

# Grandstream Networks, Inc.

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UCM series IP PBX

## Security Manual



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## OVERVIEW

This document presents a summary of security measures, factors, and configurations that users are recommended to consider when deploying the UCM.

**Note:** We recommend using firmware 1.0.14.24 or higher for improved security.

The following sections are covered in this document:

- **Web UI Access**

Web UI access is protected by username/password and login timeout. Two-level user management is configurable. Admin with limited access can be created by the default super administrator.

- **Extension Security**

Extension security utilizes SIP/IAX passwords for authentication, IP address whitelisting, and SRTP encryption.

- **Trunk Security**

Trunk security utilizes privilege levels and source caller ID filters to prevent outbound calls from unintended sources.

- **TLS Protocol**

TLS is utilized to encrypt SIP signaling.

- **Firewall Features**

Three different security measures can be configured to protect the UCM against malicious attacks: Static Defense, Dynamic Defense (UCM6102/6202/6204/6208/6510 only) and Fail2ban.

- **AMI**

AMI feature is used to let the user connect to Asterisk instance to read and track the state of telephony client, it may come with security concerns for UCM administrators that needs to be considered.



## Security Bulletins

- Potential Vulnerability Associated With Use of Allow Guest Calls Option [Grandstream Security Bulletin GS13-UCM001](#)
- Potential Vulnerability Associated With Misuse of Dial Trunk Option in IVR [Grandstream Security Bulletin GS13-UCM002](#)
- Security Vulnerability Associated With Returned Cookie from WebUI Login Session [Grandstream Security Bulletin GS17-UCM003](#)

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## WEB UI ACCESS

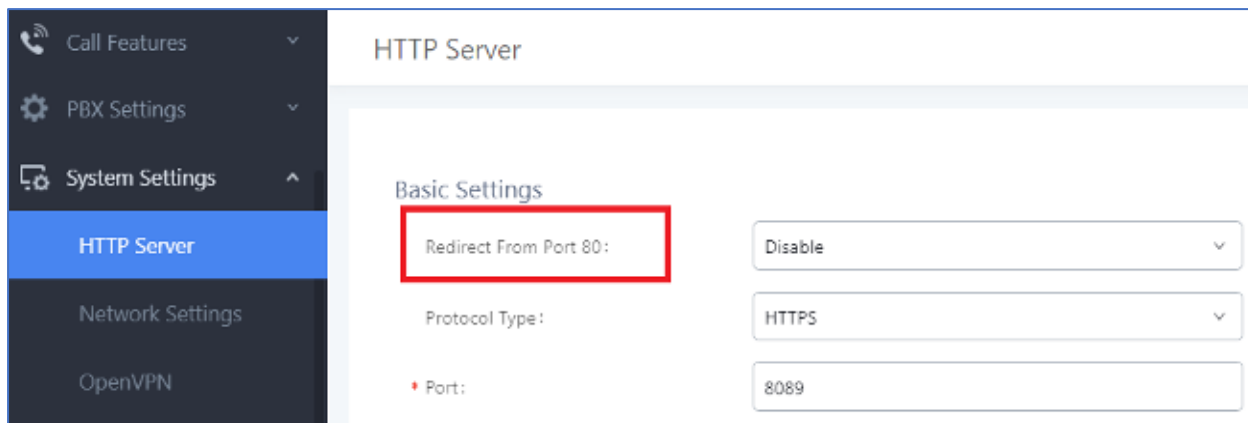
### UCM HTTP Server Access

The UCM embedded web server responds to HTTP/HTTPS GET/POST requests. Embedded HTML pages allow users to configure the device through a web browser such as Microsoft IE, Mozilla Firefox, Google Chrome, etc. With this, administrators can access and configure all available UCM information and settings. It is critical to understand the security risks involved when placing the UCM on public networks.

### Protocol Type

HTTP and HTTPS web access are supported to access the UCM web UI and can be configured under web UI→Settings→HTTP Server. The selected protocol type will also be the one used for Zero Config when configs are pushed to endpoint devices. To secure transactions and prevent unauthorized access, it is highly recommended to:

1. Use HTTPS instead of HTTP,
2. Disable option Redirect from Port 80 and,
3. Avoid using well known port numbers such as 80 and 443.

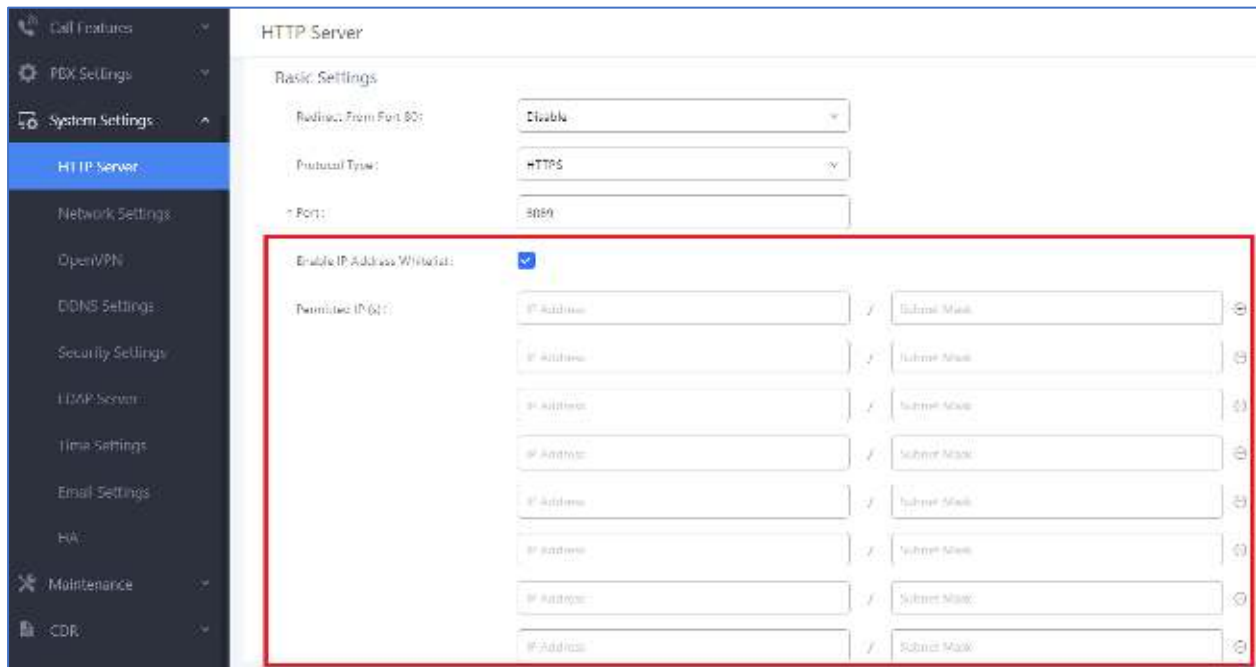


Finally, users have the option to specify a list of up to 10 IP addresses which will be allowed to access the UCM web UI. Addresses not listed will be restricted from accessing the UCM.

