAC Address : 🕕 🛈 Line Number : 1 🔻 🛈 LineKeys: 1 🔻 🛈	D11	Edit S Delete
E Paging/intercom	SIP/IAX Password: IAX: Require Call Token:	012	Edit Schelete
III Follow Me	IAX: Max Call Numbers:	013	Edit X Delete
SE Call Features	NAT: Can Reinvite: D DTME Mode: process V () inserure:	914	Edit X Delete
## System Info		1000	Luit > Delete
88 Backup	- Other Options		
an Options	💿 3-Way Calling (analog) 🛈 💿 In Directory 🛈 💿 Call Waiting (analog) 🛈		
## Asterisk Logs	ADA User 🛈 💿 Is Agent 🛈 Pickup Group: 1 🔻		
SE Call Detail Records beta			
₿ Active Channels beta	S Cancel Update		
88 Bulk Add beta			
ss File Editor			
# Asterisk CLI			
# IAX Settings			
SIP Settings			
	Copyright 2008-2011 Digium, Inc. Digium and Asterisk are registered <u>trademarks</u> of Digium, Inc. All Rights Reserved. <u>Legal I</u> GUI-version : SVN-branch-2.0-r5220	information	

Figure 5 New User

An Example of User Settings:

Manage User Extensions		
← → C ff 🗋 192	2.168.1.69:8088/asterisk/static/config/index.html	ର୍ 🖬 😭 🗾 🔳
🔢 Uygulamalar 🔛 P90X Tü	irkiye Türki 🗀 mobotix 🤌 Geleceği Yazanlar 🗀 English 🦳 Android 🦳 Ubuntu 🦳 PHP & Mysql 🦳 SSH vs	» 🧀 Diğer yer işaretleri
	Edit User Extension !	Network Logout
Strut Bultings Automation System		
at Outer Oter	INTERRA SIP SERVER	Where to Buy
System Status Trunks	Edit User Extension - 1000 Advanced Edit X	
Puttong Calling Rules Dial Plans Users	General :	Edit X Delete Edit X Delete Edit X Delete
Users is a shortcut for quickly adding and removing all the necessary configuration components	Enable Voicemail for this User VoiceMail Access PIN code: D Email Address:	Edit X Delete Edit X Delete Edit X Delete
for any new phone.	Technology Image: Sign I	Edit X Delete Edit X Delete Edit X Delete
## Ring Groups ## Time Intervals ## Incoming Calling Rules	VolP Settings MAC Address : 1000 SIP/IAX Password: 1000 ID	Edit X Delete Edit X Delete Edit X Delete Edit X Delete
Paging/Intercom Follow Me Call Features	IAX: Max Call Numbers: ① NAT: ② ① Can Reinvite: ③ ① DTMF Mode; RFC2833 ▼ ① insecure: no ▼ ①	Edit X Delete Edit X Delete
# System Info # Backup # Options	Outer Opurits 3-Way Calling (analog) ① In Directory ① Call Walting (analog) ① ADA User ① Is Agent ① Pickup Group: 1 ▼	
## Asterisk Logs	Cancel I Update	
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Figure 6 Sample User



When extra settings should be needed about users, those settings shall be performed by entering Advanced Edit in user edit screen.

Manage User Extensions ×								
← → C ㎡ 🗋 192.168.1.	69:8088/asteris	k/static/config/index	html				(2. 品 ☆ 🔀 🔳 🔳
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	-	NTERRA SIP	SER\	/ER		_ ,,_		Network Logout
# System Status	e Ne Edit User 100	0 Advanced				x		Where to Buy
Trunks Qutgoing Calling Rules Dial Plans Users Users is a shortcut for quickly adding and removing all the necessary configuration components for any new phone. definition the state of the state	1012	fullname=1000 registersip=no host=dynamic caligroup=1 qualify=yes mailbox=1000 transfer=yes context=0LPN_Dial#Ian1 collounteryes context=0LPN_Dial#Ian1 collounteryes context=000 hasvoicemaileno vmsecret=	Cancel	es	DialPlan1	1012		X Delete X Delete
# Paging/Intercom	1014	1014	-)	'es	DialPlan1	1014	4 Edit	X Delete
## Follow Me	10800	10800)	'es	DialPlan1	1080	0 Edit	X Delete
e: Call Features								
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Figure 7 User Details Screen



5 RING GROUP

It is used to make a certain ring group or a ringing order between the users created by you.

