

**VOS3000
V2.1.7.01
Manual**

Issue 01
Date 2017-05-08



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address: Nanjing University of Science and Technology Science
Guanghua Road 1st, Qinhuai
Nanjing 210014
People's Republic of China

About This Document

Purpose

This document describes the functions and operations of VOS3000 client.

Intended Audience

This document is intended for:

Installation and commissioning engineers

NM configuration engineers

Symbol Conventions

The symbols that may be found in this document are defined as follows.

Symbol	Description
 DANGER	Indicates a hazard with a high level or medium level of risk which, if not avoided, could result in service interruption.
 WARNING	Indicates a hazard with a low level of risk which, if not avoided, could result in data missing.
 CAUTION	Indicates a potentially hazardous situation that, if not avoided, could result in unanticipated results.
 TIP	Provides a tip that may help you solve a problem or save time.
 NOTE	Provides additional information to emphasize or supplement important points in the main text.

Change History

Changes between document issues are cumulative. The latest document issue contains all the changes made in earlier issues.

Issue 01 (2017-05-08)

This issue is used for first office application (FOA).

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1 Guide

About This Chapter

This chapter describes the user guide of VOS3000 client.

Operation Guide

In VOS3000, most data managements can be completed through sheets. Sheets can be opened by double-clicking corresponding nodes in “Navigation”. The following operations are supported:

Open: open current management page.

Filter: get current configuration from server.

Copy: copy the currently selected sheet line into the clipboard.

Paste: paste the line in the clipboard into a sheet with the same type.

Add: insert new lines.

Delete: delete sheet lines. If the data are at the server, the selected lines will be marked as “to be deleted”.

Apply: send currently specified operations (such as add, delete and modify) to the server to carry out.



NOTE

Before clicking “apply”, all the operations of data are saved only at the client end and will not affect the server's data; closing the management page would discard these operations.

Export: export the current sheet into local files.

Import: import data from local files into the sheet (supported by a few types of sheets).

 NOTE

Batch data operations can be fulfilled by “copy”, “paste” and column “Filling downwards” functions supported by spreadsheets in VOS3000. See the figure below:

Figure 1-1 Filling downwards



2 Function

About This Chapter

This chapter describes the functions of VOS3000.

2.1 Login

Upon running VOS3000 client, the login dialogue will be shown.



Server ip: IP address of the remote server.

User name: user names allowed by the platform.

Password: user password allowed by the platform.

Uuid: during the first installation, system will generate this id.

Even password is correct, uuid is still needed. This id can be modified any time, please refer to <Configuration Guide>.

The system will record IPs typed by users for later use. Users can also delete these historical servers IP.

The initial user name and password are admin and admin.



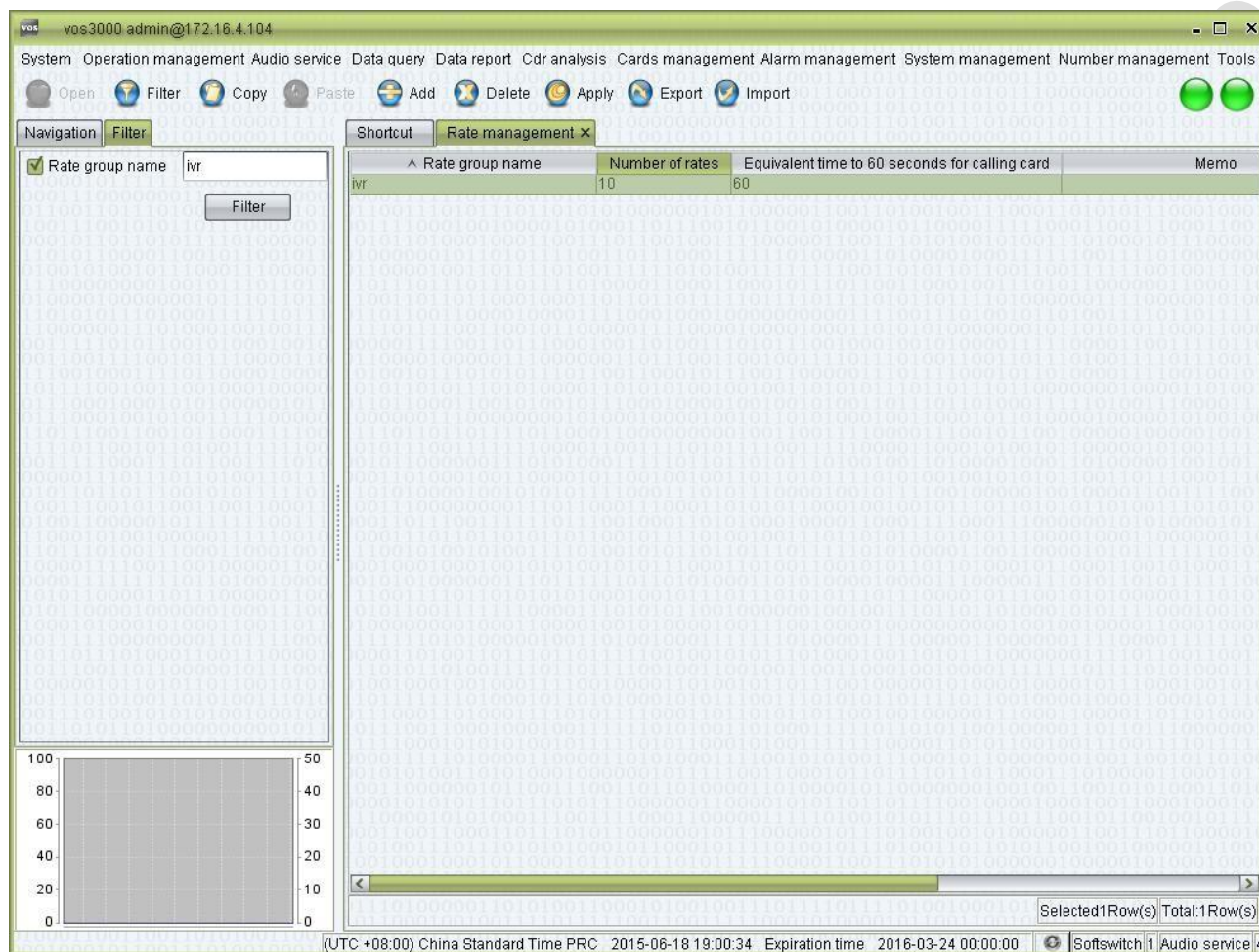
CAUTION

Please modify the initial login password as soon as possible.

2.2 Rate Management

2.2.1 Rate Group Management

This function is used to define rate groups used for billing.



How to Start

Double-click "Navigation > Rate management"

Table Items

Rate group name: the name of the rate group. When new accounts are created, one rate group must be specified and the name here will be used to identify the groups.



NOTE

Try to choose more informative names that remind people of the rate's details.

Number of rates: the number of rates contained in the group.

Equivalent time to 60 seconds for calling card: for calling card only.

Memo: additional comments.

Number of using account: the number of accounts using this rate group.

Create user: the name of the user who created this rate group.

Time of timing replace: replace time.

Mode of timing replace: full replace or append replace.

Name of timing replace: rate group name after replace.

Other Operations

Double-click the numbers at “Number of rates” to enter the rate management page.

Double-click the numbers at “Number of using account” to enter the account management page.

Right-Click Menu

Open: enter the rate management page.

Fee rate of timing replace: add timing replace fee rates.

Authorization: enter the authorization management page.



NOTE

You can simultaneously select multiple rate groups.



For agent, when login system, only those rate groups in the “Authorization” lists will be shown. As illustrated above, this rate group can be seen by "agent1", "agent2" and "agent3".

2.2.2 Rate Management

This function defines the prefixes to each number during the billing.

Rate prefix	Area prefix	Rate type	Area name	Billing rate	Billing cycle	Rate(Minute)	Lock type
0		Domestic		0.1000000	60	0.1000000	No lock
1		Domestic		0.1000000	60	0.1000000	No lock
2		Domestic		0.1000000	60	0.1000000	No lock
3		Domestic		0.1000000	60	0.1000000	No lock
4		Domestic		0.1000000	60	0.1000000	No lock
5		Domestic		0.1000000	60	0.1000000	No lock
6		Domestic		0.1000000	60	0.1000000	No lock
7		Domestic		0.1000000	60	0.1000000	No lock
8		Domestic		0.1000000	60	0.1000000	No lock
9		Domestic		0.1000000	60	0.1000000	No lock

How to Start

Select a line in the rate group management page and choose “Open” in the right-click menu.

Double-click the “Number of rate” in the rate group management page.

Filter

Billing prefix: input billing number to find used billing rate.

Add domestic area code(full):add Chinese mobile phone prefix 013,015,018 and all area code defined by “City code” automatically.

Add domestic area code(brief):add prefix 01-09 automatically

Table Items

Rate prefix: the prefix of the called number which matches this rate.

NOTE

The longest matching prefix will be used. For example, if there are two rate prefix, "0" and "01", the number "01117" will be match to "01" since it is the longest pattern that matches "01117".

Area prefix: also used for rate matching, area name will be shown according this, please refer to "Number management > Area information".

Rate type: available options are "Net", "Local", "Domestic" and "International".

NOTE

The type will not only be shown in CDR, but also used in many filters and statistics. Please correctly specify the rate type. Meanwhile, this type will be checked before calling. If a caller does not have the authorization to call the type of number specified here, the call will be banned.

Area name: the area corresponding to the rate prefix. Non-editable, which is set by "Number management > Area information", displays the "Area information" of "Area prefix".

Billing rate: the fee charged for each billing cycle.

Billing cycle: the session time that exceeds the "First time duration" will be divided into units with length specified here. When the time cannot be divided clear, it will be rounded up to the nearest integer.

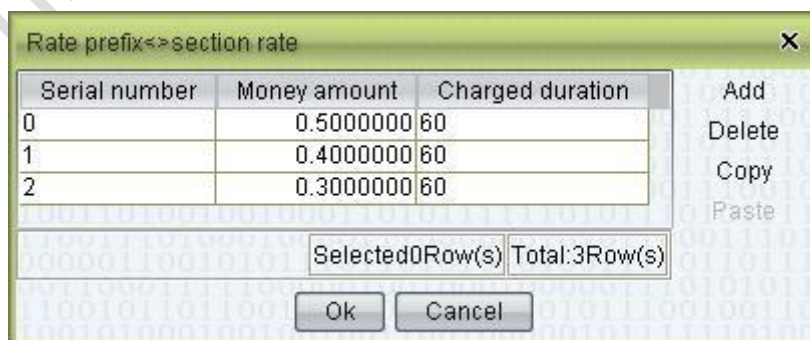
NOTE

An example: if the "First time rate" is "0.21", the "First time duration" is "180", the "Billing rate" is "0.15", and the "Billing cycle" is 60, then according to this rate, a session that lasts 250s will be charged $0.21 + 0.15 * 2 = 0.51$.

Minute cost: calculate by "Billing rate" and "Billing cycle".

Lock type: enable or disable the rate.

Section rate:



Serial number	Money amount	Charged duration
0	0.5000000	60
1	0.4000000	60
2	0.3000000	60

Selected 0 Row(s) Total: 3 Row(s)

Ok Cancel

Serial number: time section.

Money amount: money of this section.

Charged duration: charge time.



NOTE

Section rate will be used first, sessions that exceeds this time will be charged according to the "Billing rate" and "Billing cycle". If none section is set, the "Billing rate" and "Billing cycle" will be used from the beginning. (Unit: Second).

Billing rate for calling card prompt: used for IVR prompt remaining duration.

Billing cycle for calling card prompt: used for IVR prompt remaining duration.

Other Operations

The table supports "Import" and "Export" operations.

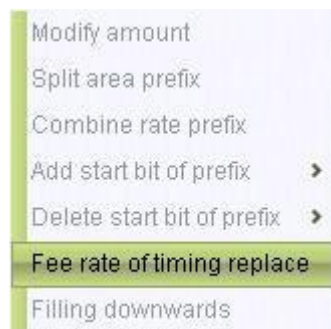


NOTE

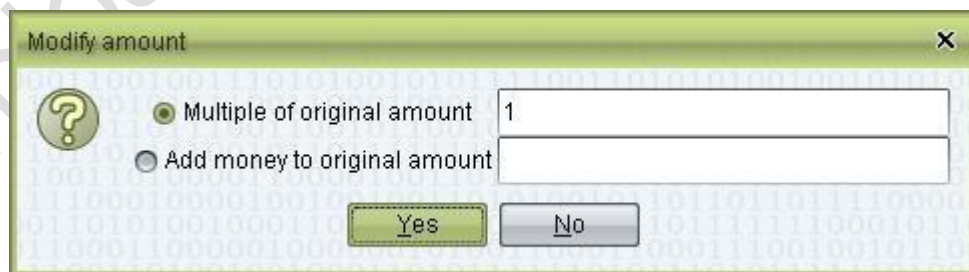
Refer to the exported files for the import format.

Right-Click Menu

Modify amount: on column "Billing rate".



Modify amount: select the need to modify the amount of the line, when the mouse is in the <Billing rate> or <Rate(Minute)>, can be selected on the amount of line adjustment.



Multiple of original amount: the original <Billing rate> amount multiplied by the set value.

Add money to original amount: the original <Billing rate> amount add the set value.

Split area prefix: for new line only.

Rate prefix	^ Area prefix	Rate type
	1,2,3,4,5	Domestic
Modify amount Split area prefix Combine rate prefix		

Rate prefix	^ Area prefix
	1
	2
	3
	4
	5

Combine rate prefix: combine “Rate prefix” and “Area prefix” to “Area prefix”.

Rate prefix	^ Area prefix	Rate type
88	1	Domestic
Modify amount Split area prefix Combine rate prefix		

Rate prefix	^ Area prefix
	881

Add start bit of prefix: when selected lines are in the status of “to be added”, the function can be used to modify their initials.

Delete start bit of prefix: when selected lines are in the status of “to be added”, the function can be used to modify their initials.

Fee rate of timing replace: At a certain time,modify the setting rates.

Fee rate of timing replace<yltest>management						
Time of timing repla...	Mode of timing replace	Rate prefix	^ Area prefix	Rate type	Area name	Billing rate
2015-04-23 00:00	Append replace		2	Domestic		0.5000000
2015-04-23 00:00	Delete		3	Domestic		0.0000000

NOTE

Some international rate list use nonstandard initials. These functions can be used to adjust the initials after import.

When creating rate for national calls, try to use rate prefixes like “01” to “09” instead of using a single “0”. Otherwise, if the international rate are incomplete, some international calls might be matched to “0” and misclassified as national calls.

TIP

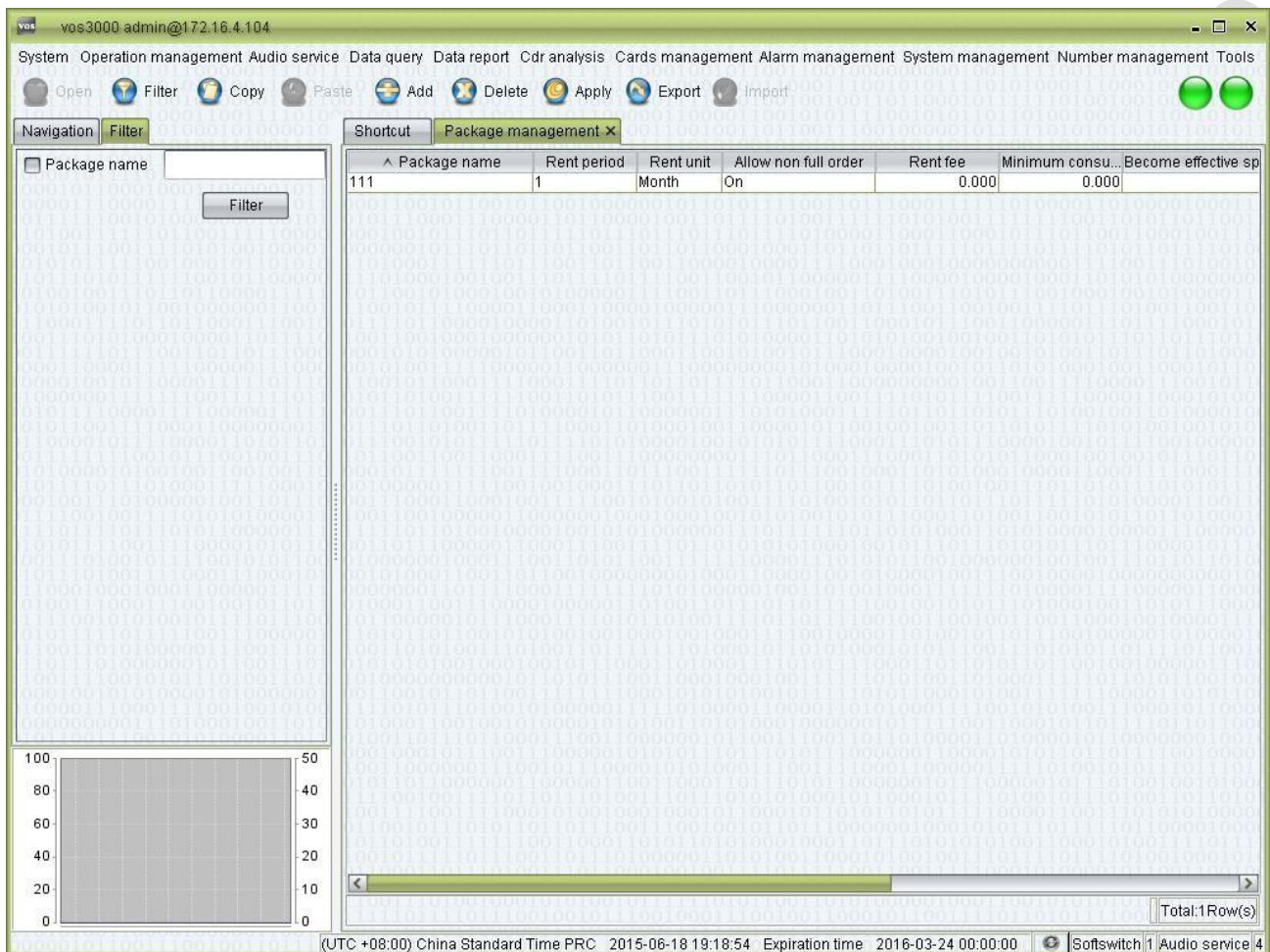
Some provider gives minute cost only, when import, you can set “Rate(Minute)”, system will calculate “Billing rate” automatically. You can use split area prefix and combine rate prefix to modify rates,obtain the rate which can be vos system accepted.