To enable and add to the IP address whitelist, navigate to System Settings→HTTP Server:

- Check the option “Enable IP Address Whitelist”
- Enter the permitted IP addresses along with the subnet masks.





**HTTP Server**

Basic Settings

Redirect From Port 80:

Protocol Type:

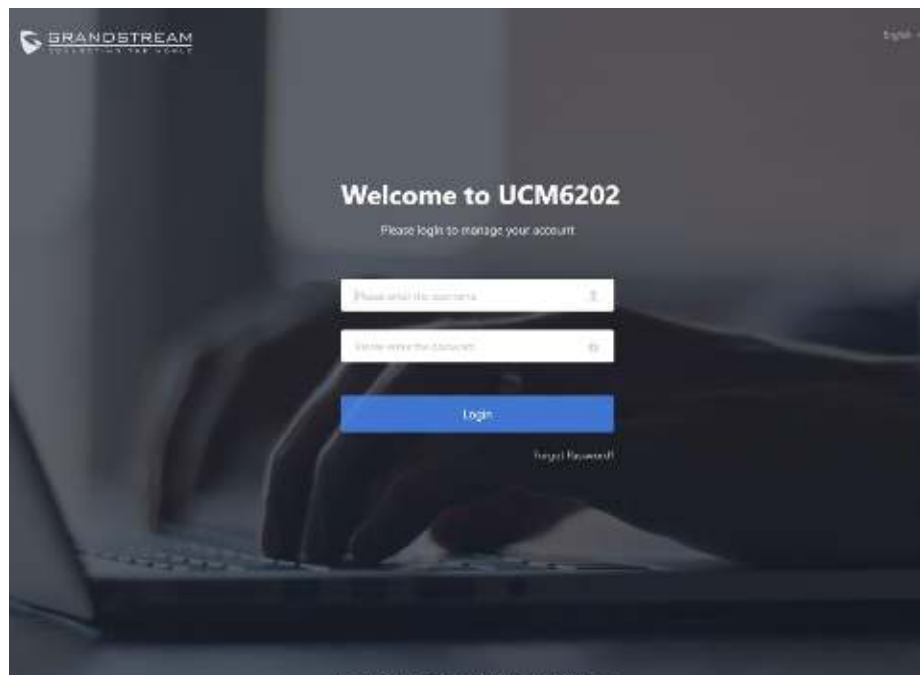
Port:

Enable IP Address Whitelist: ☒

Whitelisted IP(s)	IP Address	Subnet Mask
	<input type="text" value="IP Address"/>	<input type="text" value="Subnet Mask"/>
	<input type="text" value="IP Address"/>	<input type="text" value="Subnet Mask"/>
	<input type="text" value="IP Address"/>	<input type="text" value="Subnet Mask"/>
	<input type="text" value="IP Address"/>	<input type="text" value="Subnet Mask"/>
	<input type="text" value="IP Address"/>	<input type="text" value="Subnet Mask"/>
	<input type="text" value="IP Address"/>	<input type="text" value="Subnet Mask"/>
	<input type="text" value="IP Address"/>	<input type="text" value="Subnet Mask"/>
	<input type="text" value="IP Address"/>	<input type="text" value="Subnet Mask"/>
	<input type="text" value="IP Address"/>	<input type="text" value="Subnet Mask"/>

## User Login

Username and password are required to log into and access the UCM web UI.



**GRANDSTREAM**  
CONNECTING THE WORLD

**Welcome to UCM6202**

Please login to manage your account.

Please enter the username:

Please enter the password:

**Login**

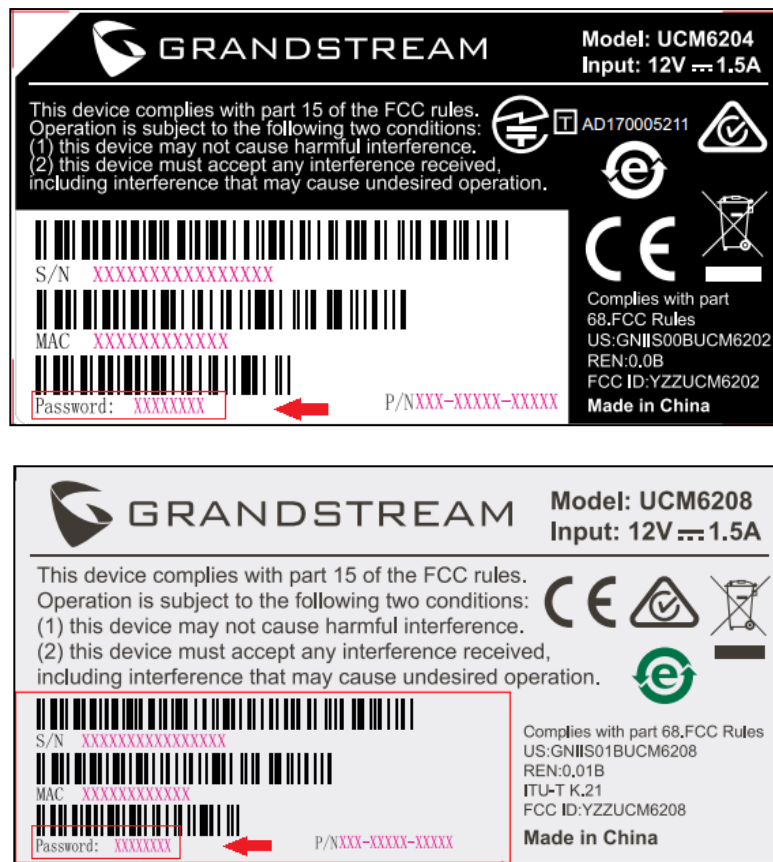
[Forgot Password?](#)

**Figure 1: UCM6202 Web UI Login**

The factory default username is “admin” while the default random password can be found on the sticker at the back of the unit.



**Note:** Units manufactured starting January 2017 have a unique random password printed on the sticker. Older units and UCM6100 series have default password “admin”.



**Figure 2: Default Random Password**

It is highly recommended to change the password after logging in for the first time.

To change the password for default user "admin", navigate to System Settings→Change Information→Change Password/Email. The password length must be between 4-30 characters.

If PBX Settings→General Settings→Enable Strong Password is toggled on, the minimum password requirements are as follows:

- Must contain at least one number.
- Must contain at least one uppercase letter, lower case letter, OR special character.