Manage RingGroups	×			
← → C ♠ D:	192.168.1.69:8088/asterisk/stat	ic/config/index.html		Q 🖬 ☆ 🔀 🔳 🔳
👯 Uygulamalar 🔤 P90X	(Türkiye Türki 📋 mobotix 🤗 Ge	leceği Yazanlar 📋 English 📋 /	Android 🛅 Ubuntu 🛅 PHP & Mysql 🛅 SSH vs	» 🗋 Diğer yer işaretleri
	INTE	RRA SIP SERVER		Network Logout
System Status	Manage RingGroups 4			
# Trunks	New RingGroup	Man	age RingGroups	
St Outgoing Calling Rules	Extension	Ring Group	Members	
	200	200	1001(SIP) 1001	Edit K Delete
a Dine Creune	201	201	1002(SIP) 1002	Edit X Delete
Dense Ringsprouse is data more han one extension imultaneously, or to may more han one phone sequentially. This feature may also be called Hungsroups. It Time Intervals It Insering Calling Rules It Paging/Intercom It Paging/Intercom It Paging/Intercom It Paging/Intercom It System Into It System Into It Caller System Into It Callerons It Asterisk Logs				
	Copyright 2008-2011 D	igium, Inc. Digium and Asterisk are registered GUI-version : SVN-b	trademarks of Digium, Inc. All Rights Reserved. <u>Legal Information</u> ranch-2.0-r5220	

Figure 8 Ring Group Creation

You will be requested to enter Ring Group Name and user id of this group when you create a new group here. Uers assigned to Ring Group Members by you shall be included in the group.

Strategy: You can determine the users to be rung in turn or simultaneously.

Seconds to ring each member: You can determine the duration to ring users in seconds.

If not answered Goto: You can determine the next action if not answered

It will be enough to ring group user ID that you gave from the user (IP Phone, Intercom) which will ring later.

(R) Manage RingGroups	×	
	2.168.1.69:8088/asterisk/static/config/index.html	요 馬 중 🔀 🔳 🚍
🙂 Uvgulamalar 🔤 P90X Ti	irkiye Türki 🗋 mobotix 🔗 Geleceği Yazanlar I 🦳 English 🦳 Android 🦳 Ubuntu 💭 PHP & Mysol 🦳 SSH vs	» Diğer ver isaretleri
	Edit RingGroup !	
YONNET		Network Logout
Smart Byrldinga Aylomation System	INTERRA SIP SERVER	
System Status Trunks	Innage RingGroups	
Soutgoing Calling Rules	Edit RingGroup - 200 X	
# Dial Plans # Users	RingGroup Name : 200 Extension for this ring group : 200	Edit X Delete
## Ring Groups	Ring Group Members Available Users	
Define Ringgroups to dial more than one extension simultaneously, or to ring more than one phone sequentially. This feature may also be called Hungroups. It Time Intervals It Incoming Calling Rules It Raging/Intercom It Call Features It Call Features	1001(SiP) 1001 Image: Simple state st	
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Figure 9 Ring Group Settings



6 DIAL PLAN

Dial Plan allows you to perform permission and authorization transactions among users.

(8) Manage Dialplans		
	92.168.1.69:8088/asterisk/static/config/index.html	ସ୍≣ ☆ 🔀 🔳 🔳
👯 Uygulamalar 🔛 P90X	Türkiye Türki 📋 mobotix 🤣 Geleceği Yazanlar 📋 English 📋 Android 📋 Ubuntu 📋 PHP & Mysql 📋 SSH vs	» 📋 Diğer yer işaretleri
	INTERRA SIP SERVER	Network Logout
# System Status # Trunks # Outgoing Calling Rules	DialPlans Manage DialPlans	
# Dial Plans A DialPlan is a set of 'Calling Rules' that can be assigned to one or more	A Dial Plan is a collection of Outgoing Call Rules . Dial Plans are assigned to Users to specify the dialing permissions they have. For example, you for local calling that only permits users of that Dial Plan to dial local numbers, via the "local" outgoing calling rule. Another user may be permitte numbers, and so would have a Dial Plan that includes both the "local" and "longdistance" outgoing calling rules.	might have one Dial Plan ed to dial long distance
users.	Default Dial Plan Calling Rules Image: Constraint of the state of th	Options sion Edit X Delete sion Edit X Delete
# Users # Ring Groups		
Incoming Calling Rules Paging/Intercom		
# Follow Me # Call Features # System Info		
# Backup # Options # Asterisk Logs		
	Copyright 2006-2011 Digium, Inc. Digium and Asterisk are registered <u>trademarks</u> of Digium, Inc. All Rights Reserved. <u>Legal Information</u> GUI-version : SVN-branch-2,0+5220	

Figure 10 Dial Plan



7 COMMAND SCREEN

You can see user IDs, IP addresses and their registration status of users which have been registered on sip server, by giving commands on command screen. You should follow **Options -> Advanced Options -> Show Advanced Options** order in order to be able to open command screen. Afterwards, **Asterisk CLI** menu shall be appeared.