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2.3 Package Management

2.3.1 Package Group Management

This function is used to define packages for billing.



How to Start

Double-click "Navigation > Package management"

Table Items

Package name: the name of the package.

Rent period: value of time.

Rent unit: Day/Month/Year

**NOTE**

Package's period = Rent period * Rent unit, e.g. rent period: 7, rent unit: day, means 7 days package, every 7 days will effect until expire.

Allow non full order:

- On:if package order is not on the first day of subscription period, the rent is calculated from ordered day to expired day of package.
- Off: if package order is not on the first day of subscription period ,then tent full deduction,The rent will be fully deducted .

Rent fee: Package cost.

**NOTE**

If rent fee is 0, rent will be always subscribed.

Minimum consumption: calculate when package expired, if account under consumption, the difference will be deducted.

Become effective spending limit :only if period suite consumption more than or equal to this value ,this package can be used.

Period rate: the rate of a certain time period. Double-click to open the period rate management. Please refer to the next section for details.

Free duration: free sessions provided by the package. Double-click to edit it.

Free money amount: gift money provided by the package. Double-click to edit it.

**NOTE**

The free duration will be used first, then the free money amount.

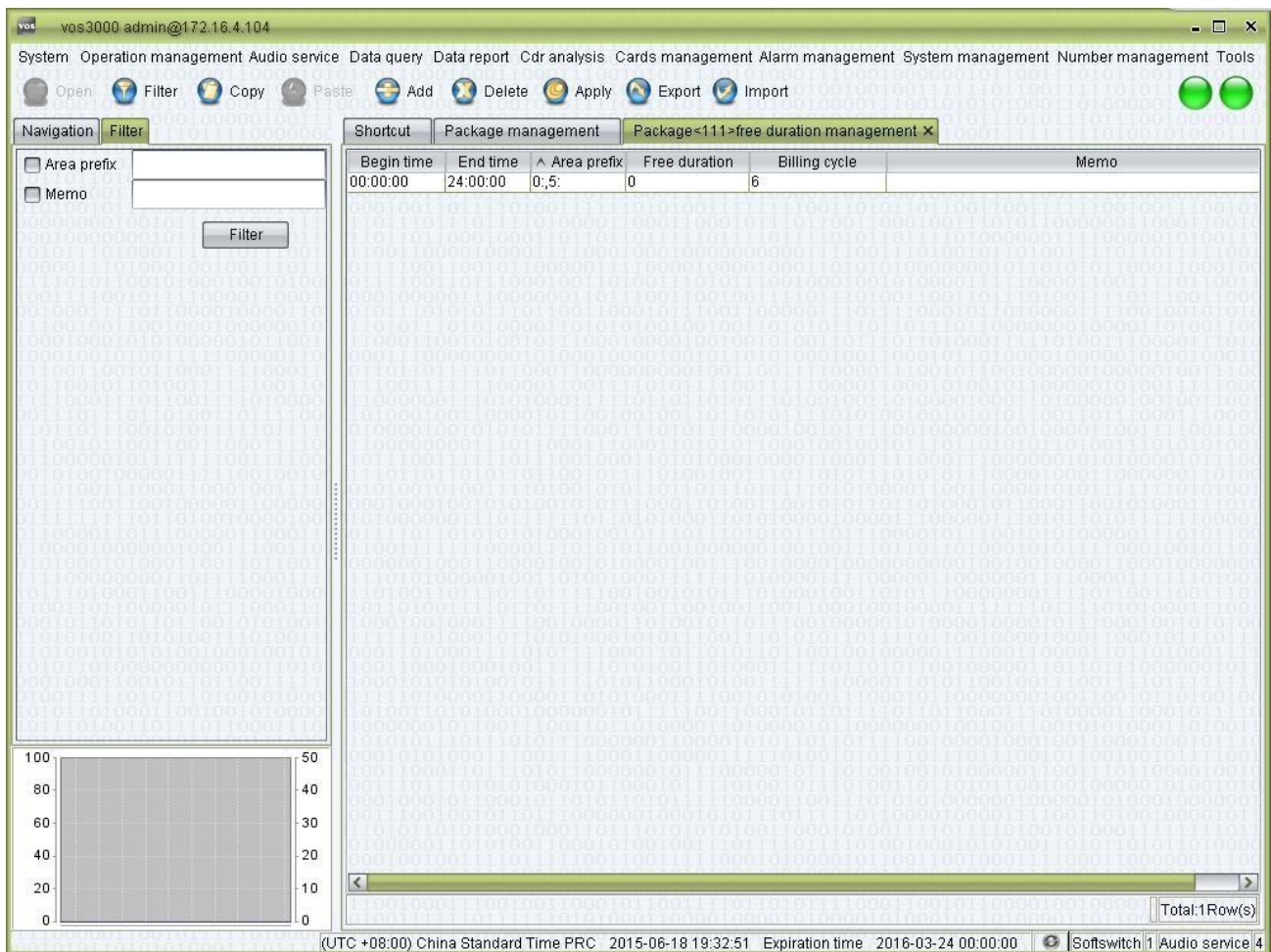
Memo: additional comments.

Create user: the name of the user who created this package.

Right-Click Menu

Authorization: refer to rate management.

2.3.2 Package Free Duration Management



How to Start

Double-click “Navigation > Package management > Package free duration management”

Table Items

Begin time

End time



NOTE

Call time within Begin Time and End Time, free duration will be used. If not whole day time, free duration is from call start to End Time.

Area prefix: the prefix of free duration.

Free duration: free time for the prefix.

Billing cycle: charge cycle.

Memo: comments on the package.

Right-Click Menu

Authorization



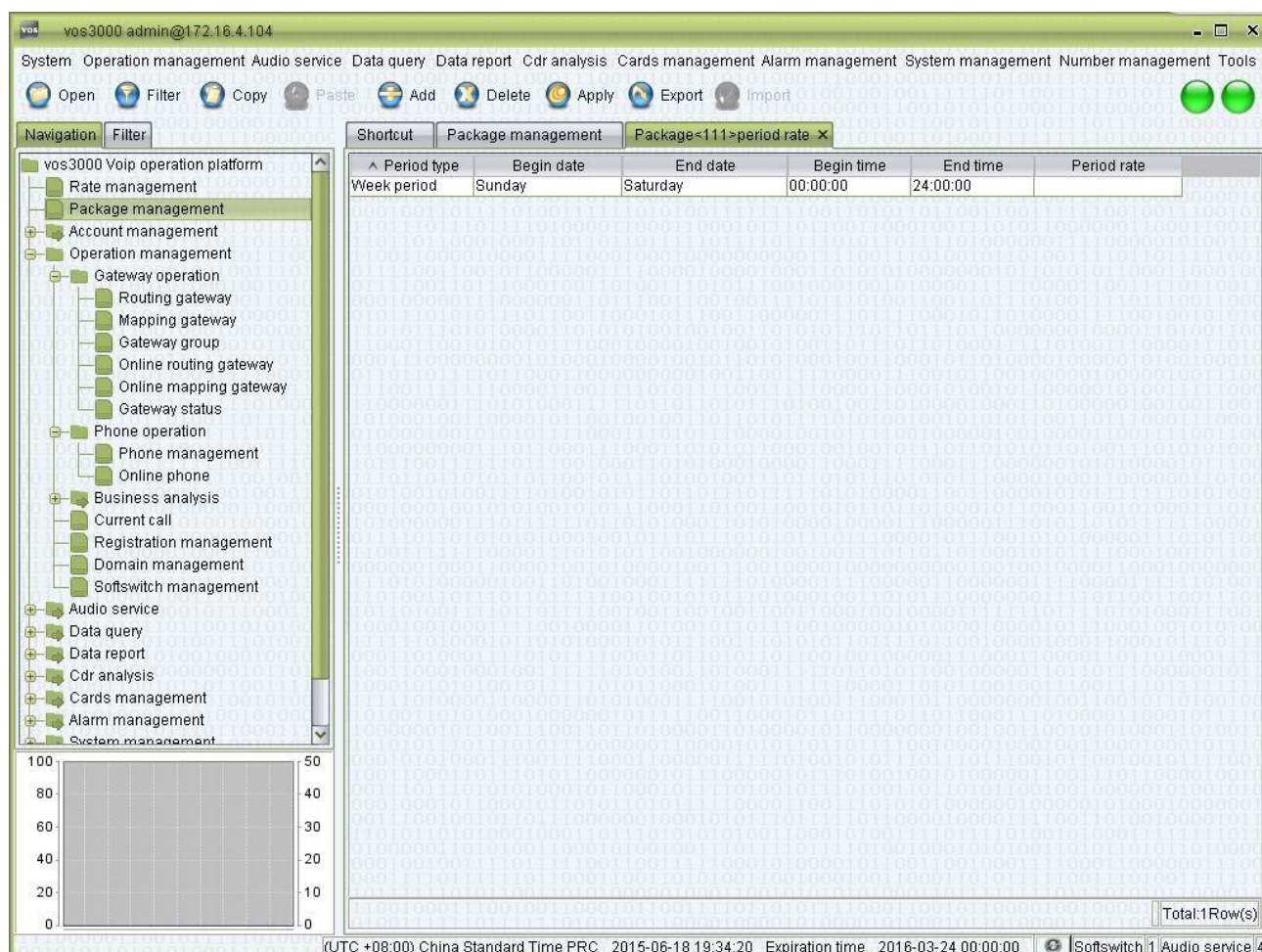
NOTE

Please refer to: rate group management.

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2.3.3 Package Period Rate Management

This function is used to define packages for period rate, One package can define different periods with different billing rates



How to Start

Double-click the numbers at the “Package management > Package period rate”

Examples:

The rate in the weekends is “0.5”.

^ Period type	Begin date	End date	Begin time	End time	Period rate
Week period	Sunday	Saturday	00:00:00	24:00:00	0.5

The rate from 0:00 to 8:00 is “0.5” and “0.6” for the rest of the time.

^ Period type	Begin date	End date	Begin time	End time	Period rate
Week period	Sunday	Saturday	00:00:00	08:00:00	0.5
Week period	Sunday	Saturday	08:00:00	24:00:00	0.6

The rate from 0:00:00, September 1st, 2013 to 21:15:00, October 21th, 2013 is “0.5”.

^ Period type	Begin date	End date	Begin time	End time	Period rate
Year period	2013-09-01	2013-10-21	00:00:00	21:15:00	0.5



NOTE

The start and expiration time for yearly periods are different from those for weekly or monthly ones.

2.4 Account Management

2.4.1 General Account

The screenshot displays the VOS3000 web interface for account management. The top navigation bar includes links for System, Operation management, Audio service, Data query, Data report, Cdr analysis, Cards management, Alarm management, System management, Number management, and Tools. Below this is a toolbar with icons for Open, Filter, Copy, Paste, Add, Delete, Apply, Export, and Import. The left sidebar contains a 'Navigation' menu with 'Filter' selected, and a 'Shortcut' menu with 'Account management' selected. The main area shows a form for creating or editing an account, with fields for Account id, Account name, Account type, Phone number, Gateway name, Bind number, Billing rate, Today consumption, Expiry date, and Expiry date. The 'Account type' is set to 'General'. Below the form is a 'Filter account' section with checkboxes for 'Direct' and 'All'. The 'Account category' section has checkboxes for 'Agent' and 'Account'. The 'Account status' section has checkboxes for 'Neg. balance', 'Pos. balance', 'Normal', 'Locked', 'Active', and 'Suspended'. A table at the bottom shows the account details for 'yjtest'.

Account id	Account name	Current balance	Overdraft limit	Billing rate	Private rate	Today consumption	Number
yjtest	yjtest	397.630	0.000	yjtest	0	0.000	3

How to Start

Double-click "Navigation > Account management"

Table Items

Account id: the unique identification of the account. This must be unique and cannot be modified once the account is created.

Account name: the name of the account (such as the full name of the user).

Current balance: the current balance of the account.

Overdraft limit: the maximum credit of the account.

Billing rate: the rate group for billing.

Private rate: rate for this account only, if billing rates contains private rate, the lower will be chosen.

Today consumption: today's consumption.

Number of gateway: non-editable. The number of gateways under this account. Double-click to enter the gateway management page of this account.

Number of phone: non-editable. The number of phones under this account. Double-click to enter the phone numbers management page of this account.

Number of active phone card: non-editable. The number of phone cards under this account. Double-click to enter the phone card management of this account.

Bind number: non-editable. The bind numbers of this account.

Number of package: non-editable. It shows the number of packages subscribed by this account.

Current package: non-editable. The number of packages effected.

Agent id: the "Account id" of its parent account. The parent account must exist. Upon designation, the parent account will become the "Agent" type.

Customer information: information about the customer. Click "Edit" to change.

Directory: non-editable. The number of this account's phonebook. Double click to enter short number settings.

Directory limit: number of phonebooks.

Account category: "Account" or "Agent", non-editable. When an account has sub accounts, it automatically becomes an agent.

Account type: "General", "Phone card" or "Clearing".

Memo: comments on the account.

Account status: "Normal" or "Locked".

CTD billing model: just used for callback business, and account is "Agent" type. Standard: callback billing mode of "Agent" account is decided by the subsidiary account's in callback business

Flow: callback billing mode of "Agent" account is decided by its own setting in callback business.

 TIP

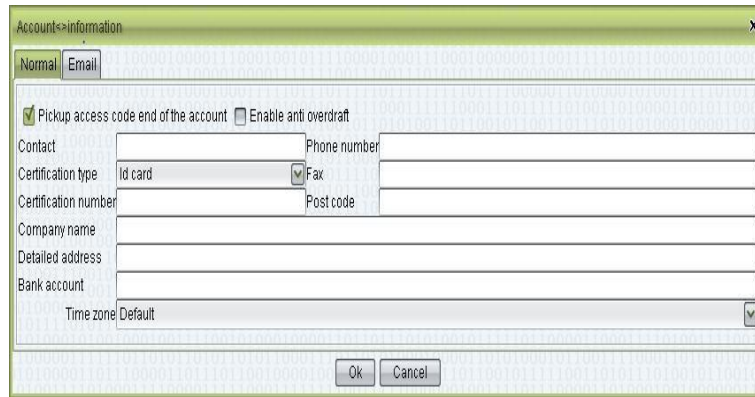
CTD billing mode setting is often used for the condition where the first line free billing and the second line billing in callback business.

Expiry date: the expiration date of the account.

Advance amount: advanced money for ongoing calls

 TIP

Additional settings ----Enable anti overdraft



Date of open account: non-editable. The date when the account is created.

Other Operations

Double-click the number at “Number of gateway” to enter “Mapping gateway management” page for the account.

Double-click the number at “Number of phone” to enter “Phone number management” page.

Double-click the number at “Number of package” to manage the packages subscribed by the account.

Double-click the number at “Current package” to edit the effected package of the account.

Right-Click Menu



Disable account: disable the account and all its sub accounts. Phone numbers belonging to these accounts will not be able to make phone calls.

Enable account: enable the disabled account.



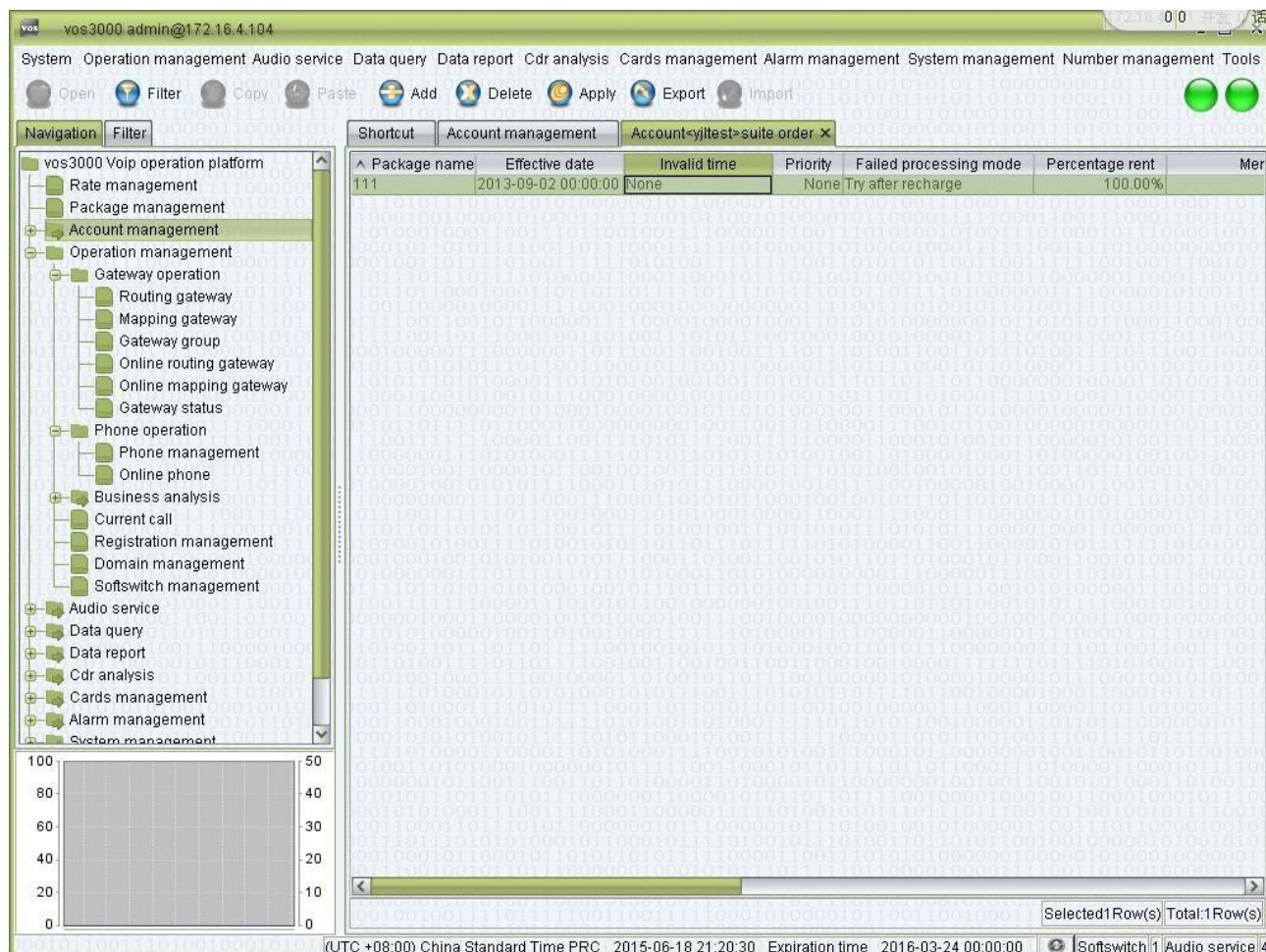
NOTE

If an account is deleted, his phone and gateway will belong to no account.