Strong passwords with a combination of numbers, uppercase letters, lowercase letters, and special characters are always recommended for security.



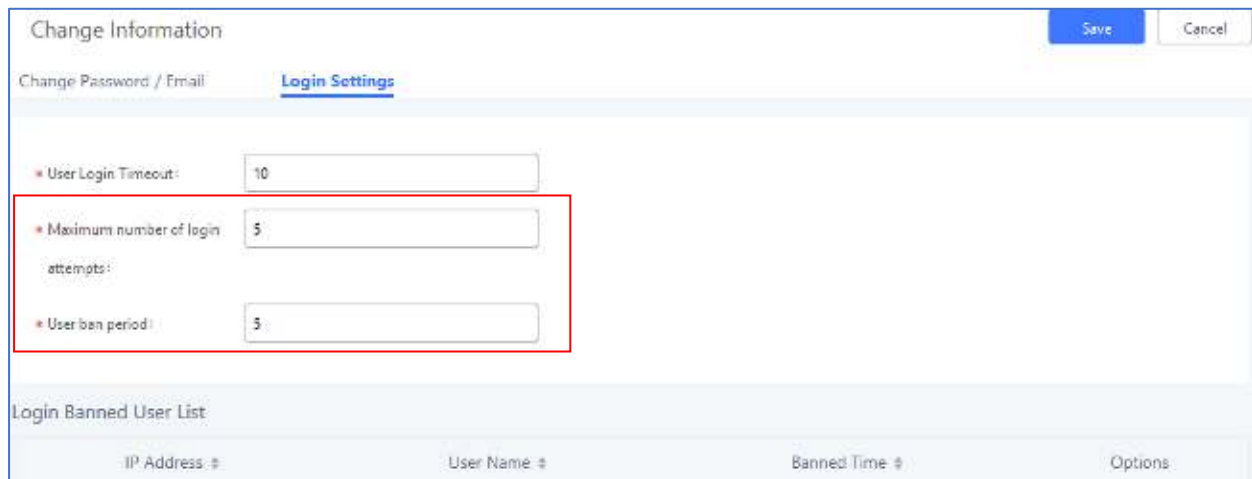


## Login Settings

To further prevent unauthorized access to the UCM web UI, users will automatically be logged out after the configured period of inactivity. Username and password will be required to access the web UI again.

The default login timeout is 10 minutes and can be changed by navigating to Maintenance→Change Information→Login Settings and modifying the User Login Timeout field.

Additionally, the UCM can also ban users after a specified number of failed login attempts for a specified amount of time. By default, users will be banned for 5 minutes after 5 failed login attempts.



Change Information Save Cancel

Change Password / Email [Login Settings](#)

\* User Login Timeout:

\* Maximum number of login attempts:

\* User ban period:

Login Banned User List

IP Address	User Name	Banned Time	Options
------------	-----------	-------------	---------

**Figure 3: Login Settings**



## User Management Levels

Four user privilege levels are currently supported:

- **Super Admin**
- **Admin**
- **Custom level**
- **Consumer**

Super Admin has access to all of the UCM's pages and can execute any operation. Admin can access most of the UCM's pages with the exception of the following:

- Maintenance→Upgrade
- Maintenance→Backup
- Maintenance→System Cleanup/Reset
- Maintenance→Operation Log

A "Super Admin" user with username "admin" is initially configured on the UCM with factory settings. It is the only allowed "Super Admin" account and cannot be deleted and changed. This super administrator could create, edit and delete new user accounts with lower privileges "Admin", "Consumer" and "Custom".

Only Super Admin has the authority to view the activity of all users via the Operation Log.

If there is more than one PBX administrator managing the UCM, it is highly recommended to create Admin level users instead of giving out Super Admin access.

Custom privilege user levels can also be created and modified to have specific permissions including but not limited to:

- Status
- Conference
- System Events
- Feature Codes
- CDR
- PMS-Wakeup Service

To create a new custom privilege level, navigate to Maintenance→User Management→Custom Privilege, name the new custom user level and assign the desired modules as shown in the figure below.



### Create New Custom Privilege

\* Privilege Name:

Reporter

Custom Privilege:

☐ 4 items Available Modules

Search

Q

☐ Status

☐ Conference

☐ CDR API

☐ Wakeup Service

<

>

☐ 2 items Selected Modules

Search

Q

☐ CDR Records

☐ System Events

**Figure 4: Creating Custom Privilege Levels**

"Consumer" level access is the default privilege level assigned to newly created extensions and users. This level does not allow system-wide changes and access to advanced maintenance operations.



## EXTENSION SECURITY

### SIP/IAX Password

When creating a new SIP/IAX extension, the UCM administrator is required to configure “SIP/IAX Password” which will be used for account registration authentication.

If “Enable Random Password” (PBX Settings→General Settings) is enabled, “SIP/IAX Password” is automatically filled with a randomly generated secure password when creating the extension on the UCM. “Passwords must contain 1) at least one number and 2) at least one lowercase letter, uppercase letter, OR special character.”

### Strategy of IP Access Control

The UCM administrator could control what IP address(s) is allowed to register to a certain extension by editing “strategy” option under extension configuration dialog→“Media” tag. **Make sure to configure the “strategy” option to the smallest set to block registration attempts from anyone that doesn’t need to register to the account.**

The strategy options are:

- **“Local Subnet Only”**: allows register requests from local IPs only. By default, the local subnet where the UCM is location is allowed. User could also add more local subnets where devices are allowed to register to this extension.
- **“A Specific IP Address”**: allows register requests from one user specified IP only.
- **“Allow All”**: allows registration requests from anywhere, which is not recommended.

### Example: Local Subnet Only

1. Assuming there are multiple subnets within the office and the devices in all subnets can reach each other, the network administrator would like to allow only devices in 192.168.40.x network to register to this UCM.
2. Under UCM web UI extension dialog, configure “Local Subnet Only” for “Strategy” option and 192.168.40.0 for “Local Subnet”.