() Show/Hide Advanced opt						
← → C ♠ 192,168,1,69:8088/	asterisk/static/config/index.html	Q 55 1 1 = =				
🗰 Uvgulamalar 🔤 P90X Türkiye I Türki 🧰 m	iobativ. 🔗 Geleceăi Vazaplar I 🦳 English. 🦳 Android. 🦳 Ubuntu. 🦳 PHP & Mysol. 🦳 SSH ys	» Dičer ver isaretleri				
	INTERRA SIP SERVER	Network Logout				
## System Status Advanced Options						
## Trunks						
Outgoing Calling Rules	General Preferences Language Change Password Reboot Advanced Options					
88 Dial Plans	A transmitter of Operations					
## Users	Advanced Options					
88 Ring Groups						
ss Time Intervals	Clicking the Snow Advanced Options button below provides the additional menu items on the left hand sidebar					
B Incoming Calling Rules	Notice! Digium does not provide support for the options configurable in the Advanced menu items. Digium does not					
St Paging/intercom	provide support for bugs uncovered in the Advanced menu items. If your unit becomes inoperable due to editing of the Advanced menu items, Diglum Technical Support will request that you reset your unit toFactory Default					
ss Coll Features	configuration. Continue at your own risk.					
•• System Info						
## Backup	Show Advanced Options					
and the second s						
Admin Settings.						
88 Asterisk Logs						
Canad	inkt 2008-2011 Dinium Inc. Dinium and Astarisk are renistered trademarks of Dinium. Inc. All Richts Researed. Land Information					
Сорун	GUI-version : SVN-branch-2.0-5220					
192.168.1.69:8088/asterisk/static/config/flipadvanced.	html					

Figure 11 Command Screen

Command screen shall request a command from you.

Sip show pers is the command you will need most. You can follow the status of registered users by this command.

'INTERRA

() Asterisk CLI Emulator	×		plan, plan hape	-	-				ing and the	-	
← → C ↑ D 192.168.1.69:8088/asterisk/static/config/index.html										옥 🖩 ☆ 🔀 🔳 🔳	
🔢 Uygulamalar 🔛 P90X	Türkiye Türki	🗀 mobotix	🔗 Geleceği Yazanlar	🗀 English	🗀 Andr	oid 📋 Ubuntu	📋 PHP & Mysql	📋 SSH vs	S 3S Surveillance IP Ca	indir 🗀	» 📋 Diğer yer işaretleri
	INTERRA SIP SERVER										Network
## System Status	Asterisk CL	sip show pe	ers								
St Trunks	Command>sip sh	ow peers									
B Dial Plane	Name/username				Dyn Forc	ernort ACL Port	Status				
	1000/1000	19	92.168.1.202		D N	5060	OK (7 ms)				
# Users	1001/1001	19	92.168.1.65		D N	3294	5 OK (1 ms)				
## Ring Groups	1002/1002	19	92.168.1.119		D N	4765	2 OK (1 ms)				
88 Time Intervals	1003/1003	((Inspecified)		D N	0 A	UNKNOWN				
# Incoming Calling Rules	1005/1005	19	92.168.1.207		D N	4559	OK (17 ms)				
as Paging/Intercom	1006/1006	19	92.168.1.138		D N	4358	5 OK (321 ms)				
aa Faying/intercom	1007/1007	19	92.168.1.140		D N	5066	4 OK (2 ms)				
SE Follow Me	1008/1008	19	32.168.1.138 Inspecified)		D N	4136	UNKNOWN				
# Call Features	1010/1010	19	92.168.1.67		D N	4151	2 OK (34 ms)				
:: System Info	1011/1011	(1	Unspecified)		D N	e	UNKNOWN				
te Backun	1012/1012	19	92.168.1.171		D N	4345	3 OK (67 ms)				
as Ontiono	1013/1013	((Inspecified)		D N	0	UNKNOWN				
	10800/10800	(1	Unspecified)		D N	ĕ	UNKNOWN				
## Asterisk Logs	16 sip peers [M	onitored: 9 or	nline, 7 offline Unmonito	ored: 0 online	e, 0 offli	ne]					
88 Call Detail Records beta											
# Active Channels beta											
📽 Bulk Add ^{beta}											
## File Editor											
## Asterisk CLI											
Asterisk Command Line											
Interface											
*											
## IAX Settings											
:: SIP Settings											
			Copyright 2006-2011 Digium,	Inc. Digium and	Asterisk are	registered tradema	ks of Digium, Inc. All Ri	ights Reserved	. Legal Information		

Figure 12 Command Entry