You cannot change clearing account into other type.

2.4.1.2 Customer Package Management

This function is used to manage accounts packages.



How to Start

Double-click "Navigation > Account management > Number of package"

Table Items

Package name

Effective date

Invalid time



NOTE

Add packages for the account then choose package name and set the effective date and invalid time of packages.

The "Invalid time" is the date after which the package cannot be subscribed. For example, if a monthly lease has effective date (2013-9-2) and invalid time (2013-9-15), on the 2th of September; the package will be subscribed, and effective through 2013-9-2 to 2013-9-30. But after September 15th, the package will not be available for subscription. That means this package cannot be used in October.

Priority : package with high priority will take effect firstly (the smaller the value, the higher the priority), if packages' priority are the same but not "none", account balance will determine packages together to take effect or not. priority with "none" will finally take effect. if packages' priority are all "none", system will determine the effective order of package.

Failed processing mode: the processing mode after suite order failed.

- Try after recharge: immediately try to order suite again once account be recharged.
- Try next cycle: try to order suite again in next cycle.
- Delete order: delete the order directly.

Percentage rent: rent of percentage

Memo:

Rent period

Rent unit

Rent fee

Minimum consumption

Become effective spending limit

Period rate

Free duration

Free money amount

Suite memo

2.4.2 Payment

This function is used to pay money for account.

How to Start

Double-click “Navigation > Account management”, select an account; click “Payment” in the right-click menu.

Operation Details

Type of payment: If the phone card is selected, the card number and password will be asked.

Payment type: Payment/Credit/Make Zero.

Amount: the amount of payment.

Memo: comments will be kept as historical records.



NOTE

Historical payments can be view in the payment record page.

Payment will change account's expiration. Please refer to system parameter:

SERVER_PAY_DELAY_CUSTOMER_EXPIRE_DAY

2.4.3 Agent Account

Agent accounts differ with ordinary accounts in that there are accounts belonging to agent accounts. Once an account becomes an agent account, it will occur in the navigation tree. Double-click the agent account in the navigation tree to open the “Sub account management”.



NOTE

Use the filter “Direct” and “All”, respectively, to show the directly sub accounts and all (direct and indirect) affiliations of the account.

2.4.4 Billing

Phones, gateways and bind numbers will be charged according the account they currently belong to. If the number being called does not match any rate, the call will be terminated. And the cause of such termination will be shown in CDR. When the account belongs to other accounts, the call will cause the agent account to be charged according to its own rate (this backtracking process ends up at accounts that

belong to no other accounts). If the billing turns the account or any of its agent accounts into “disabled” status, the phones, gateways and bind numbers will no longer be able to make calls.

 NOTE

Billing principle: optimal rate, the deduction amount is calculated by period fee rate, account fee rate, account private fee rate or phone private fee rate, choose the cheapest.

When account is in debt, he can still make free call.

2.4.5 Authorization Management

This function is used for manage agent’s operation of accounts.



User<=>authorization	Allow	Reject
Add/delete/modify account	☒	☐
Add/delete/modify phone	☒	☐
Delete and modify phone card	☒	☐
Add/delete gateway	☐	☒
Modify gateway prefix	☐	☒
Modify gateway capacity	☐	☒
Payment for this account	☐	☒
Payment for sub account	☒	☐

How to Start

Double-click “Navigation > Account management”, select an account, click “Authorizations” in the right-click menu

Operation Details

Authorizations

Add/delete/modify account: the right to create delete or modify accounts.

Add/delete/modify phone: the right to manipulate phones belonging to the account.

Delete/modify phone card: the right to manipulate phone cards belonging to the account.

Add/delete gateway: the right to manipulate gateways belonging to the account.

Modify gateway information: the right to modify information about gateways except capacity.

Modify gateway capacity: the right to modify the number of lines.

Payment for this account: the right to perform payment for the current account (including changing the amount of overdraft).

Payment for sub accounts: the right to pay for the sub accounts.

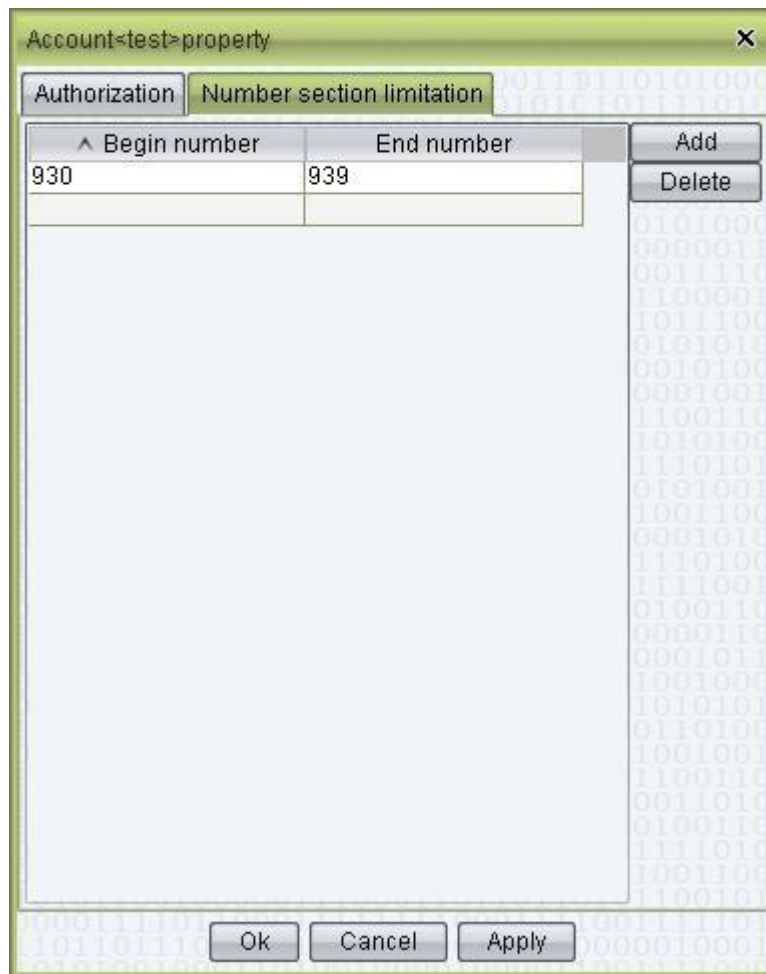


NOTE

This function is usually used to facilitate agent development. An agent user can have an agent-typed account in the system. Administrators can create one or more accounts for them, limiting their rights to recharge their own account, yet granting those authorizations to add new accounts, phones and gateways, and to recharge their sub accounts. The agent can create new accounts for its sub-agents. The agent account can only manipulate its sub accounts. Note that, accounts created by agent accounts must be designated to an agent account, and the creator must have the authorization to manipulate the designated agent account. Users logged in with an agent account can only see those accounts that authorized to the agent. This restriction applies to all account-related operations.

2.4.6 Number Section Limitation

This function is used to limit the phone prefix of the account.



How to Start

Double-click "Navigation > Account management", select an account, click "Authorizations" in the right-click menu.

Operation Details

Begin number: the smallest number of the segment (including the number itself).

End number: the largest number of the segment (including the number itself).



NOTE

Number restriction specifies the phone numbers that can be added to the current account, in order to avoid competition of number resources among agents.

If account type is "agent", the numbers of its sub accounts should also be in this range.

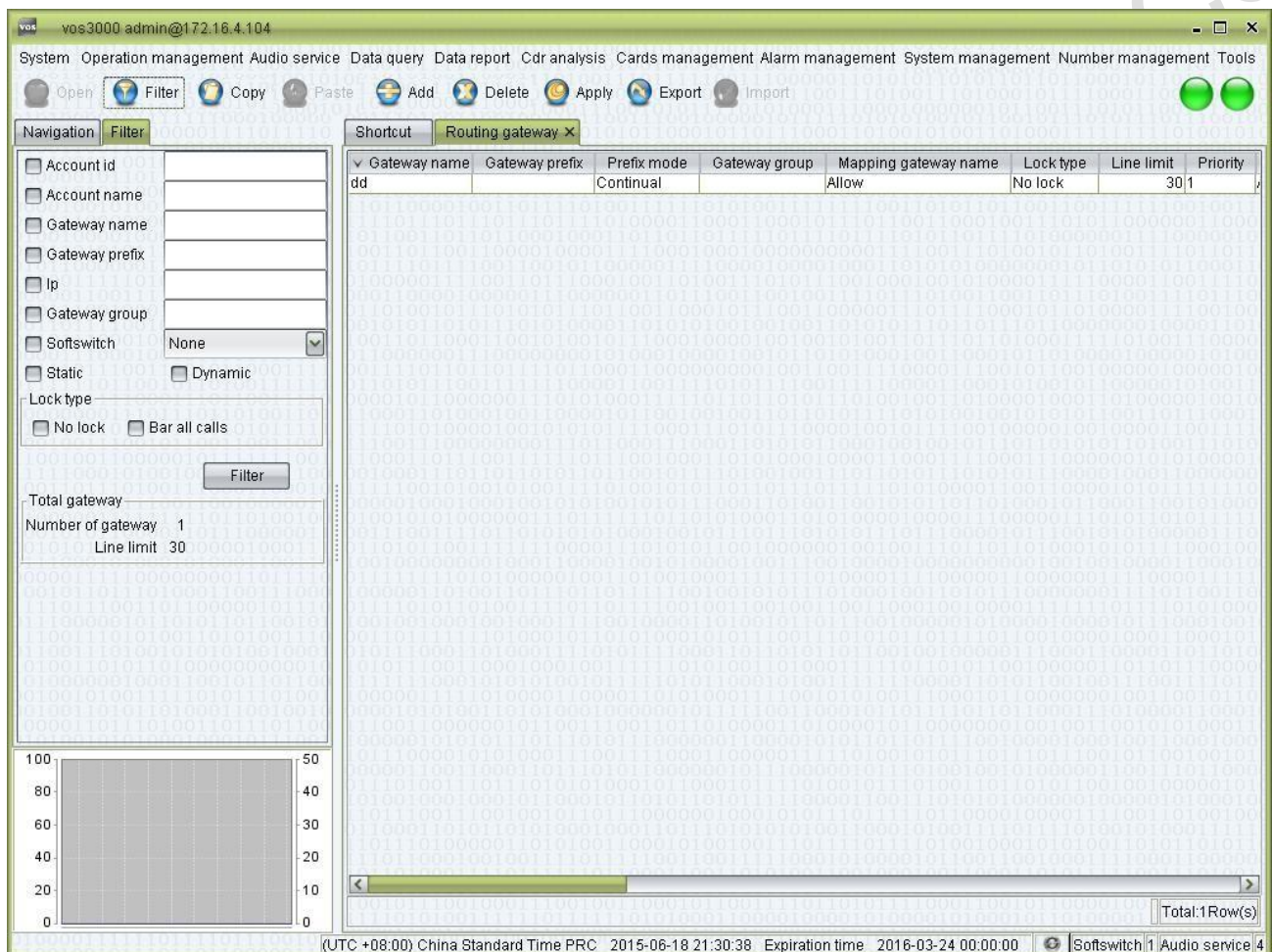
Otherwise there will be error prompt from the system. If the account type is "ordinary", the appropriate number segments will automatically added by the system.

2.5 Operation Management

2.5.1 Gateway Operation

2.5.1.1 Routing Gateway

This function is used to manage routing gateway.



How to Start

Double-click "Navigation > Operation management > Gateway operation > Routing gateway"

Table Items

Gateway id: the unique id of the device, used for the authentication of dynamic gateways. For static gateways (usually relay gateways), the only requirement is their ids do not conflict with each another.

Gateway prefix: when the number being called is not registered in the system, the call will be routed only to gateways which match the prefix specified here. Multiple prefixes can be specified, separated by commas. Different gateways can be designated with the same prefix. When conflict occurs, the gateway will be chosen according to following numbers (the smallest comes first): priority number, the ratio of the number of current calls to the number of channels, the number of historical calls, and the gateway id.

Prefix mode:

- Extension: shorter prefixes will be tried if the routing gateway matched by this prefix cannot deliver the call.
- Expiration: no more prefixes will be tried if the routing gateway matched by this prefix cannot deliver the call.

Examples:

Gateway id	Gateway prefix	Prefix mode	Gateway group	Lock type	Line limit	Priority
gw1	9	Continual		No lock	30	1
gw2	900	Terminal		No lock	30	0
gw3	9	Continual		No lock	30	1
gw4	900	Continual		No lock	30	1

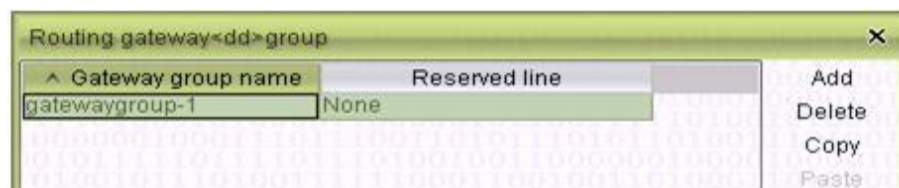
If the prefix mode of “gw2” is set to “Terminal”, the prefixes being tried for the number “90080001” will be “gw2” and “gw4” in order.

If the prefix mode of “gw2” is set to “Continual”, while others remain the same, the prefixes being tried for the number “90080001” will be “gw2”, “gw4”, “gw3”, and “gw1” in order.

Gateway group: the group which gateway belongs to.

Reserved line: line at least to be reserved for gateway group.

Group line limit will restrict gateway line limit when this gateway belong to this group. If remaining available line of group is below the value of reserved line, this gateway will be disabled. Means when line resource strained, function of reserved line used to assure line resource is available for important customers.



NOTE

Suppose that client A and client B use the same routing gateway, which has a line limit of 600(client A and client B have different callee prefixes). If we want to give client A and client B different priorities as : client A is permitted to use all the 600 lines ;client B is permitted to use only 200 lines for the most; if client A's concurrency is over 550 lines, the routing gateway shuts down the service for client B and only provide service for client A, we can do as the following.

Create gateway group G,Line limit is 600.

Create routing gateway A which only allow client A to pass through, line limit is 600, belonging to group G and the reserved line is "none".

Create routing gateway B which only allow client B to pass through, line limit is 200, belonging to group G and the reserved line is "50".

According to the above configuration,when call can use the gateway A,concurrent gateway group G can reach 600 (Gateway Group line limit),when call can use the gateway B,concurrent gateway group G can not exceed 550 (Gateway Group line limit minus 50 lines),If the gateway A current concurrency is 520,the gateway B current concurrency is 40,the call can use the gateway A,and can not use the gateway B.From another point of view can be considered,at the peak of the priority to protect the ability to connect the client A,and in control of the maximum-peak concurrent client B's.

Note that this ground if the overall concurrent gateway less than 550,the maximum concurrent client B can reach 200.Client A and client B who call first,only to restrict client B calls,when gateway concurrent more than 550.

NOTE

Instructions for combination use of period capacity and reserved line

Create gateway group G, line limit is "none".

Create routing gateway A, line limit is "none", belonging to G and reserved line is "none", gateway A's period capacity is that during 00:00:00-18:00:00, line limit is 200 ,during 18:00:00-24:00:00, line limit is 400.

Create routing gateway B, line limit is 200, belonging to G and reserved line is 50.

As above setting, during 00:00:00-18:00:00 every day, when gateway A is available, G's concurrency can reach 400() when gateway B is available, G's concurrency can reach 350() , during 18:00:00-24:00:00 every day, when A is available, G's concurrency can reach 600. when B is available, G's concurrency can reach 550.

Mapping gateway name:Set the mapping gateways which are allow/forbidden to use the routing gateway.

Lock type: "No lock" or "Bar all calls".

Line limit: lines of this routing gateway.

Priority: when the prefixes are the same, gateway with high priority will be used firstly.

Softswitch: specify the softswitch used by this routing gateway.

Additional settings

Ip: see descriptions in the “Additional settings”.

Configuration password: the password used for gateway registration, which is also the password used for web configuration.

Self service password: the password used for web query.

Caller black/white list group: can set numbers in “Number management > Black/white list group”.

Callee black/white list group: can set numbers in “Number management > Black/white list group”.

Memo: comments on the gateway.

Routing clearing account id: the billing account charged when the gateway is called.

Routing clearing account name: the name of the billing account.

Routing clearing account balance: the balance of the billing account.

Clearing billing number

- Before rewrite
- After rewrite

Other Operations

Double-click the “Routing clearing account name”, can directly enter to the manage interface of this account.