Edit Extension: 1001

Save

Cancel

Basic Settings
Media
Features
Specific Time
Follow Me

SIP Settings:

NAT: ☒

\* Can Direct Media: No

\* DTMF Mode: RFC2833

\* TEL URI: Disabled

\* Alert-info: None

\* Fax Mode: None

Tax to Email: Yes

Enable T.38 UDPTL: ☐

SRTP: Disabled

Strategy: Local Subnet Only

\* Local Subnet: 192.168.40.0

Codec Preference:

☐ G.726
☐ G.726-32
☐ ADPCM
☐ G.723
☐ H.263

☐ PCMU
☐ PCMA
☐ GSM
☐ G.726

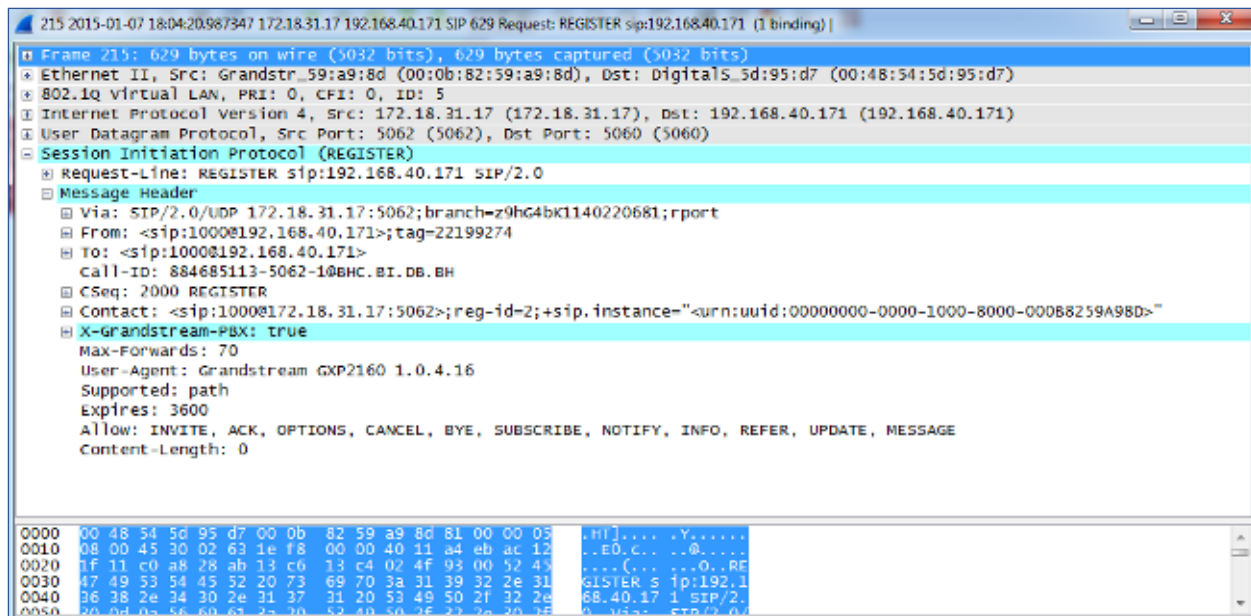
**Figure 5: Strategy – Local Subnet Only**

### 3. Save and Apply changes.

Now if the SIP end device is in the subnet other than 192.168.40.x, e.g., 172.18.31.x subnet, the UCM will not allow registration using this extension. The following figure shows the SIP device IP address is 172.18.31.17. The UCM on IP 192.168.40.171 replies 404 Not Found for the registration request.

No.	Time	Source	Destination	Protocol	Length	Info
213	2015-01-07 18:04:20.987347	172.18.31.17	192.168.40.171	SIP	629	Request: REGISTER sip:192.168.40.171 (1 binding)
216	2015-01-07 18:04:20.989556	192.168.40.171	172.18.31.17	SIP	491	Status: 404 Not Found
304	2015-01-07 18:04:42.194344	172.18.31.17	192.168.40.171	SIP	629	Request: REGISTER sip:192.168.40.171 (1 binding)
336	2015-01-07 18:04:42.196110	192.168.40.171	172.18.31.17	SIP	491	Status: 404 Not Found
341	2015-01-07 18:05:03.242985	172.18.31.17	192.168.40.171	SIP	629	Request: REGISTER sip:192.168.40.171 (1 binding)
342	2015-01-07 18:05:03.244559	192.168.40.171	172.18.31.17	SIP	491	Status: 404 Not Found

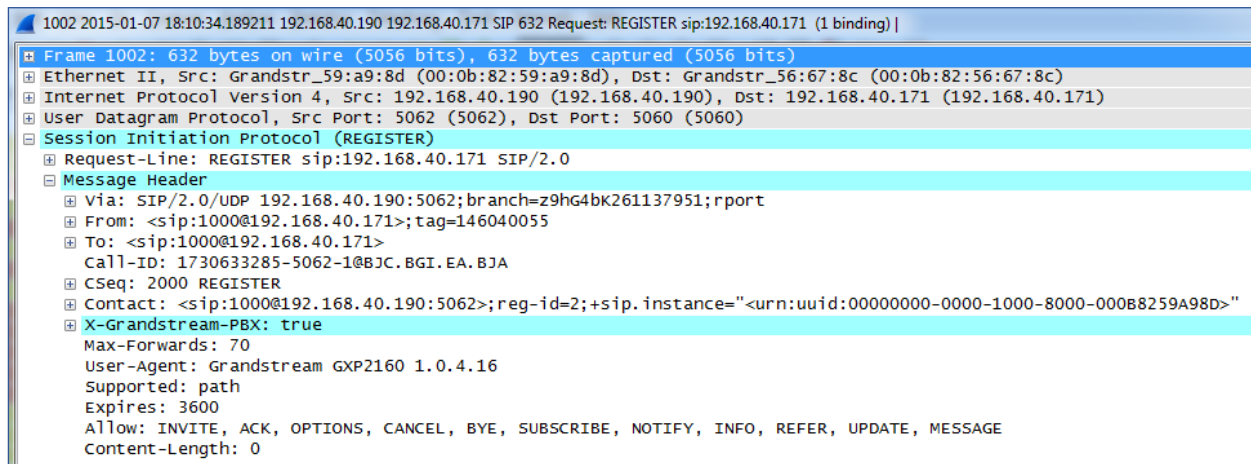




**Figure 6: Registration Failed from Subnet Not Allowed for Registration**

Once the device is moved to the 192.168.40.x subnet, registration will be successful. The following figure shows the IP address for the same SIP endpoint is 192.168.40.190. The UCM on IP address 192.168.40.171 replies 200 OK to the registration request.

1002	2015-01-07 18:10:34.189211	192.168.40.190	192.168.40.171	SIP	632 Request: REGISTER sip:192.168.40.171 (1 binding)
1003	2015-01-07 18:10:34.190632	192.168.40.171	192.168.40.190	SIP	376 Status: 401 Unauthorized
1004	2015-01-07 18:10:34.198412	192.168.40.190	192.168.40.171	SIP	796 Request: REGISTER sip:192.168.40.171 (1 binding)
1005	2015-01-07 18:10:34.206894	192.168.40.171	192.168.40.190	SIP	383 Status: 200 OK (1 binding)



**Figure 7: Registration Successful from Allowed Subnet**

## SRTP

UCM supports SRTP for encrypted audio streams. It is disabled by default. To enable it, navigate to Extensions→Edit Extension→Media and select the desired SRTP option.

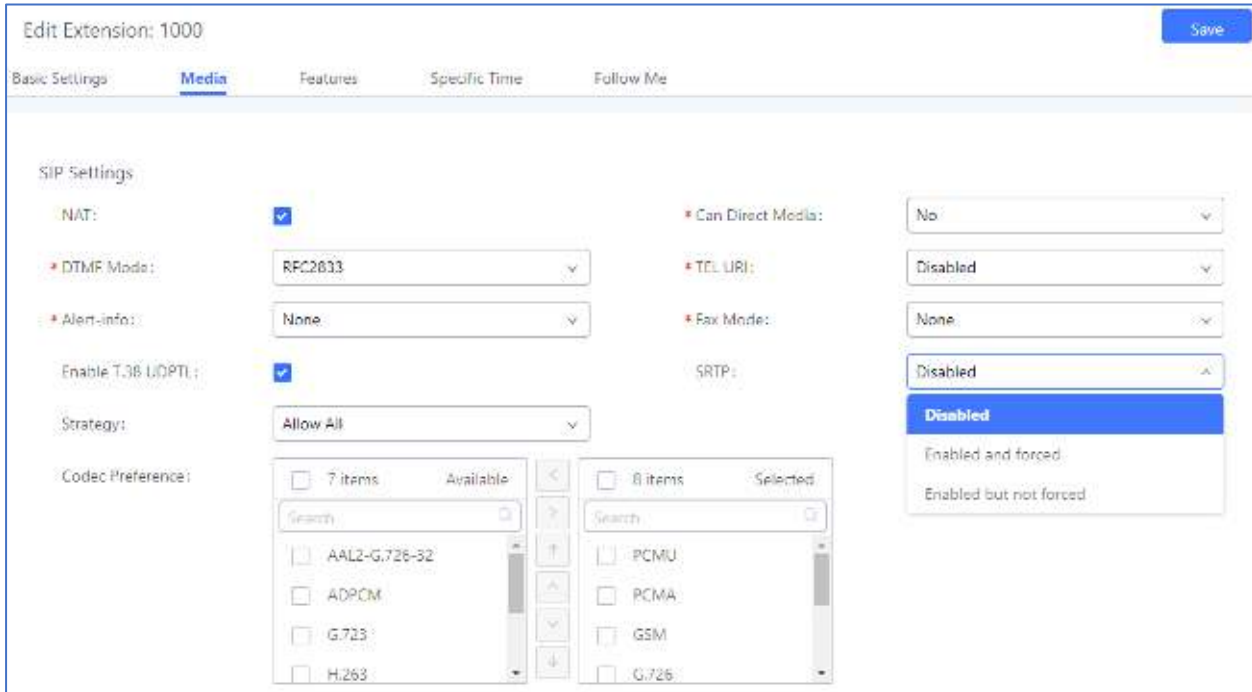


Figure 8 shows the 'Edit Extension: 1000' interface with the 'Media' tab selected. The 'SRTP' dropdown menu is open, displaying three options: 'Disabled' (highlighted in blue), 'Enabled and forced', and 'Enabled but not forced'. Other visible settings include NAT (checked), DTMF Mode (RFC2833), Alert-info (None), Enable T.38 UDP/TLS (checked), Strategy (Allow All), and Codec Preference (7 items available, 8 items selected).

**Figure 8: Enabling SRTP**

As shown above, users have two options while enabling SRTP under extension parameters:

- **Enabled and forced:** Extension will support SRTP and will not allow any calls without it.
- **Enabled but not forced:** Extension will support SRTP and will allow negotiation with calls without SRTP.



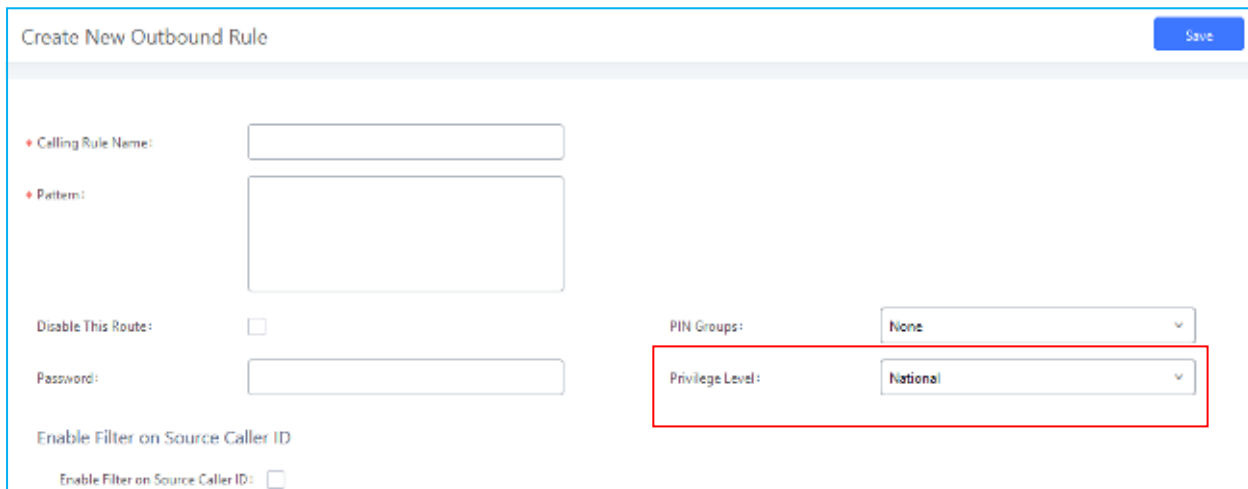
## TRUNK SECURITY

A potential risk for trunks is unauthorized international and long-distance calls, which can result in unexpected high charges before the issue is noticed. Therefore, administrators must take special care when configuring trunks that would be charged for placing certain calls such as long-distance calls and international calls.

### Outbound Rule Permissions

Users can apply one of four security measures on the UCM to control outbound calls.

1. Privilege Level
2. Enable Filter on Source Caller ID
3. Password protection
4. PIN groups



**Figure 9: Outbound Rule Permissions**

### Privilege Level

On the UCM, the supported 4 privilege levels are “Internal”, “Local”, “National” and “International” from the lowest to the highest. Outbound calls through trunk can be placed only if the permission level assigned to the caller is higher or equal to the privilege level of the outbound rule. Outbound call requests from users with privilege lower than the outbound rule will be rejected.

### Source Caller ID Filter

UCM administrators can specify the extensions/extension groups that can use the outbound rule. This can be done by selecting extension/extension groups or defining pattern for the source caller ID in “Custom Dynamic Route” field. The extension allowed to make outbound call will either need to be an extension in the selected list or match the defined pattern.