Right-Click Menu



Current call

Condition monitor

Connect analysis

Interrupt analysis

Alarm monitor

Total gateway

Filling downwards

Random generate

Downloaded from VOS3000.Com

Additional settings > Normal

Gateway<testgw>additional settings

Normal Routing prefix Period control Routing settings Protocol Codec Others

Gateway type: Dynamic Protocol: SIP

Ip: Signaling port: 5060

Local ip: Default Local port: Default

Signaling encryption: None

Media proxy: Auto Rtp interrupt detection: None ☐ Recording

Switch gateway until connect: Default Conversation limitation(seconds): Default

Stop switch gateway when rtp start: Default Callee busy stop switch: Default

Real time computing asr: Default Real time computing acc: Default

☐ Check rate ☐ Sort by lowest rate per second ☐ Lowest profit rate limit: 0 %

☐ Max minute rates: 0.0000000

Ok Cancel

Gateway type

- Static: no registration is required. These are usually relay gateways (i.e. routing gateway). Static IP addresses and ports can be specified for them.
- Dynamic: registration is required before use.
- Registration: register to other server; please refer to "Registration management".

Protocol: H323 or SIP.

Ip: gateway's IP.

Signaling port

Local ip: set IP of local network used for sending, use "Auto" to choose by Linux routing table.

Local port: set port of local network used for sending

Media proxy:

- Auto: let the system decide whether enable media proxy. (Recommended)
- On: always enable media proxy.
- Off: always disable media proxy.

Signaling encryption:

RTP interrupt detection

- None: disable detection.
- Server to remote: detect audio send from server to device.
- Remote to server: detect audio send from device to server.
- Bidirection: detect both side, if any one side no audio, the call will be interrupt.

Recording: record the call

Conversation limitation (seconds): set the max call duration of the gateway.

- None: no limit.
- Default: set by "Operation management > Softswitch management > Additional settings > System parameter > SS_MAX_CALL_DURATION".

Switch gateway until connect

- Default: set by "Operation management > Softswitch management > Additional settings > System parameter > SS_GATEWAY_SWITCH_UNTIL_CONNECT".
- On: switch gateway until softswitch got connect signal (SIP 200 OK/H323 Connect).



NOTE

This option priors to "Protocol > Stop switch gateway after olc" and "Stop switch gateway after receive sdp".

- Off: try next gateway except: call connected, ringing, receive busy or no answer and settings in "Protocol".

Stop switch gateway when rtp start

- Default: set by "Operation management > Softswitch management > Additional settings > System parameter > SS_GATEWAY_SWITCH_STOP_AFTER_RTP_START".
- On: if using media proxy, when softswitch received RTP packet from the gateway, won't try next gateway any more.

 NOTE

This option is NOT affected by “Switch gateway until connect”. When “Switch gateway until connect” is on, if received RTP packet, stop switch gateway.

- Off: ignore the RTP starting conditions for gateway switch.

Callee busy stop switch

- Default: set by “Operation management > Softswitch management > Additional settings > System parameter > SS_GATEWAY_SWITCH_STOP_AFTER_USER_BUSY”.
- On: when softswitch received busy signal from the gateway, won't try next gateway any more.

 NOTE

This option is NOT affected by “Switch gateway until connect”. When “Switch gateway until connect” is on, if received busy signal, stop switch gateway.

- Off: ignore the RTP starting conditions for gateway switch.

Real time computing asr

- On: softswitch will calculate this gateway's ASR.
- Off: softswitch will not calculate this gateway's ASR.
- Default: set by “Operation management > Softswitch management > Additional settings > System parameter > SS_GATEWAY_ASR_CALCULATE”.

Real time computing acd

- On: softswitch will calculate this gateway's ACD.
- Off: softswitch will not calculate this gateway's ACD.
- Default: set by “Operation management > Softswitch management > Additional settings > System parameter > SS_GATEWAY_ACD_CALCULATE”.

Check rate: if the call has clearing fee rate, this gateway will be tried.

Sort by lowest rate per second: use rate per second when sorting. Sort order can be set in “Operation management > Softswitch management > Additional settings > System parameter >

SERVER_GATEWAY_ROUTE_FEE_RATE_BEFORE_QUALITY” and “SERVER_GATEWAY_ROUTE_FEE_RATE_SORT_CONFIG”.

- Enable: during fee rate sorting, use actual fee rate.

- Disable: during fee rate sorting, see the gateway has the lowest fee rate.

Lowest profit rate limit: lock this gateway when profit below settings. When the difference, calculate by rate per second, between caller fee rate and clearing fee rate lower than the value, this gateway won't be tried.



NOTE

Negative is supported.

Max minute rates: when minute rate above the value, this gateway won't be tried.

Additional settings > Routing prefix

Routing caller prefix

- Allow: prefixes of the caller numbers allowed to pass through (left blank to allow all numbers)
- Forbidden: prefixes of the caller numbers disallowed to pass through



NOTE

Only one of the "Allow" and "Forbidden" options can be chosen.

Routing callee prefix

- Allow: prefixes of the called numbers allowed to pass through
- Forbidden: prefixes of the called numbers disallowed to pass through

Forwarding prefix

- Allow: forwarding prefixes to pass through (left blank to allow all numbers)
- Forbidden: disallow forwarding prefixes to pass through



NOTE

Forwarding number is in the To field of forwarding signal.

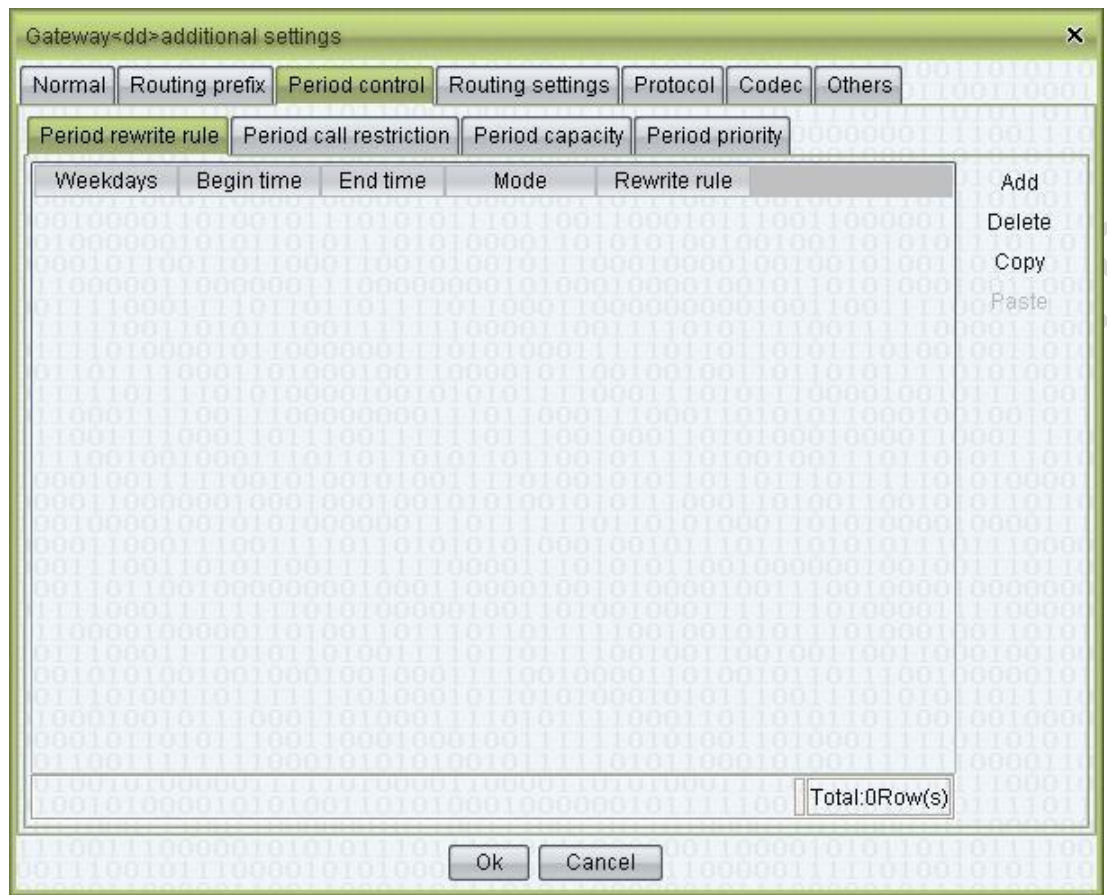
Call Restriction on Caller

- Allow: allow the caller with particular prefix to dial callee with particular prefix.
- Forbidden: forbidden the caller with particular prefix to dial callee with particular prefix.

Routing caller rewrite rule: rewrite rules for the caller number when called out through this gateway.

Routing callee rewrite rule: rewrite rules for the called number when called out through this gateway.

Additional settings > Period control > Period rewrite rule



Weekdays

Begin time: time when the rewrite rule comes into effect.

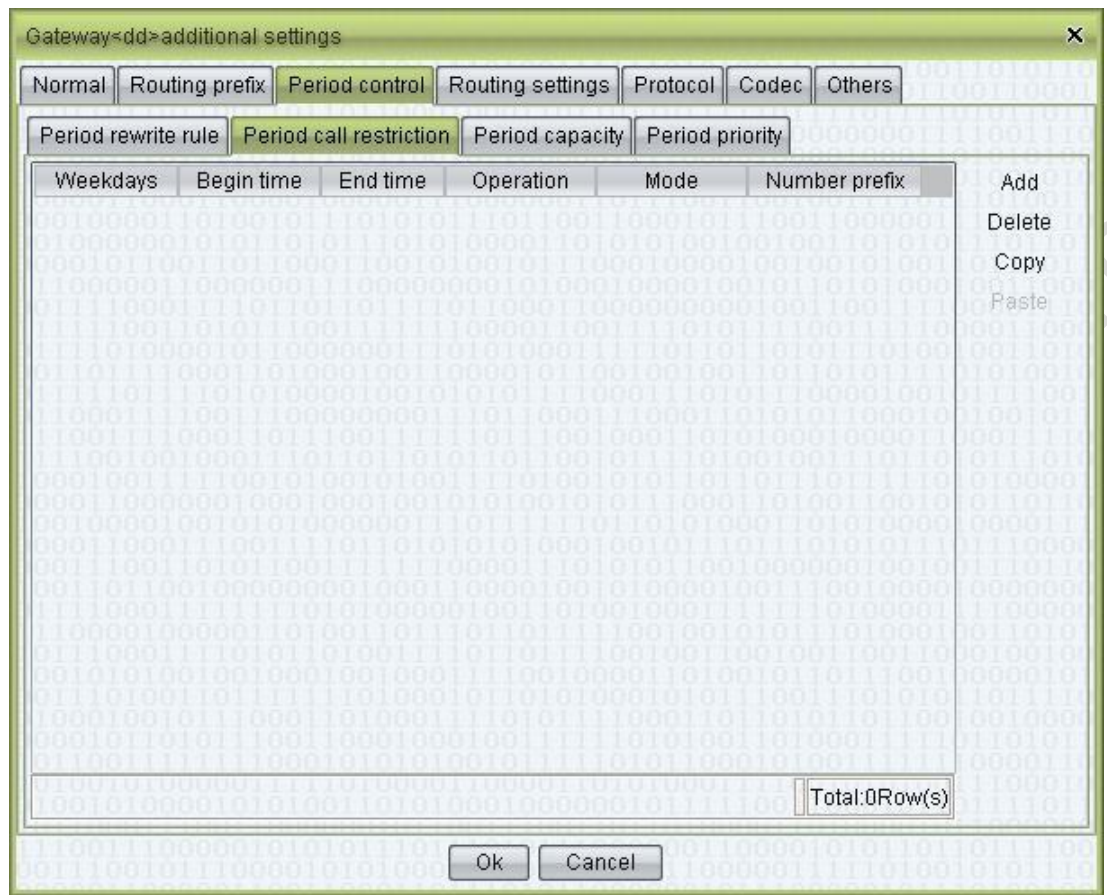
End time: time when the rewrite rule expires.

Mode

- The called: the rewrite rule applies to the called number.
- The caller: the rewrite rule applies to the caller number.

Rewrite rule: the content of the rule.

Additional settings > Period control > Period call restriction



Weekdays

Begin time: time when the rule comes into effect.

End time: time when the rule expires.

Operation

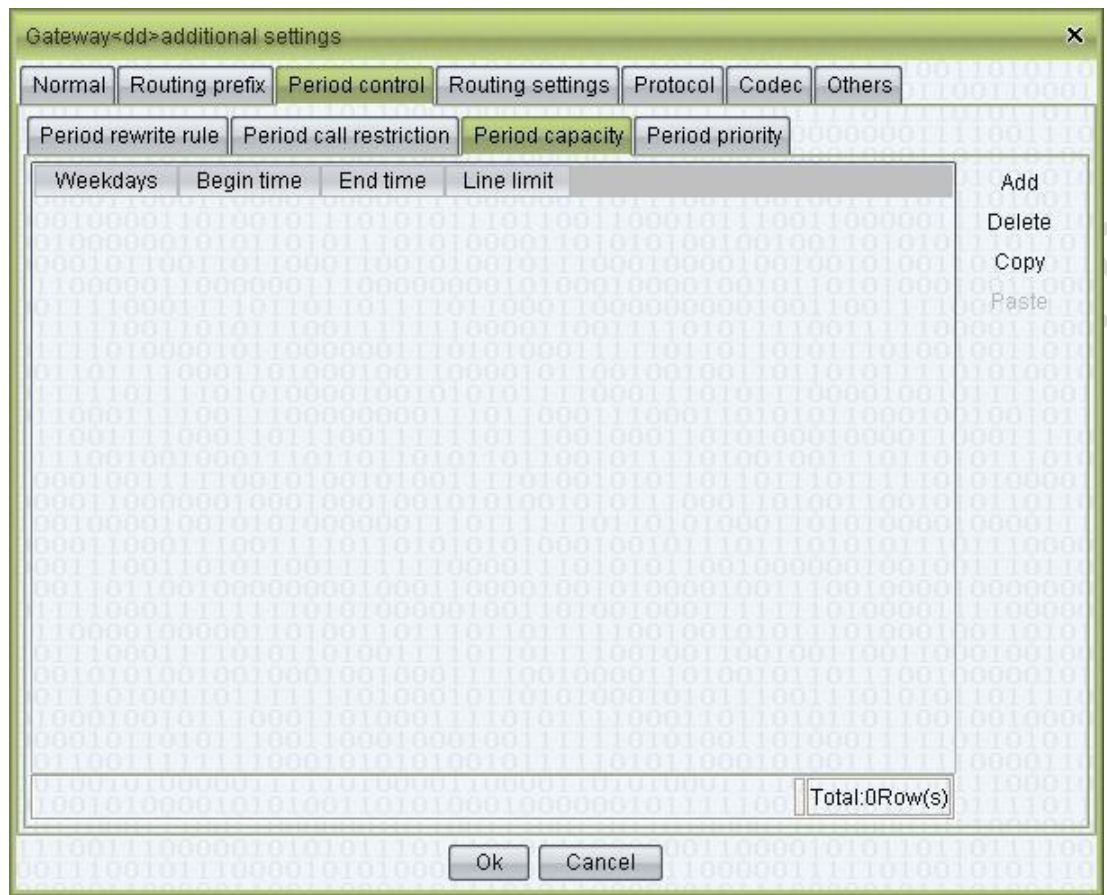
- Forbidden: forbidden operations for the matched prefixes.
- Allow: allow operations for the matched prefixes.

Mode

- By caller: matches the prefixes of the caller numbers.
- By callee: matches the prefixes of the called numbers.

Number prefix: the prefix of the number. Multiple prefixes can be specified, separated by commas.

Additional settings > Period control > Period capacity



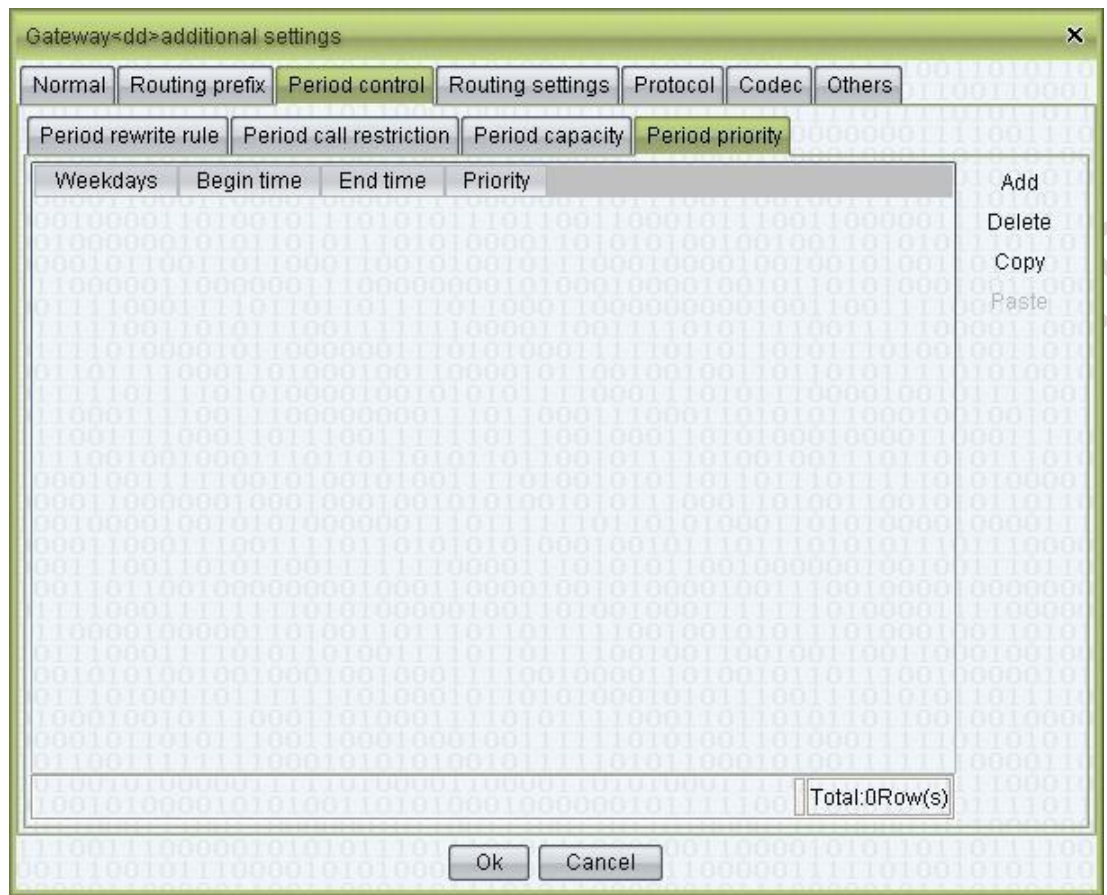
Weekdays

Begin time: time when the rule comes into effect.

End time: time when the rule expires.

Line limit: the maximum number of channels allowed for this period.

Additional settings > Period control > Period priority



Weekdays

Begin time: time when the rule comes into effect.

End time: time when the rule expires.

Priority: priority for this gateway for this period.

Additional settings > Period control > Routing settings

Add area code for mobile automatically: “Number management > Mobile area” is needed, default: Off. Softswitch will automatically add area code for mobile number.

Regional restriction for mobile: restrict callee e.g. allow 25, means caller can dial callee with prefix 21 only.

Call restriction on local city: e.g. allow 25, means only allow area 25 to dial local number.

Number restriction on caller: only allow specified caller number. “Number management > Mobile area” and “City code” is needed.

Number restriction on callee: only allow specified callee number.

NOTE

This function can reduce invalid numbers.

Additional settings > Protocol > H323

NOTE

Default value is set in "Operation management > Softswitch management > Additional settings > H323 parameter".

Setup/CallProceeding/Alerting/CallProceeding(OLC): if signal timeout, softswitch will try next gateway or hang up.

- Setup default:set by"Softswitch management > Additional settings > H323 parameter > SS_H323_TIMEOUT_SETUP".
- CallProceeding default:set by" Softswitch management > Additional settings > H323 parameter > SS_H323_TIMEOUT_CALLPROCEEDING".
- Alerting default:set by" Softswitch management > Additional settings > H323 parameter > SS_H323_TIMEOUT_ALERTING".
- CallProceeding(OLC) default:set by" Softswitch management > Additional settings > H323 parameter > SS_H323_TIMEOUT_CALLPROCEEDING_OLC".

Stop switch gateway after olc:

- Off: won't affect gateway switch.

- On: stop switch gateway after receive faststart or OpenLogicalChannel (H245).
- Default:set by” Softswitch management > Additional settings > H323 parameter > SS_H323_STOP_SWITCH_AFTER_OLC”.



NOTE

If “Switch gateway until connect” is on, this setting is invalid.

Q931 NumberingPlan: see H323’s RFC.

Q931 NumberType: see H323’s RFC.

Q931 PresentationIndicator: see H323’s RFC.

Q931 ScreeningIndicator: see H323’s RFC.

G729 negotiation mode:

- Auto: keep original G729 codec.
- G729: treat G729a or G729 as G729.
- G729a: treat G729 or G729a as G729a.
- G729&G729a: treat G729 or G729a as G729 and G729a.

FastStart: check to enable.

H245Tunneling: check to enable.

H245InSetup: check to enable.

Allow t38: check to enable.

Additional settings > Protocol > SIP



NOTE

Default value is set in "Operation management > Softswitch management > Additional settings > SIP parameter".

Invite/Trying/Ringing/SessionProgress(183)/ SessionProgress(SDP): if signal timeout, softswitch will try next gateway or hang up.

- Invite default:set by" Softswitch management > Additional settings > SIP parameter > SS_SIP_TIMEOUT_INVITE".
- Trying default:set by" Softswitch management > Additional settings > SIP parameter > SS_SIP_TIMEOUT_TRYING".
- Ringing default:set by" Softswitch management > Additional settings > SIP parameter > SS_SIP_TIMEOUT_RINGING".
- SessionProgress(183) default:set by" Softswitch management > Additional settings > SIP parameter > SS_SIP_TIMEOUT_SESSION_PROGRESS".
- SessionProgress(SDP) default:set by" Softswitch management > Additional settings > SIP parameter > SS_SIP_TIMEOUT_SESSION_PROGRESS_SDP".

Stop switch gateway after receive sdp:

- Off: won't affect gateway switch.
- On: stop switch gateway after receive SDP.
- Default: set by "Softswitch management > Additional settings > SIP parameter > SS_SIP_STOP_SWITCH_AFTER_SDP"



NOTE

If "Switch gateway until connect" is on, this setting is invalid.

Reply address: after receive sip request message, send reply signal to which address.

- Socket: send reply signal to request address (recommend).
- Via port: send reply signal to request IP and use port from Via. (Due to network reasons, some systems will use this mode.)
- Via: send reply signal to address from Via. (May have security risk.)

Request address: after call establish, send request signal to which address.

- Socket: send request signal to sender address (recommend).
- Contact Port: send request signal to sender IP and use port from Contact.
- Contact: send request signal to address from Contact.

G729 negotiation mode

- Auto: keep original G729 codec.
- G729: treat G729a or G729 as G729.
- G729a: treat G729 or G729a as G729a.
- G729&G729a: treat G729 or G729a as G729 and G729a.

G729 annexb

- Auto: send routing's G729 annexb setting to routing.
- yes: annex=yes
- no: annexb=no
- None: no annexb
- Transmit: send caller's G729 annexb setting to routing.

G723 annexa: refer to G729 annexb.

Enable timer protocol: enable routing gateway's timer protocol to detect over time.

Enable 100rel: send 1xx to routing gateway (intermediate state like 183).

Allow t38: allow send T38 signal to routing gateway.

Allow Retry-After: when routing gateway's signalling contain Retry-After domain, the values will determine the disabled length of gateway, disabled gateways are displayed in "Online routing gateway".

Allow Publish: this protocol can make routing gateway control concurrency automatically.

Enable call forward signal: if call from mapping gateway uses call forward format or phone's call forward, then use call forward signal format to send the call.



NOTE

After enable call forward signal, the outbound caller of CDR will be "number in To field | number in From field".

Peer number information: set select mode to SIP signal's caller.

- Display:
- Remote-Party-ID screen:

Privacy

- None: no this field
- Passthrough: send received settings to routing gateway
- Id: privacy field

P-Preferred-Identity

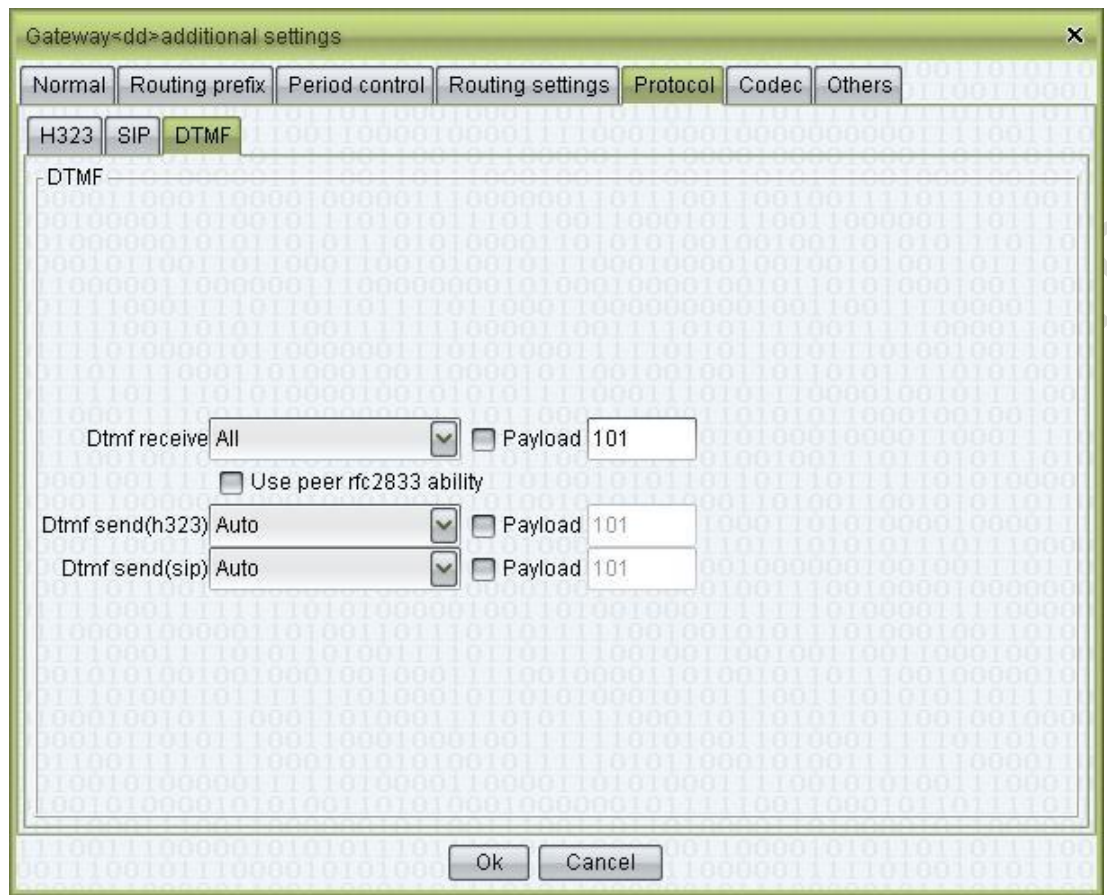
- None: no this field
- Passthrough: send received settings to routing gateway
- Caller: use caller number

P-Asserted-Identity

- None: no this field
- Pass through: send received settings to routing gateway
- Caller: use caller number

Caller rewrite rule: rewrite rules for the caller number in "P-Asserted-Identity" field.

Additional settings > Protocol > DTMF



Dtmf receive: specifies the way by which DTMF signals are received. The <All> option is recommended, which asks the system to accept all kinds of DTMFs. Once a certain kind of DTMF is received, this channel will accept the same kind of DTMFs only, thus effectively avoiding duplicate receptions.

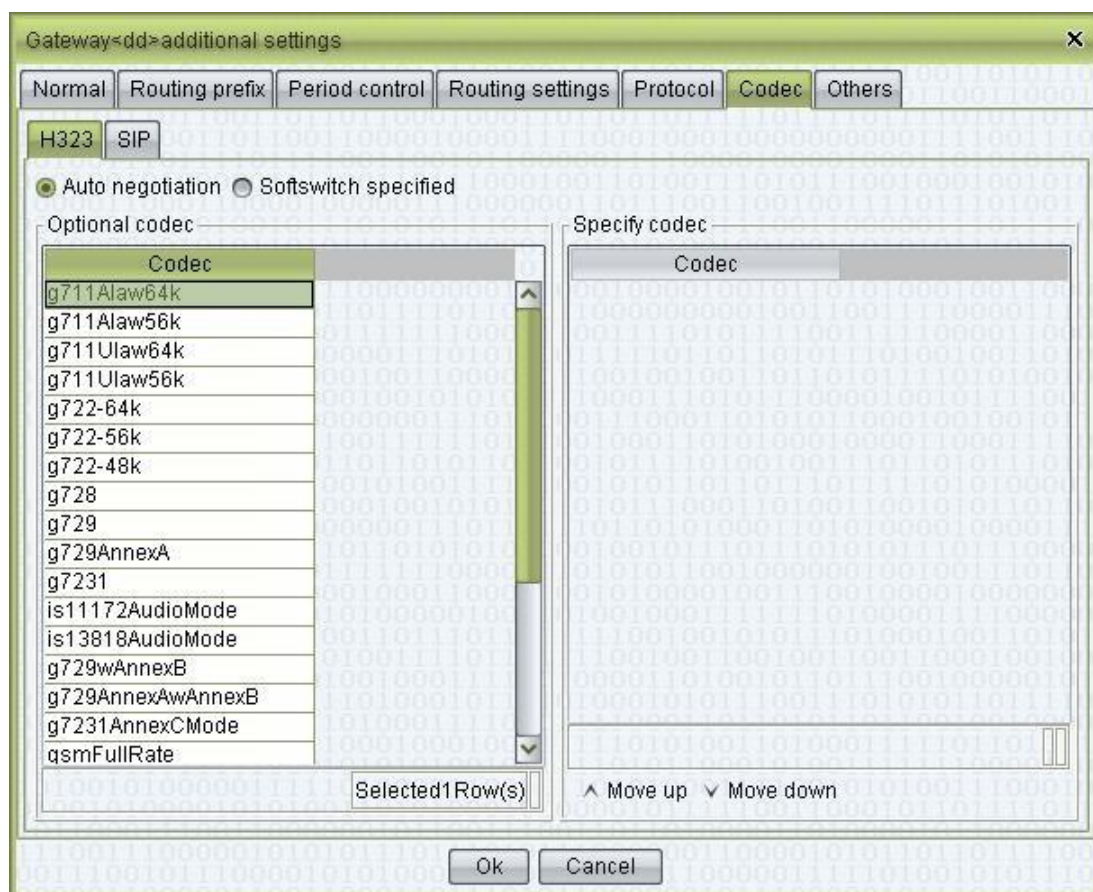
Use peer rfc 2833 ability: rfc 2833 ability is determined by peer ability.

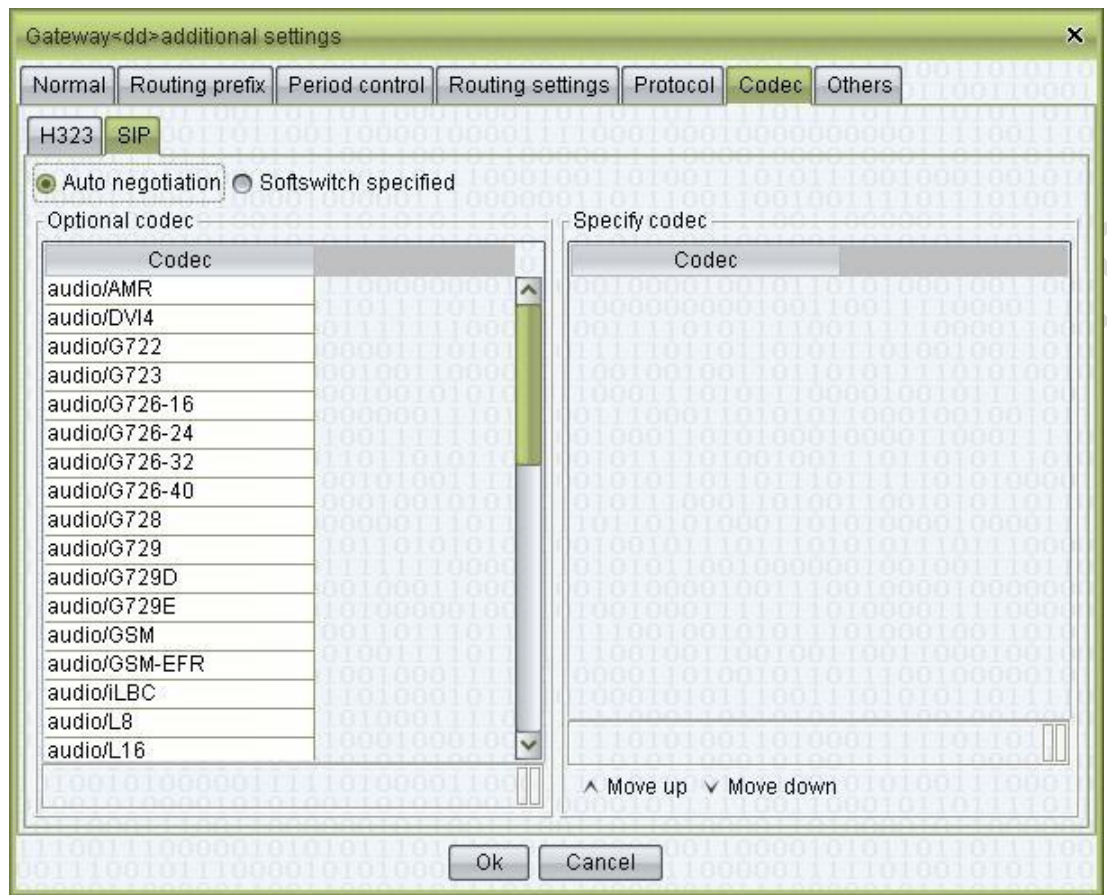
Payload: the payload value in RTP, for the DTMF of the RFC2833 mode.

Dtmf send(h323): it is set to "Auto" by default, indicating that the system would determine the best way to send DTMFs based on the receiver's capacity. If the receiver provides no capacity set, the system will send according to the default mode. The RFC2833 mode can only be specified for media proxy.

Dtmf send(sip): it is set to "Auto" by default. The details are the same as those of "Dtmf send(h323)" described above.

Additional settings > Codec > H323/SIP





Auto negotiation: determined by caller and callee.

Softswitch specified: codec can only be sent.

TIP

Select one codec, and then double click to add or delete.

Select some codecs, and then right click to add or delete.

NOTE

For specified codec, "Softswitch specified" is needed.

Additional settings > Others

Gateway<testgw>additional settings

Normal Routing prefix Period control Routing settings Protocol Codec Others

Number length limit

Caller number allow length

Callee number allow length

Number restriction on callee: None ☐ Callee transform ☐

☒ Enable phone display number

☐ Set to protect route

☐ Enable bilateral reconciliation ☐ Passthrough rtp loss rate

☐ Enable forwarding signal rewrite pool

☐ Rate limit: Max calls every 1000 ms

Signaling tracing

☐ Call tracing ☐ Register tracing End time: 2017-05-04 00:00:00

Ok Cancel

Number length limit

Caller number allowable length: the lengths of the caller numbers allowed to pass through the gateway (e.g. fill in "11, 14" to allow numbers of 11 digits or 14 digits only.)

Callee number allowable length: the lengths of the called numbers allowed passing through the gateway.

NOTE

Left blank to allow numbers of all length to pass through, and fill in "0" to allow no numbers to pass through.

Number restriction on callee:

- None: no restriction.
- Phone number: allow call platform's phone numbers only.
- Other number: allow call numbers except platform's phone number.

Callee transform

Enable phone display number: when caller is phone, check to use phone's display number, uncheck to use phone number.