Enable Filter on Source Caller ID: ☒

Available Extensions/Extension: 

1000 x 1002 x 1003 x

Groups: 

▼ ☒ All

☒ 1000

☐ 1001

☒ 1002

☒ 1003

☐ 1004

☐ 1005

☐ 1006

☐ 1007

Call Duration Limit

Call Duration Limit:

Send This Call Through Trunk

\* Use Trunk:

Strip:

### Figure 10: Source Caller ID Filter

For detailed configuration instructions, please refer to MANAGING OUTBOUND ROUTE section in white paper: [How to manage inbound/outbound route on UCM6xxx](#)

## Password Protection

Administrators can protect outbound rules with passwords that would be requested by the UCM when callers try to use the configured routes.



Create New Outbound Rule

Save

\* Calling Rule Name:

\* Pattern:

Disable This Route:

☐

PIN Groups:

None

Password:

Privilege Level:

National

Enable Filter on Source Caller ID

☐

**Figure 11: Password Protection**

## PIN Groups

In the scenario where multiple users share the same phone, and the phone is used to make outbound calls, outbound routes can be set to require PIN entry to use. To set up a PIN group:

1. Navigate on to Extension/Trunk→Outbound Routes→PIN Groups
2. Click Add to create a new PIN group and enter the user names and passwords.

Create New PIN Group

\* Name:

Store\_Staff

Record in CDR:

☒

Members

+ Add

PIN Number
PIN Number: 9854 PIN Name: John
PIN Number: 7741 PIN Name: Adam
PIN Number: 6535 PIN Name: Jane

**Figure 12: Adding PIN Groups**

3. Save and apply. The PIN Group can now be selected in Outbound Routes. Each time one of the PIN Group members makes an outbound call, he/she will be requested to enter their PIN.



Create New Outbound Rule

Save

\* Calling Rule Name:

\* Pattern:

Disable This Route:

☐

Pin Groups:

None

Password:

Privilege Level:

National

Enable Filter on Source Caller ID

Enable Filter on Source Caller ID:

☐

**Figure 13: Outbound route with PIN group**

## IVR Dial Trunk

When creating/editing an IVR, the administrator can decide whether or not to allow the calls entering the IVR to make outbound calls through trunks by configuring “Dial Trunk” and “Permission”. If “Dial Trunk” option is enabled, the caller calling into the IVR will be able to dial external numbers through a trunk if the IVR’S permission is higher than or equal to the privilege of the trunk. The potential risk here is that unwanted users may call into IVR and then dial external numbers. This could possibly generate unexpected high charges especially if an IVR is configured as the destination of an inbound route of a PSTN trunk, in which case, anyone can call into the IVR and then dial long distance or international calls.

Edit IVR: testIVR

Basic Settings
Key Pressing Events

\* Name:

\* Extension:

Dial Trunk:
☒

\* Permission:

Dial Other Extensions:

☒ Extension
☐ Conference

☐ Call Queue
☐ Ring Group

☐ Paging/Intercom Groups

☐ Voicemail Groups
☐ Fax Extension

☐ Dial By Name

☐ All

**Figure 14: IVR Dial Trunk**

We recommend disabling “Dial Trunk” option unless the risk associated with it is clearly understood or the PBX administrator intentionally configures it to do so for specific reasons.

For more information about IVR permissions, please refer to IVR PERMISSION section in white paper: [How to manage inbound/outbound route on UCM6xxx](#)

## Allow Guest Calls

“Allow Guest Calls” option can be found in PBX Settings→SIP Settings→General page. **It is highly recommended to NOT enable this option for any deployments. This option will stop the UCM from authenticating incoming calls from unknown/anonymous callers. This allows hackers to send unauthorized calls, which may result in high charges.** Administrators should also check CDR regularly to ensure there are suspicious calls in the early stages of deployment.



## TLS

UCM has the option to secure SIP packets with TLS encryption, allowing for safe transactions over untrusted networks and with authenticated parties.

TLS can be configured in PBX Settings→SIP Settings→TCP/TLS.

SIP Settings

General
Misc
Session Timer
TCP/TLS
NAT
ToS
SIP Trunk Prompt Tone

TCP Enable:

☐

TCP Bind IPv4 Address:

TCP Bind IPv6 Address:

TLS Enable:

☐

TLS Bind IPv4 Address:

TLS Bind IPv6 Address:

TLS Do Not Verify:

☒

TLS Self-signed CA:

Delete

TLS Cert:

Delete

TLS Key:

Delete

TLS CA Cert:

**Figure 15: PBX Settings→SIP Settings→TCP/TLS**

1. Tick the "TLS Enable" box to activate TLS.
2. Configure "TLS Do Not Verify", "TLS Self-Signed CA" and "TLS Cert" to achieve basic TLS authentication and encryption.



- **TLS Self-Signed CA**

Used when UCM is acting as the client to authenticate a server. If the server uses a self-signed certificate, that certificate should also be installed on the UCM for successful verification. If it is signed by a CA, then a copy of the CA server certificate should be installed.

- **TLS Cert**

Used when UCM acts as the server. TLS Cert is sent to the client during the TLS handshake and includes the key and server certificate. The "common name" field in the server certificate should match the server host (either by IP address or domain name). This is required if the client is another UCM (not a standard, as some clients do not have this requirement for server authentication). If the common name does not match, authentication on the client side will fail, and the TLS connection will fail to establish.

- **TLS Do Not Verify**

Effective only when the UCM is acting as the client. If enabled, the certificate sent by the server will not be verified. This is useful for authentication between two peered UCMs using the factory certificate, which has "common name" configured with the value "localhost", which is not a valid host and will result in authentication failure. Skipping verification will not have any effect on the encryption itself.

If disabled, the client will verify the server certificate with the "TLS Self-Signed CA".

**Note:** SIP endpoints will still need to be configured to use TLS in order to encrypt SIP messages sent to the UCM.



# FIREWALL

The UCM firewall functionality consists of three components: Static Defense, Dynamic Defense, and Fail2ban. Each can be configured for specific types of malicious attacks.

## Static Defense

Static Defense uses pre-configured filtering rules to protect the UCM. Three filtering rule types are supported: ACCEPT, REJECT, and DROP. These rules can be applied to specific source/destination IP addresses and ports. Static Defense can be configured in System Settings→Security Settings→Static Defense.

When creating or editing rules, the following options are available:

- **Rule Name:** A name to easily identify the rule and its purpose.
- **Action:** The action to perform for this rule. Accept, Reject, or Drop the connection if criteria is met.
- **Type:** The direction of the network traffic. IN for inbound traffic. OUT for outbound traffic.
- **Interface:** Select network interface where the traffic will go through (this option is only available for devices with both WAN and LAN ports).
- **Service:** The specific services to affect. FTP, SSH, Telnet, HTTP, and LDAP are available. Custom allows administrators to specify source/destination IP addresses and ports and the protocol.