NOTE

Gateway's rewrite rule will still be used.

Set to protect route: if set to protect route, this gateway will not be used to sort with other gateways. Between protect routes, sort order as normal. If normal gateways are not connected, the call will switch to protect routes. Protect route enable time can be set, within the time call is not connected, will be send to protect routes.

Allow bilateral reconciliation: VOS will check the amount deviation of customer and vendor automatically.

Passthrough rtp lost rate: transparent media message to peer platform

Enable forwarding signal rewrite pool: use number in pool as caller.



NOTE

Number sequence is decided by <FORWARD_SIGNAL_REWRITE_SEQUENCE> setting in config file softswitch.conf

0: random

1:poll

Default is random

Rate limit: Restrict the number of call-establish signaling in a certain period.

Signaling tracing: Setting this gateway, call tracing, register tracing, end time .This setting will not affected by the system<Debug trace> setting.

Other Operations

Double-click the content of "Routing clearing account name" to open the account management page for this account.

Right-Click Menu

Current call: open the list of "Current call" for this gateway

Condition monitoring: monitor the gateway status and show the connection information of this gateway in the "Gateway status" page.

Connect analysis: open "Connect analysis" sheet for this gateway.

Interrupt analysis: open "Interrupt analysis" sheet for this gateway.

Alarm monitoring: open "Routing gateway alarm" settings for selected gateways.

Total gateway: get total capacity of the selected gateways.

Filling downwards

Random generate: to generate password.

Downloaded from VOS3000.Com

2.5.1.2 Mapping Gateway

This function is used to manage mapping gateway.

Gateway name	Lock type	Authorization type	Gateway group	Line limit	Routing gateway group	Routing gateway name
test	No lock	Domestic		30/Allow		Allow

How to Start

Double-click “Navigation > Operation management > Gateway management > Mapping gateway”

Table Items

Gateway name: see the descriptions in “Routing gateway”.

Lock type: see the descriptions in “Phone management”.

Authorization type: see the descriptions in “Phone management”.

Gateway group: set the gateway’s group, used for control several gateways’ total currency.

Line limit: the number of concurrent calls allowed by the gateway.

Routing gateway group: allow or forbidden.

Routing gateway name: Set mapping gateway allow/forbidden to try those routing gateways.

Softswitch: see the descriptions in "Routing gateway".

Additional settings

Ip: see the descriptions in "Additional settings".

Account id: the number of the billing account for this mapping gateway.

Account name: the name of the billing account for this mapping gateway.

Configuration password: see the descriptions in "Routing gateway".

Self service password: see the descriptions in "Routing gateway".

Priority: for static gateway with the same IP, use this priority to match.

Caller black/white list group: see the descriptions in "Routing gateway".

Callee black/white list group: see the descriptions in "Routing gateway".

Memo: comments on this gateway.

Other Operations

Double-click the "Account name", can directly enter to the manage interface of this account.

Right-Click Menu



Same as "routing gateway"

Routing analysis.

Additional settings > Normal

Gateway<mapping-1>additional settings

Normal Mapping prefix Period control Routing settings Protocol Codec Others

Gateway type: Dynamic

Ip:

Process timeout(seconds): Default Protect route enable time(seconds): Default

Conversation limitation(seconds): Default Rtp interrupt detection: None

Media proxy: Auto Recording: ☐

☐ Lowest profit rate limit: 0 % ☐ Max minute rates: 0.0000000

Real time computing asr: Default Real time computing acd: Default

First routing strategy: None

Second routing strategy: None

Ok Cancel

Gateway type

- Dynamic: registration is required.
- Static: gateway mapping is achieved directly through IP addresses.

Ip: IP addresses of mapping gateways. Multiple addresses and signaling port can be specified, separated by commas.

Process timeout(seconds): the maximum time waited after the call has reached the gateway. If the connection has not been establish within the time limit, the system server will send a reject signal to the mapping gateway.

- Default: set by "Operation management > Softswitch management > Additional settings > System parameter > SS_MAPPINGGATEWAYTIMEOUT".
- None: no limitation.

Protect route enable time(seconds):

Conversation limitation(seconds)

- Default: set by "Operation management > Softswitch management > Additional settings > System parameter > SS_MAXCALLDURATION".
- None: no limitation.

Media proxy: see the descriptions in “Routing gateway”.

Recording: record the call

RTP interrupt detection: see the descriptions in “Routing management”.

Lowest profit rate limit: this gateway will be locked if profit below this value.

Max minute rates: max minute rate allowed for callee number.

Real time computing asr: see the descriptions in “Routing management”.

Real time computing acd: see the descriptions in “Routing management”.

First routing strategy

- None: System default.
- Asr: Sort by ASR.
- Lowest rate per second: sort by rate per second.

Second routing strategy: see “First routing strategy”.

Additional settings > Mapping prefix

Gateway<mapping-1>additional settings

Normal Mapping prefix Period control Routing settings Protocol Codec Others

Mapping caller prefix
☒ Allow ☐ Forbidden

Mapping callee prefix
☒ Allow ☐ Forbidden

Caller rewrite rule

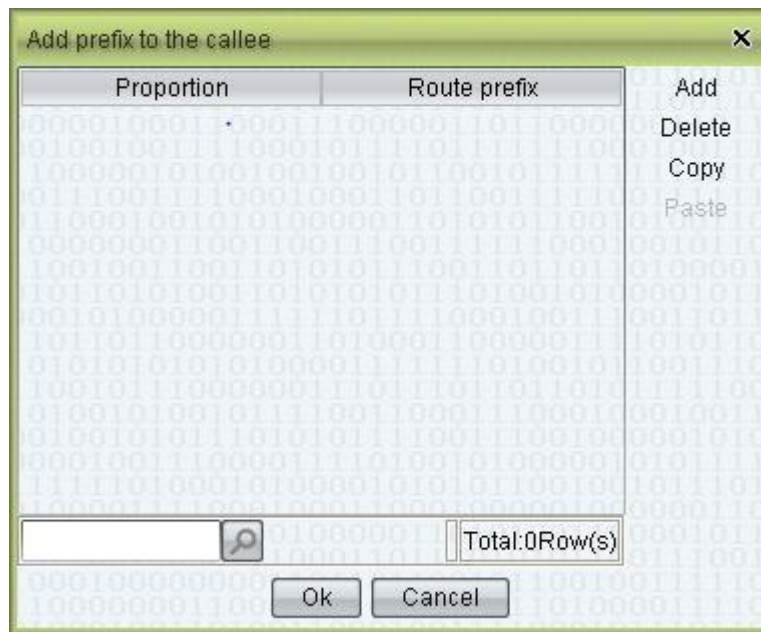
Callee rewrite rule

Add prefix to the callee

Ok Cancel

See the descriptions in “Routing gateway”

Add prefix to the callee: add prefix for callee number by proportion



NOTE

When “callee rewrite rule” and “add prefix to callee” are present rules for same callee number at the same time, callee will be applied by rule of “callee rewrite rule” at first, then rewritten number will be added prefix by proportion.

Additional settings > Period control

Gateway<test>additional settings

Normal Mapping prefix **Period control** Routing settings Protocol Codec Others

Period call restriction Period capacity Period rewrite rule

Weekdays	Begin time	End time	Operation	Mode	Number prefix	
						Add
						Delete
						Copy
						Paste

Total: 0 Row(s)

Ok Cancel

See the descriptions in “Routing gateway”.

Additional settings > Routing settings

Gateway<test>additional settings

Normal Mapping prefix Period control **Routing settings** Protocol Codec Others

Add area code for mobile automatically

☐ On ☒ Off

Regional restriction for mobile

☒ Allow ☐ Forbidden

Call restriction on local city

☒ Allow ☐ Forbidden

Number restriction on caller

☐ On ☒ Off

☒ Allow city number ☒ Allow mobile number ☐ Allow uncertainty number

Number restriction on callee

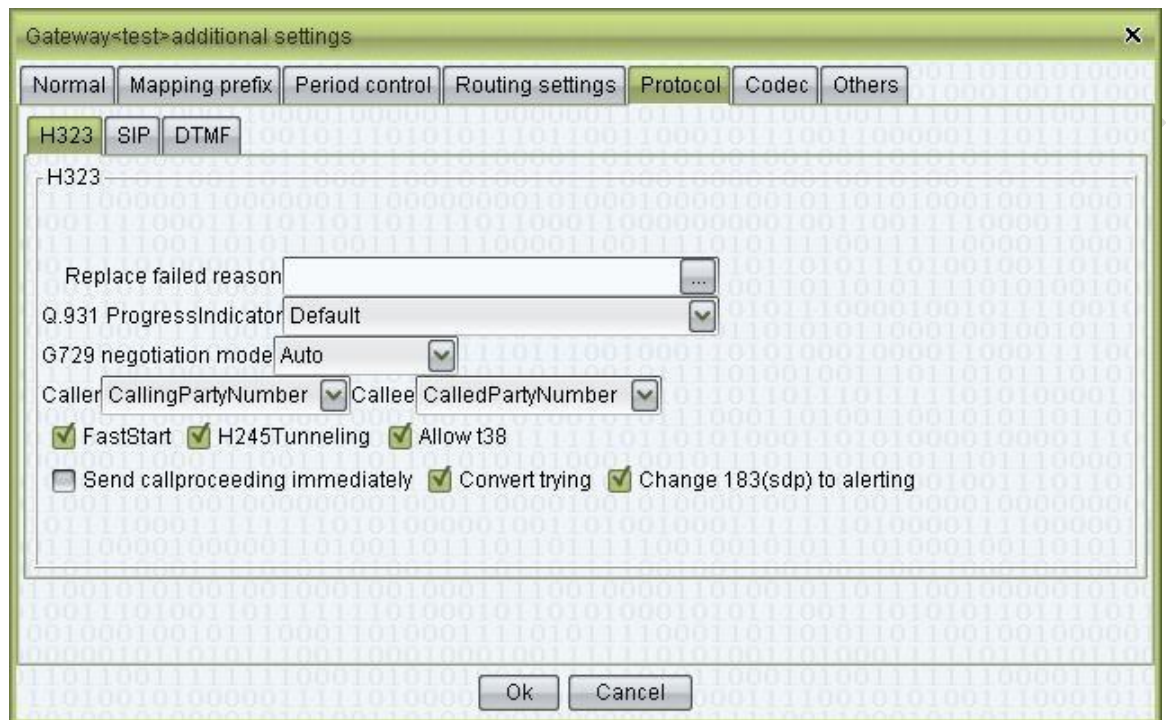
☐ On ☒ Off

☒ Allow city number ☒ Allow mobile number ☐ Allow uncertainty number

Ok Cancel

See the descriptions in “Routing gateway”.

Additional settings > Protocol > H323



Replace failed reason: specify the error message sent to the mapping gateway when the call cannot be established, can define their replacement rules according to different termination reason.



Q.931 ProgressIndicator: see the standard reference for H323. See the descriptions in “Phone management” for details about signaling checkboxes.

G729 negotiation mode

Caller: CallingPartyNumber/SourceAddress/Display

Callee: CalledPartyNumber/DestinationAddress

FastStart: check to enable.

H245Tunneling: check to enable.

Allow t38: check to include T38.

Send callproceeding immediately: check to send callproceeding when receives setup.

Convert trying: check, if caller uses H323 and callee uses SIP, when callee returns Trying, VOS will send callproceeding to caller.

Change 183(sdp) to alerting: check, if caller uses H323 and callee uses SIP, when callee returns 183 with SDP, VOS will send alerting to caller; Uncheck, VOS will send callproceeding to caller.

Additional settings > Protocol > SIP

Gateway<test>additional settings

Normal Mapping prefix Period control Routing settings Protocol Codec Others

H323 SIP DTMF

SIP

Reply address	Via Port	Request address	Contact Port
Caller	From	Callee	To
Display	Default	G723 annexa	Auto
G729 negotiation mode	G729	G729 annexb	Auto

☒ Enable timer protocol ☐ Enable 100rel ☒ Allow t38 ☐ Support privacy

☐ Recognize call forward signal

Replace failed reason

Remote ring back mode Passthrough

Call authentication mode Ip

Ok Cancel

See the descriptions in "Routing gateway".

Replay address

Request address

Caller: get caller number from which field of signal.

- From
- Remote-Party-ID
- Display

Callee: get callee number from which field of signal.

- To

- Request-Line

Support Privacy: pass through mapping gateway private domain.

Recognize call forward signal: call with forward format will be identified as forward call

Replace failed reason: specify the error message sent to the mapping gateway when the call cannot be established, can define their replacement rules according to different termination reason.



Remote ring back mode

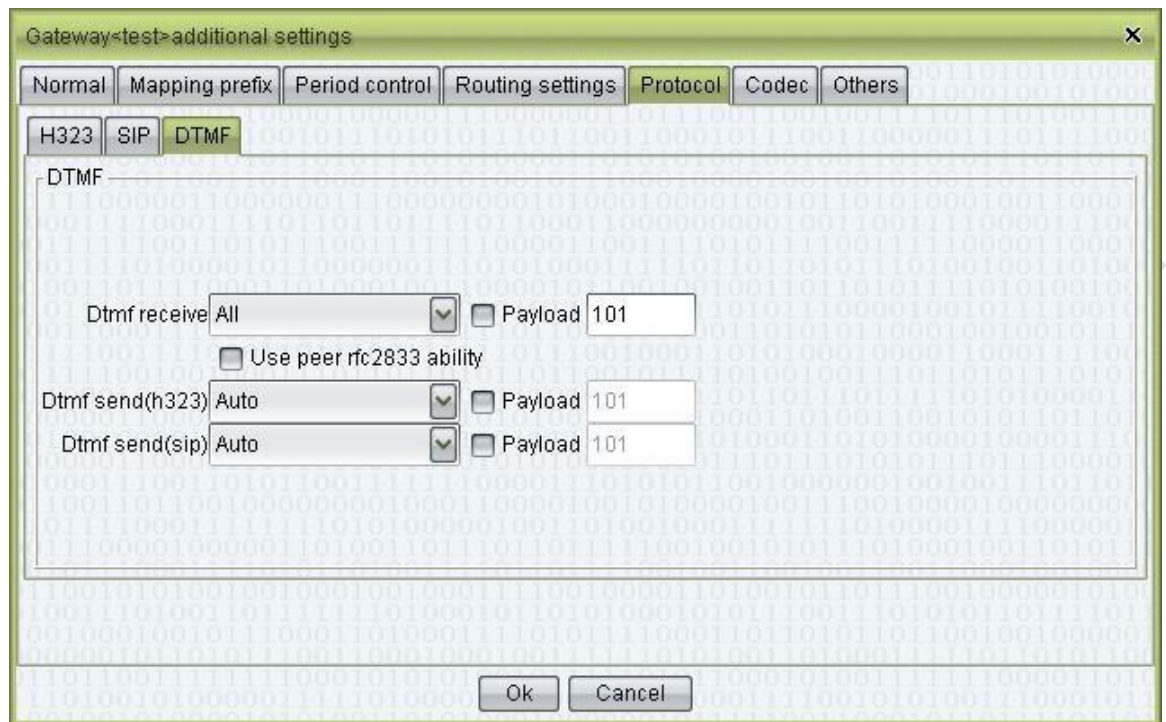
- Passthrough: according to routing gateway.
- 183 Session Progress + SDP: when callee ringing, send 183 Session Progress + SDP to mapping gateway.
- 180 Alerting + SDP: when callee ringing, send 180 Alerting + SDP to mapping gateway.

Call authentication mode

- IP
- IP Address and Port
- Authenticate by Password

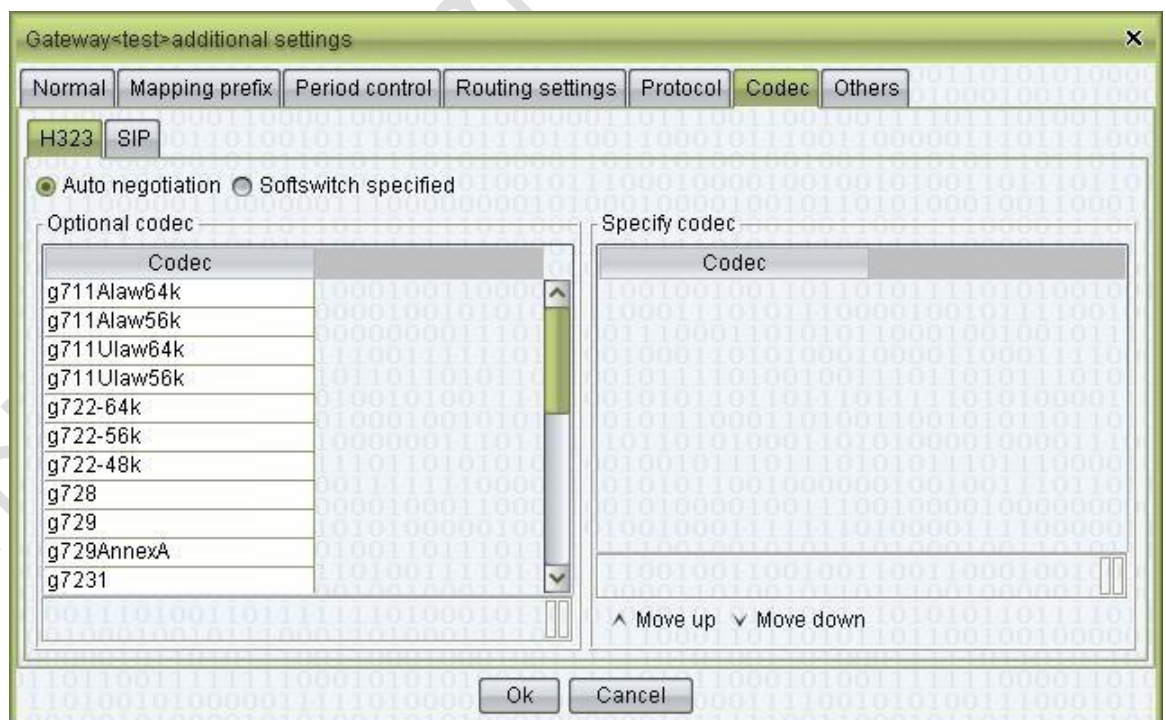
Recognize call forward signal: if mapping gateway send calls in forward signal format, do recognize.

Additional settings > Protocol > DTMF



See the descriptions in “Routing gateway”.

Additional settings > Codec



See the descriptions in “Routing gateway”.

Additional settings > Others

Gateway<mapping-1>additional settings

Normal Mapping prefix Period control Routing settings Protocol Codec Others

Number length limit

Caller number allow length

Callee number allow length

☐ Caller transform

☐ Allow bind number billing ☐ Allow phone number billing ☐ Enable phone settings

☐ Audio prompts for can't connect Language name

☐ Not recorded no hold time cdr

☐ Allow bilateral reconciliation ☐ Passthrough rtp loss rate

Number restriction on callee None

☐ Enable forwarding signal rewrite pool

☐ Rate limit Max calls every 1000 ms

Signaling tracing

☐ Call tracing ☐ Register tracing End time 2017-05-04 00:00:00

Ok Cancel

See the descriptions in "Routing gateway".

Caller number allow length: allowed length of caller number (eg. fill in "11, 14", means the gateway only allow 11-digit caller number and 14-digit caller number to pass through)

Callee number allow length: allowed length of callee number

Caller transform: use number in "Number Transformation" table to replace caller ID.

Allow bind number billing: if caller is bind number, use account that bind number belongs to billing.

Allow phone number billing: if caller is platform's phone number, use account that phone number belongs to billing.

Audio prompts for can't connect: if calls fail to connect, system will broadcast the failure reason

Language name: Chinese/English

Not recorded no hold time cdr: system will not record the cdr with no conversation time

Number restriction on callee

- None: no restriction
- Phone number: only platform's phone number is allowed
- Other number: only number not platform's number is allowed

Allow bilateral reconciliation: VOS will check the amount deviation of customer and vendor automatically.

Passthrough rtp lost rate: transparent media message to peer platform

Enable forwarding signal rewrite pool: use number in pool as caller.

Query lrn



NOTE

In American, location routing number (LRN) is saved in signal control point (SCP). LRN is used for number portability.

LRN server can be set in "Softswitch management > Additional settings > System parameter > SS_LRN_SERVER_IP" and "SS_LRN_SERVER_PORT".

Eat prefix length: cut the number, then use left to billing.

Failure action: when query failed, "Reject" or "Continue".

Routing using number:

Interstate billing prefix: add prefix for interstate call.



NOTE

Identify of interstate can be refereed to "Number Management > Lerg Data", if caller and callee's state are different, the call is interstate.

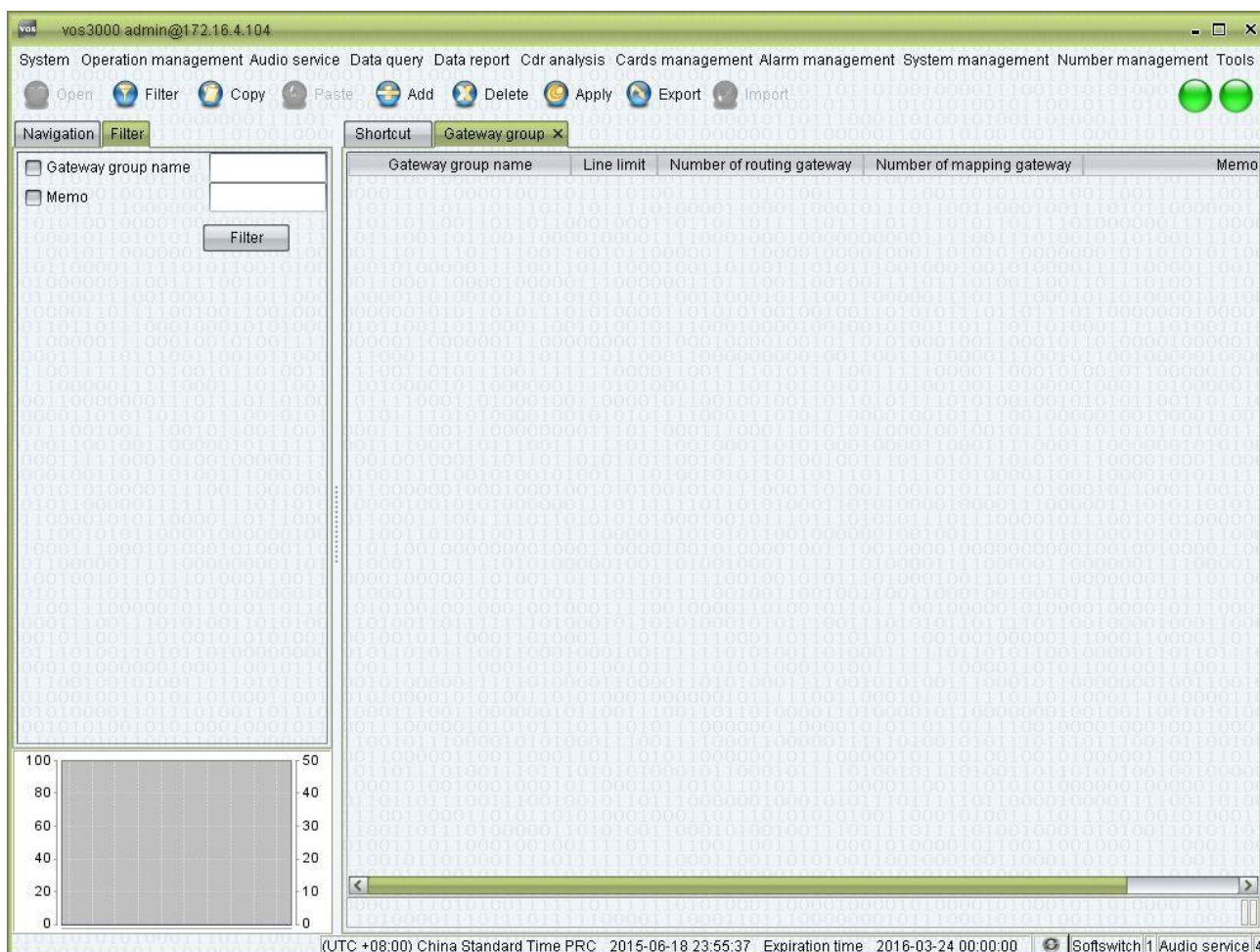
Undetermined billing prefix: add prefix for the calls which cannot be recognized interstate.

Rate limit: Restrict the number of call-request in a certain period.

Signaling tracing: Setting this gateway, call tracing, register tracing, end time. This setting without restriction of the system <Debug trace> setting.

2.5.1.3 Gateway Group

This function is used to define gateway groups, which is used to limit gateways during routing.



How to Start

Double-click “Navigation > Operation management > Gateway management > Gateway group”

Table Items

Gateway group name: the name of gateway group.

Line limit: total capacity of the gateway group.

None: means lines of all gateways included in the group are all available.

Use gateway setting: means gateway’s line limit setting will be used.

Number of routing gateway: double-click to manage.

Number of mapping gateway: double-click to manage.

Memo



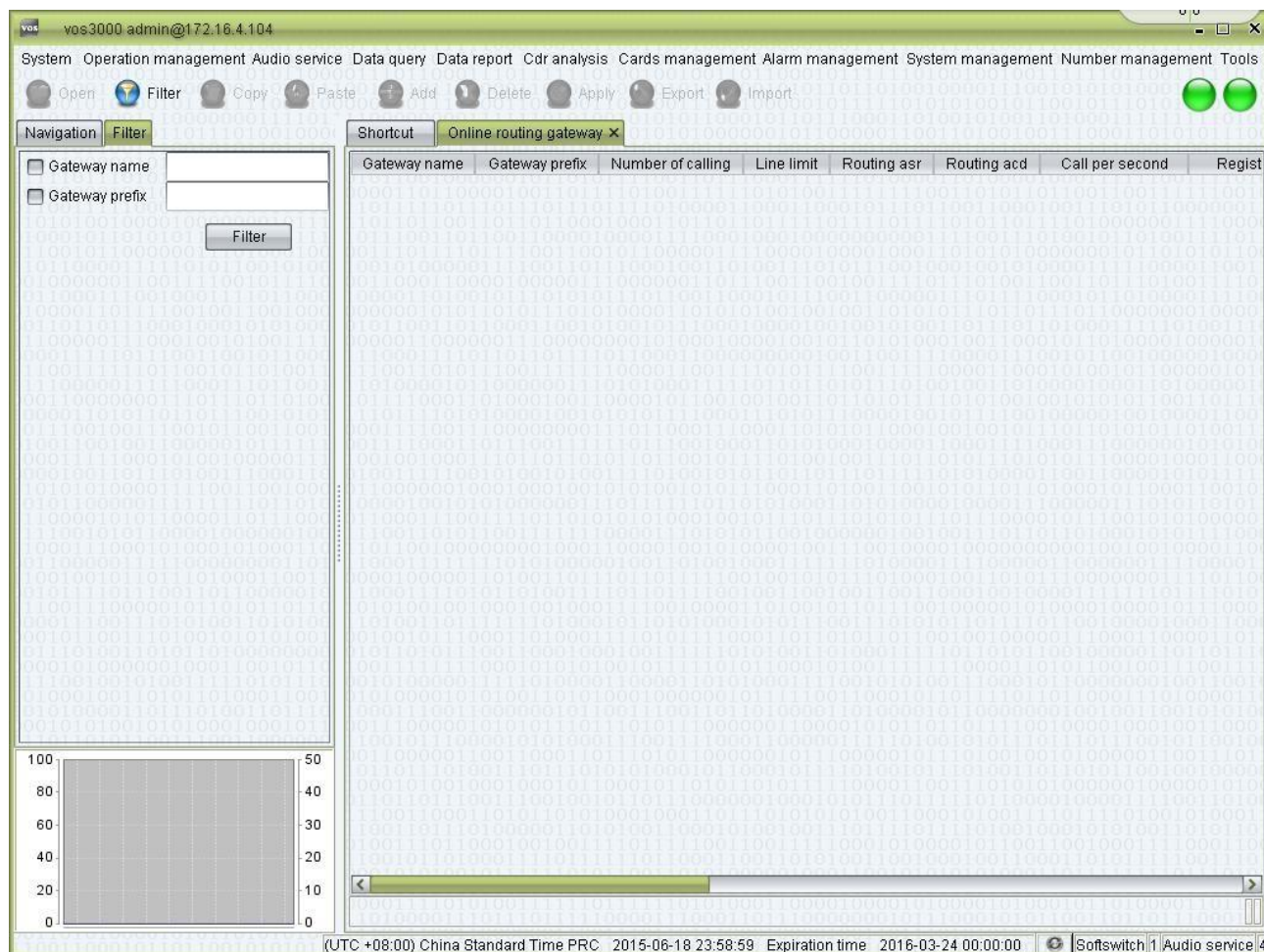
NOTE

If phone or mapping gateway's Allow gateway groups is deleted totally, means forbidden all routing gateways.

If phone or mapping gateway's Forbidden gateway groups is deleted totally, means allow all routing gateways.

2.5.1.4 Online Routing Gateway

This function is used to query online routing gateway.



How to Start

Double-click "Navigation > Operation management > Gateway management > Online routing gateway"

Table Items

Gateway name: the device id of gateway.

Gateway prefix: the prefix of gateway.

Number of calling: the number of current sessions maintained by the gateway and the total capacity of it.

Line limit

Routing asr

Routing acc

Call per second: Display the current call rate when routing gateway <rate limit> is turned on .

Registered ip: the current IP of the gateway.

Registration time: the server time of the platform's most recent registration.

Update time: the time of the most recent confirmation that the platform is online.

Duration: the time elapsed since the most recent registration (for dynamic gateways). (* There is no "Time elapsed" item for static gateways.)

Encryption type: the type of encryption used by the gateway.

Register name

Disabled time

Call tracing

Register tracing

Local ip

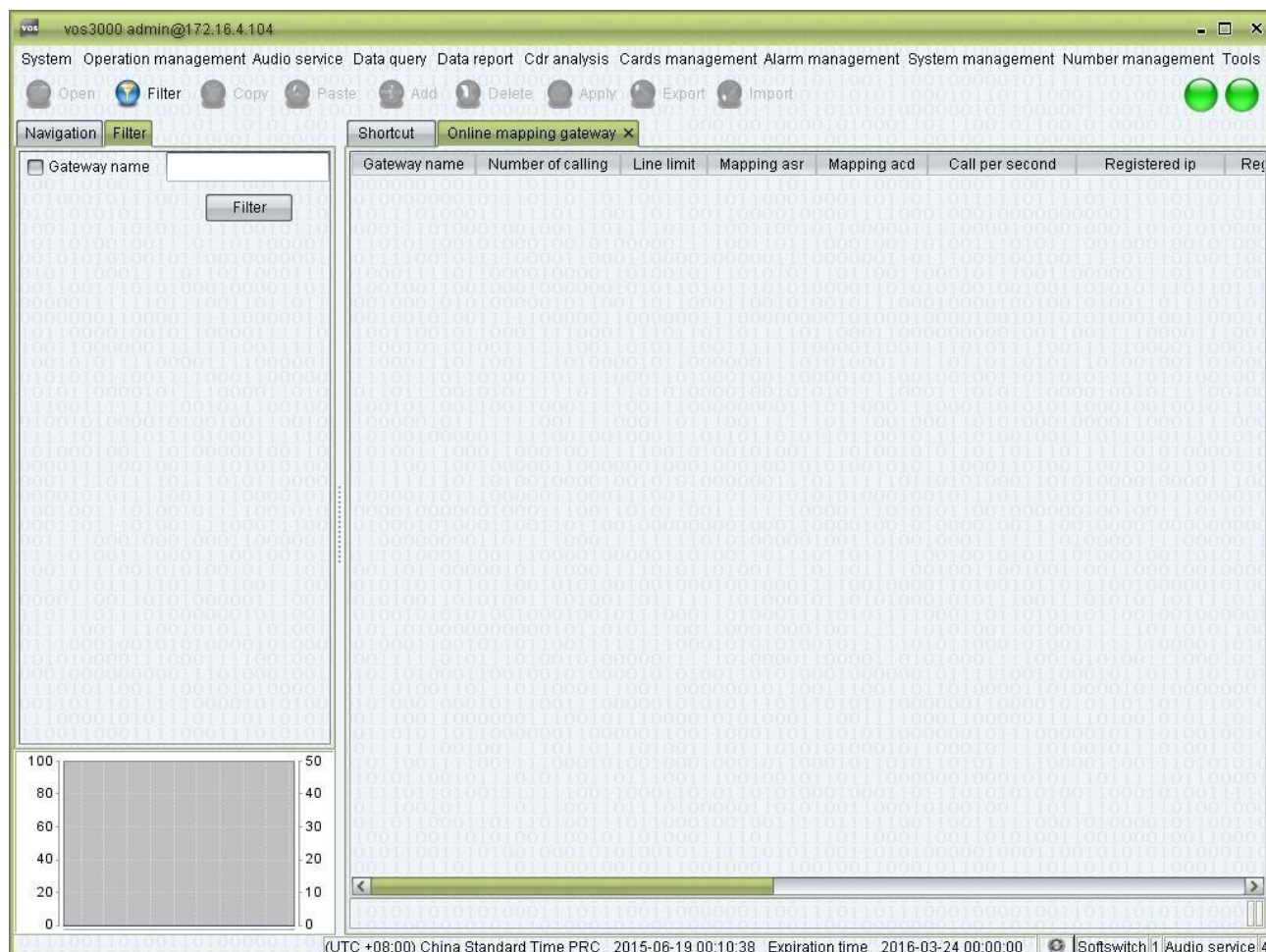
Softswitch name: the name of the softswitch that the gateway registered.

Right-Click Menu

Current call: open the "Current call" page for this gateway.

2.5.1.5 Online Mapping Gateway

This function is used to query online mapping gateway.



How to Start

Double-click "Navigation > Operation management > Gateway management > Online mapping gateway"

Table Items

Gateway name: the device ID of gateway.

Number of calling

Line limit

Mapping asr

Mapping acd

Call per second: Display the current call rate when routing gateway <rate limit> is turned on .

Registered ip: the current IP of the gateway.

Registration time: the server time of the platform's most recent registration.

Update time: the time of the most recent confirmation that the platform is online.

Duration: the time elapsed since the most recent registration (for dynamic gateways).



NOTE

There is no "Time elapsed" item for static gateways.

Encryption type: the type of encryption used by the gateway.