Additionally, the following features are also available to further increase security:

### 1. Ping Defense

Protects against ping floods by not responding to ICMP echo requests.

### 2. SYN-Flood Defense

Protects against SYN floods by limiting the amount of SYN requests accepted per second.

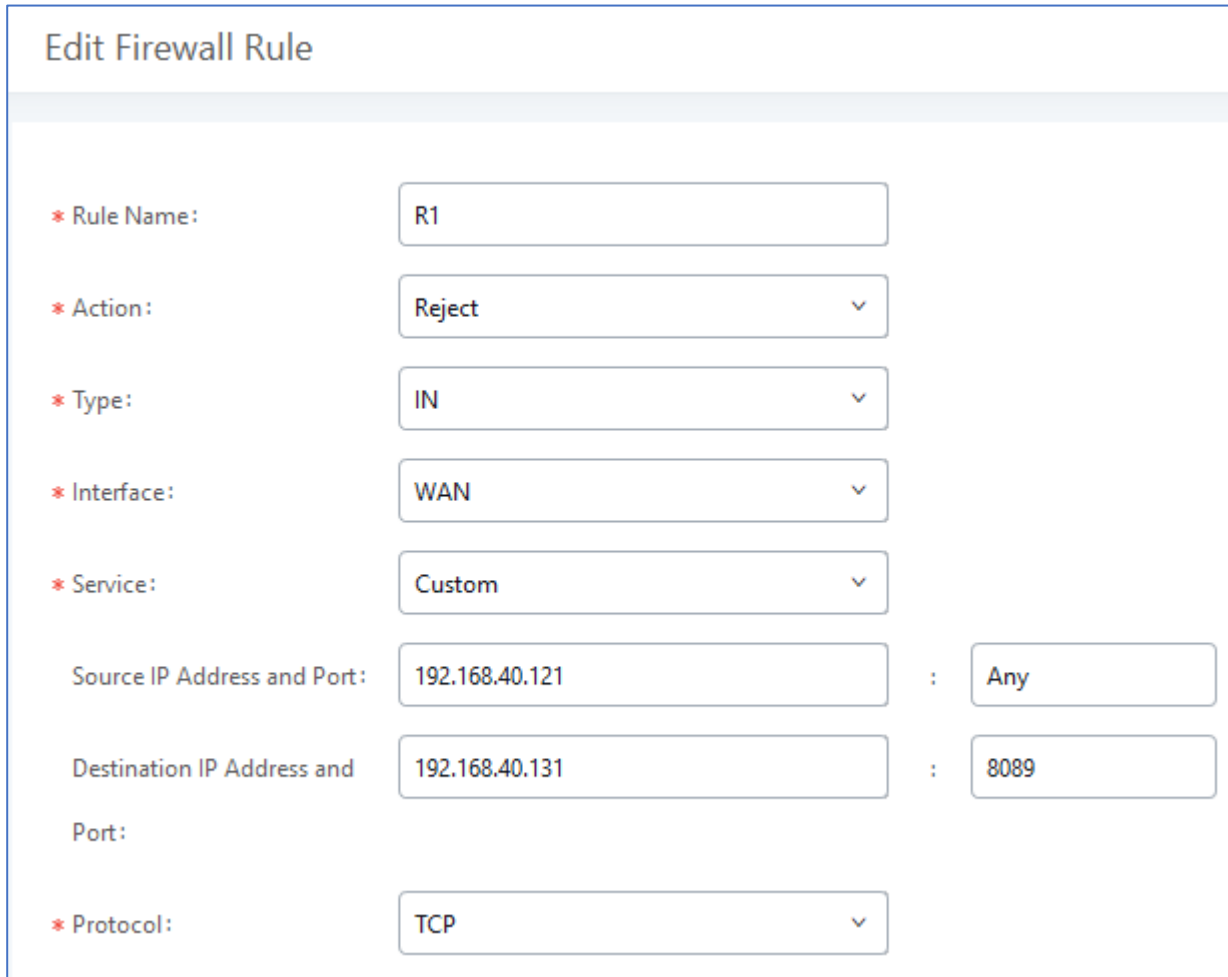
### 3. Ping-of-Death defense

Protects against ping packets larger than 64 KB.



## Static Defense Example: Blocking TCP Connection from a Specific Host

This example demonstrates how to set up a new rule to block a host with a specific IP address to connect to UCM using TCP connection. In the following figure, 192.168.40.121 is the host IP address and 192.168.40.131 is the UCM's IP address. Port 8089 on UCM is used for HTTP server/web UI access. This setting will block host on 192.168.40.131 to access UCM port 8089 using TCP connection.



**Edit Firewall Rule**

\* Rule Name: R1

\* Action: Reject

\* Type: IN

\* Interface: WAN

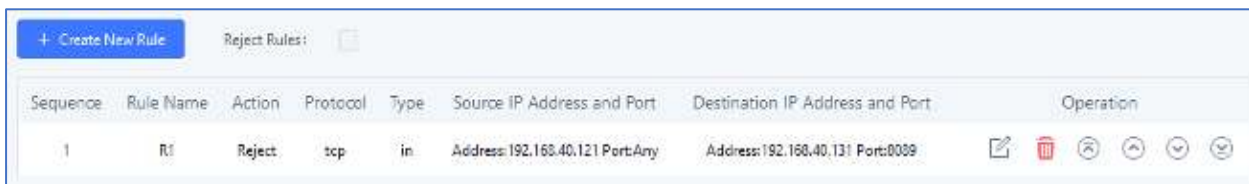
\* Service: Custom

Source IP Address and Port: 192.168.40.121 : Any

Destination IP Address and Port: 192.168.40.131 : 8089

\* Protocol: TCP

Figure 16: Firewall Rule Custom Configuration







Sequence	Rule Name	Action	Protocol	Type	Source IP Address and Port	Destination IP Address and Port	Operation
1	R1	Reject	tcp	in	Address:192.168.40.121 Port:Any	Address:192.168.40.131 Port:8089	   

Figure 17: Static Defense Blocking Host 192.168.40.121 Using TCP Connection

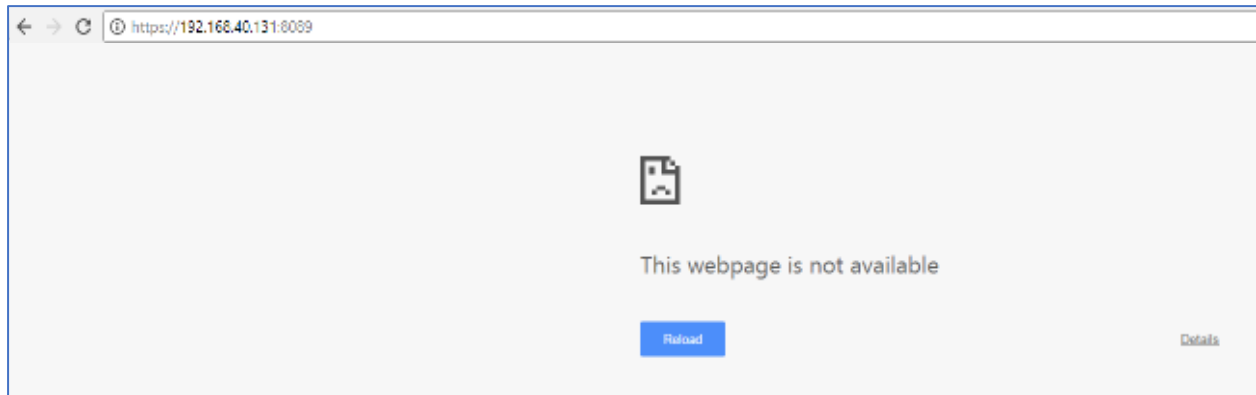
After saving and applying the change, host 192.168.40.121 will not be able to access UCM web UI anymore.





```

Connection-specific DNS Suffix  . : 
Link-local IPv6 Address . . . . . : fe80::bca4:247b:116:9d66%11
IPv4 Address. . . . . : 192.168.40.121
Subnet Mask . . . . . : 255.255.255.0
Default Gateway . . . . . : 192.168.40.3
  
```



**Figure 18: Host blocked by UCM**

## Dynamic Defense

Dynamic Defense protects the UCM by blacklisting individual hosts that send excessive connection attempts or brute force attacks. On UCM6102/6202/6204/6208/6510 devices, this requires Network Settings→Basic Settings→Method to be set to "Route". Dynamic Defense can be configured in Security Settings→Dynamic Defense, it will try to blacklist massive connection attempts or brute force attacks made by individual host.

Dynamic Defense's connection threshold and blacklist update interval can be adjusted in accordance to administrator preferences. Whitelist entries can be added to prevent specific IP addresses and ports from getting blocked.

For more configuration details, please refer to [UCM User Manual](#).

## Fail2ban

Fail2ban protects the UCM by detecting excessive failed login attempts and SIP requests and blocking any further attempts from the same host.

Fail2Ban can be configured in System Settings→Security Settings→Fail2Ban. By default, Fail2Ban is disabled (see figure below).



Security Settings

Static Defense
Dynamic Defense
**Fail2ban**
SSH Access

Global Settings

Enable Fail2Ban: ☐

Banned Duration:

Max Retry Duration:

MaxRetry:

Fail2ban Whitelist:  +

Local Settings

Asterisk Service: ☐

Login Attack Defense: ☐

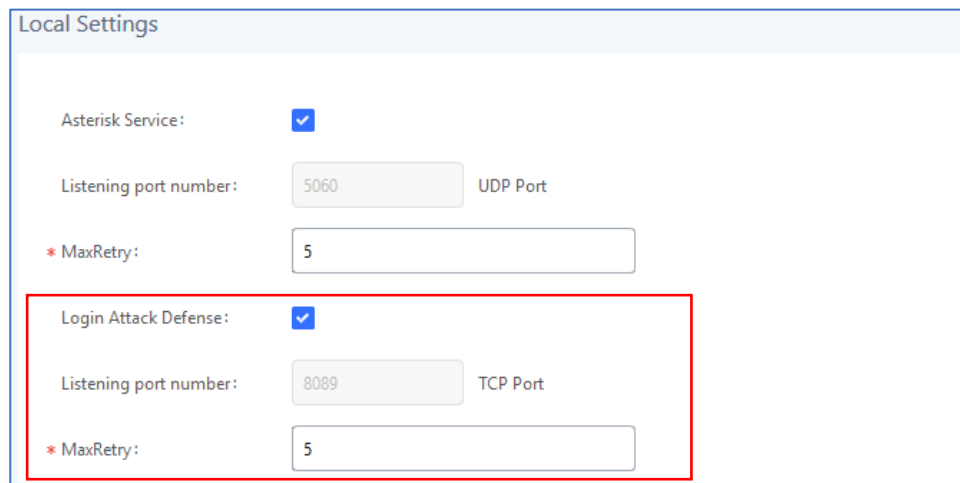
Blacklist

Banned Type	IP	Operation
-------------	----	-----------

**Figure 19: Fail2Ban Default Configuration**

- **Enable Fail2ban:** Tick the box to enabled Fail2ban.
- **Banned Duration:** The amount of time in seconds that a host is banned for. A value of "0" means an permanent ban. Set to 600 seconds by default.
- **Max Retry Duration:** The maximum amount of failed attempts allowed within this window of time. Set to 600 seconds by default.
- **MaxRetry:** The maximum number of failed attempts allowed before the host is banned. Set to 5 attempts by default.
- **Fail2ban Whitelist:** The list of hosts that would not be blacklisted even if the number of failed SIP and login attempts exceeds the allowed limit.





Local Settings

Asterisk Service: ☒

Listening port number: 5060 UDP Port

\* MaxRetry: 5

Login Attack Defense: ☒

Listening port number: 8089 TCP Port

\* MaxRetry: 5

**Figure 20: Asterisk Service Fail2Ban setting**

If Enable Fail2ban is ticked under the Global Settings section, either Asterisk Service, Login Attack Defense, or both must be enabled under the Local Settings section. The MaxRetry values under these options will override the value set in the Global Settings section.

**Asterisk Service** - Protects against excessive failed SIP REGISTER, INVITE, and SUBSCRIBE requests. If the retry limit has been reached, the UCM will not respond to any more SIP requests from the same host. The Listening port number value will match the configured SIP port number in the PBX Settings→SIP Settings.

**Login Attack Defense** - Protects against excessive failed login attempts. If the retry limit has been reached, the UCM will prevent the host from accessing the UCM web UI. The Listening port value will match the configured port number in System Settings→HTTP Server→Port.



## AMI

Asterisk Manager Interface (AMI) is supported on UCM with restricted access. Please refer to the following link for detailed documentation on the feature:

[http://www.grandstream.com/products/ucm\\_series/UCM/documents/UCM\\_ami\\_guide.pdf](http://www.grandstream.com/products/ucm_series/UCM/documents/UCM_ami_guide.pdf)

It is highly recommended to not enable AMI on a UCM that is placed on a public or untrusted network unless precautions have been taken to prevent unauthorized access and restrict permissions. AMI access allows users to originate calls and view sensitive information on the UCM.

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