Call tracing

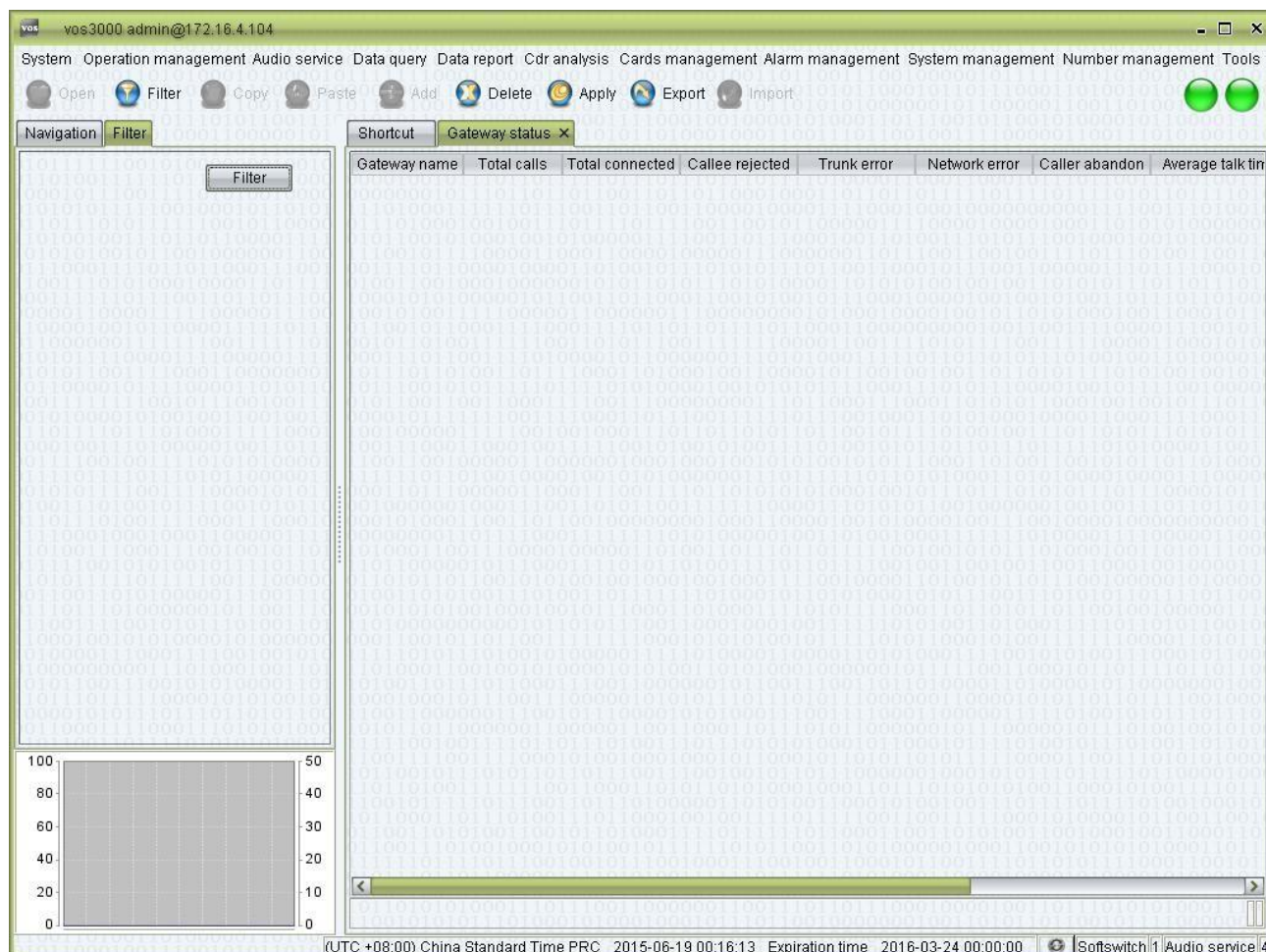
Register tracing

Local ip

Softswitch name: the name of the softswitch that the gateway registered.

2.5.1.6 Gateway Status

This function is query gateway status.



How to Start

Double-click “Navigation > Operation management > Gateway management > Gateway status”

Table Items

Gateway name

Total calls

Total connected

Callee rejected

Trunking error

Network error

Caller abandon
Average talk time
Total talk time
lp
Starting time

Downloaded from VOS3000.Com

2.5.2 Phone Operation

2.5.2.1 Phone Management

This function is used to manage phone.

Phone number	Configuration password	Display caller id	Lock type	Authorization type	Monthly consumption	Month
100	100	100	No lock	Domestic	0.000	
15687964569	1	15687964569	No lock	Domestic	0.000	
200	200	200	No lock	Domestic	0.000	
20004	984077	20004	No lock	Domestic	0.000	
300	300	300	No lock	Domestic	0.000	
400	400	400	No lock	Domestic	0.000	
4001	4001	Remote-Party-ID	No lock	Domestic	0.000	
4002	4002	Remote-Party-ID	No lock	Domestic	0.000	
4003	4003	Remote-Party-ID	No lock	Domestic	0.000	
4011	4011	4011	No lock	Domestic	0.140	
4012	4012	4012	No lock	Domestic	0.000	
4013	4013	4013	No lock	Domestic	0.000	
4014	4014	4014	No lock	Domestic	0.000	
4015	4015	Remote-Party-ID	No lock	Domestic	0.000	
4016	4016	4016	No lock	Domestic	0.000	
5001	23761649	5001	No lock	Domestic	0.000	
5002	23599800	5002	No lock	Domestic	0.000	
5003	85149112	5003	No lock	Domestic	0.000	
5004	30209796	5004	No lock	Domestic	0.000	
8001	8001	Remote-Party-ID	No lock	Domestic	0.000	
8002	8002	Remote-Party-ID	No lock	Domestic	0.000	
8003	8003	Remote-Party-ID	No lock	Domestic	0.000	
8004	84423017	8004	No lock	Domestic	0.000	
8012	8012	Remote-Party-ID	No lock	Domestic	0.000	
9001	9001	9001	No lock	Domestic	0.000	
9002	9002	9002	No lock	Domestic	0.970	
9005	9005	4014	No lock	Domestic	0.000	
9006	9006	4014	No lock	Domestic	0.000	

How to Start

Double-click "Navigation > Operation management > Phone operation > Phone management"

Table Items

Phone number: the number used as caller ID and the called number for the terminal.

Configuration password: the password used for terminal registration (For H32, will be the H323ID)

Display caller id: the caller ID shown at the called end.

- “Remote-Party-ID” to use the number of the original caller.
- “Display” to use number in “From” field of SIP message.

Lock type:

- No lock: none restrictions to the terminal.
- Bar outgoing: the terminal is not allowed to call out.
- Bar incoming: the terminal is not allowed to be called.
- Bar all calls: the terminal is denied from any service.

Authorization type: when a call is initiated by this number, this type will be compared with the “Rate Type” of the rate. If the “Rate type” of the matching rate has higher precedence than the type specified here, the call will be denied.



NOTE

The precedence of authorization is: International call > Domestic call > Local call > Net call.

Monthly consumption

Monthly min: consumption at least per month.



NOTE

At the beginning of the month, system will calculate the cost of last month. If the phone's consumption less than the value, system will take off the difference.

Monthly max: consumption at most per month.

Monthly service fee: rent fee.



NOTE

Take off at the beginning of month.

Billing rate: private rate of the phone.

Supplementary service

Advanced configuration

Rewrite rule: rewrite rules for the called number.



TIP

Rewrite rules can be used to implement dial local number without add area code, please refer to “rewrite rule” chapter.

Routing gateway group: allow routing gateways.

Account id: editable. Designating the billing account for this terminal.

**NOTE**

Rate of this account will be used to bill and perform authorization checks upon calling. The operator that edits this number must have authorizations to manipulate phones of the concerned account.

Account name: non-editable. When the account number is correctly set, the corresponding name of the account will be shown here.

**NOTE**

Changing the account number will cause changes to the account and the number segment restrictions of its agent.

Double-click to the account management page for this account.

DID/DDI: after the phone on line, the other numbers allowed as caller ID or callee number.

**NOTE**

Number only as signaling transfer, not as billing and reporting data.

Softswitch

Reverse charging: if "On", any call to the number will be charge on this account, instead of the caller.

Self service password: the password used by users to login from the web and query bills. When left blank, the "Configuration password" will be used.

Call in limitation: when phone is called, maximum of incoming calls.

Call out limitation: when phone is calling, maximum of outgoing calls.

Line limit: the maximum number of channels for this phone, which limits the maximum sum of incoming and outgoing calls processed simultaneously.

Directory: number of phone's phonebook.

Directory limit: max number of phone's phonebook.

Caller black/white list group: see the descriptions in "Routing gateway".

Callee black/white list group: see the descriptions in "Routing gateway".

Memo: descriptions to this phone.

Supplementary service:

Display caller id: display the caller's ID.

Call transfer: forward the calls.

Do not disturb: reject all calls.

Call forwarding on period: forward calls in specified time period.

Call forwarding unconditional: forward all calls to specified number.

Call forwarding no reply: forward calls to specified number when the call is not answered or the phone is out of connection.

Offline forward: forward calls when phone is not online

Call forwarding on busy: forward incoming calls when the phone is busy.

Forward display number:

- Default: Use "Softswitch management > Additional settings > System parameter > SS_CALL_FORWARD_USING_ORIGINAL_CALLER".
- Original caller: display caller's number.
- Local number: display the phone's number.

Normal call transfer display:

- Default: Use "Softswitch management > Additional settings > System parameter > SS_CALL_TRANSFER_NORMAL_DISPLAY".

- Original caller: display caller's number.
- Local number: display the phone's number.

Ask call transfer display:

- Default: Use "Softswitch management > Additional settings > System parameter > SS_CALL_TRANSFER_ASK_DISPLAY".
- Original caller: display caller's number.
- Local number: display the phone's number.

Polyphonic ringtone: audio file should be imported to "Audio service > Value added > Phone polyphonic ringtone".

Language: Chinese/English.

Audio prompts for can't connect:

- Default: Use "Softswitch management > Additional settings > System parameter > SS_CALL_FAILED_INDICATION".
- On: if calls failed connect, system will broadcast the failure reason
- Off: if calls failed connect, system will hang up the phone directly

Insufficient alarm:

- Default: Use "Softswitch management > Additional settings > System parameter > SS_ACCOUNT_INDICATION_METHOD".
- Prompt balance: Use "Softswitch management > Additional settings > System parameter > SS_ACCOUNT_INDICATION_MONEY"
- Prompt duration: Use "Softswitch management > Additional settings > System parameter > SS_ACCOUNT_INDICATION_TIME"

Spots remaining duration:

- Default: Use "Softswitch management > Additional settings > System parameter > SS_CALL_FAILED_INDICATION".
- On: Use "Softswitch management > Additional settings > System parameter > SS_CALL_REMAIN_TIME_INDICATION"
- Off: system will not broadcast the remaining duration

Advanced configuration > Normal

Phone<0601>advanced configuration

Normal | Number prefix | Routing settings | Protocol | Codec | Period control | Audio service | Others

Register type: Dynamic (dropdown)
Protocol: SIP (dropdown)
Ip: (text field)
Signaling port: 5065 (text field)
Local ip: Default (dropdown)
Local port: Default (dropdown)
Signaling encryption: None (dropdown)
Media proxy: Auto (dropdown) | Rtp interrupt detection: None (dropdown) | ☐ Recording
Conversation limitation(seconds): Default (dropdown)
☐ Lowest profit rate limit: 0 % | ☐ Max minute rates: 0.0000000
First routing strategy: None (dropdown)
Second routing strategy: None (dropdown)
Ok Cancel

See the descriptions in “Routing gateway”.

Register type

- Static: use IP to access.
- Dynamic: use register to access.
- Mapping gateway: use gateway id to access.

Protocol

Ip

Signaling port

Local ip

Media proxy

Rtp interrupt detection

Recording: record the call

Conversation limitation(seconds)

Free billing to phone

- Default: set by "Operation management > Softswitch management > Additional settings > System parameter > SS_NO_BILLING_TO_PHONE".
- On: if caller is platform's phone, the call will be free.
- Off: Even caller is platform's phone, the call will still be charged.

Number restriction on callee

Display caller short e164

Use caller phone display id when number not in phonebook

Try to use routing gateway when phone is offline: when the phone being called is offline, try to find a matching route in the routing gateway.

- Default: set by "Operation management > Softswitch management > Additional settings > System parameter > SS_REDIRECTOFFLINEPHONETOGW".
- On: try routing gateway.
- Off: won't try routing gateway.

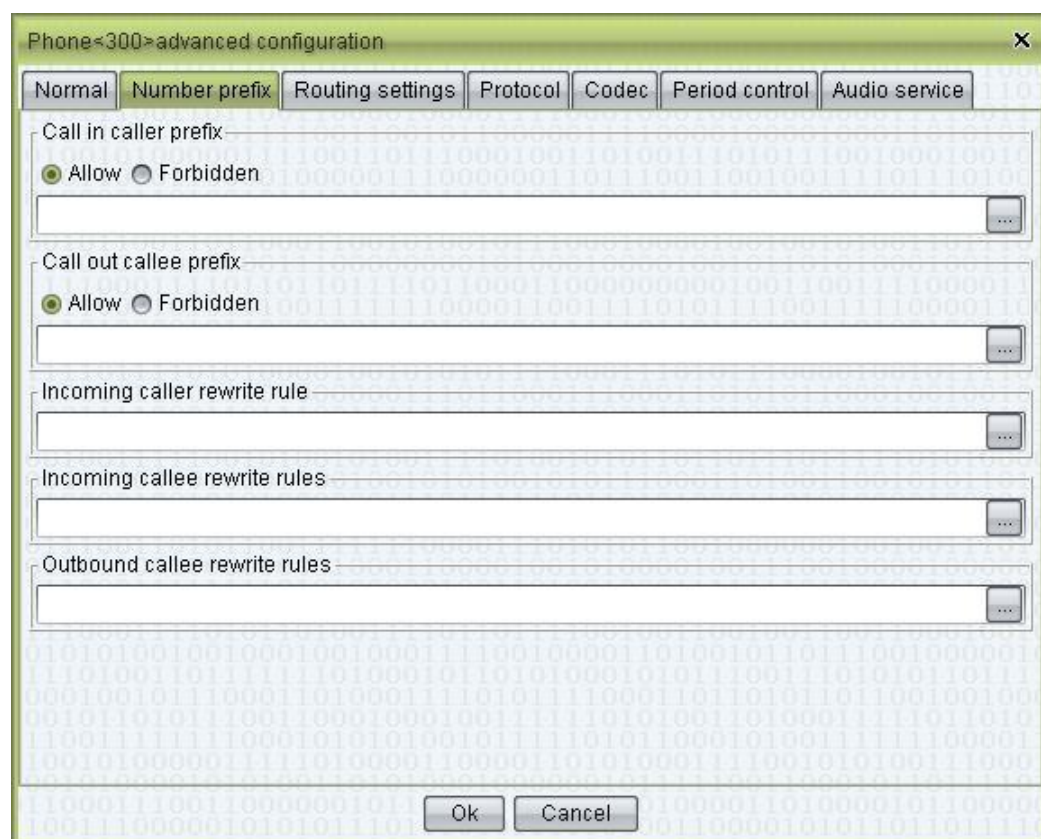
Lowest profit rate limit

Max minute rates

First routing strategy

Second routing strategy

Advanced configuration > Number prefix



The image shows a software configuration window titled "Phone<300>advanced configuration". It has a tabbed interface with the following tabs: "Normal", "Number prefix" (which is selected), "Routing settings", "Protocol", "Codec", "Period control", and "Audio service".

Under the "Number prefix" tab, there are five sections, each with a text input field and a button with three dots (indicating a list or search function):

- Call in caller prefix:** Includes radio buttons for "Allow" (selected) and "Forbidden".
- Call out callee prefix:** Includes radio buttons for "Allow" (selected) and "Forbidden".
- Incoming caller rewrite rule:**
- Incoming callee rewrite rules:**
- Outbound callee rewrite rules:**

At the bottom of the window are "Ok" and "Cancel" buttons.

Advanced configuration > Routing settings

Phone<300>advanced configuration

Normal Number prefix **Routing settings** Protocol Codec Period control Audio service

Add area code for mobile automatically
☐ On ☒ Off

Regional restriction for mobile
☒ Allow ☐ Forbidden

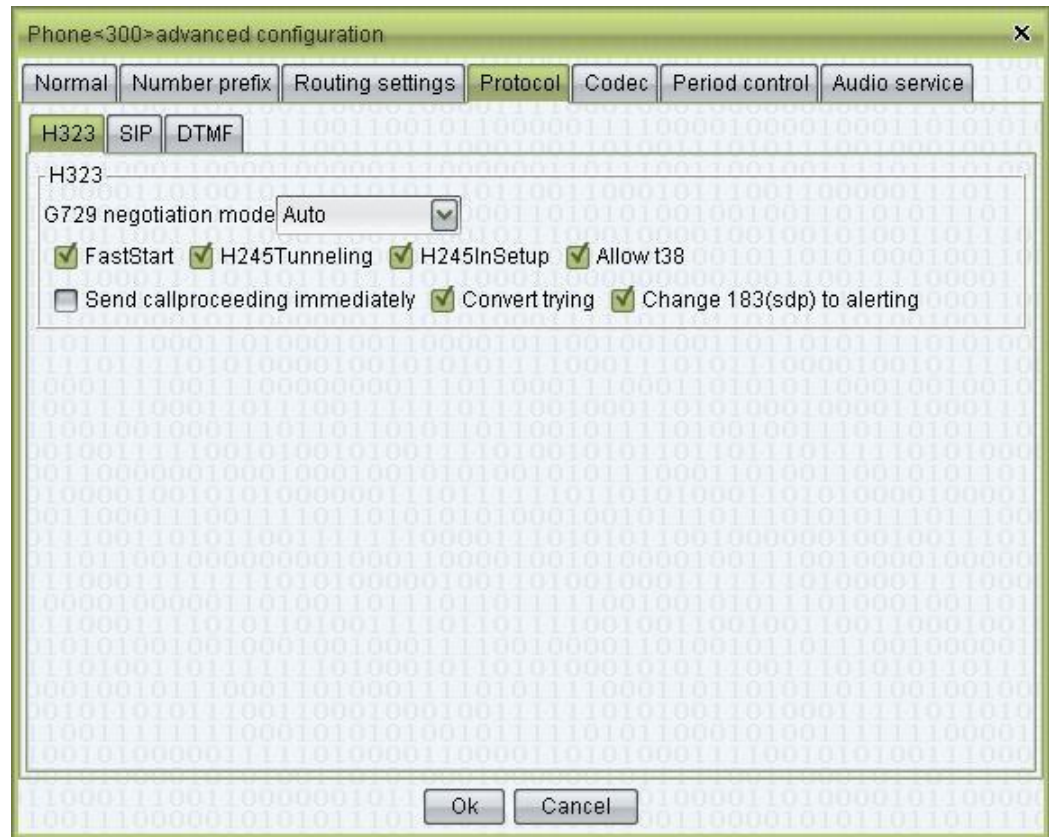
Number restriction on caller
☐ On ☒ Off
☒ Allow city number ☒ Allow mobile number ☐ Allow uncertainty number

Number restriction on callee
☐ On ☒ Off
☒ Allow city number ☒ Allow mobile number ☐ Allow uncertainty number

Ok Cancel

See the descriptions in “Routing gateway”.

Advanced configuration > Protocol > H323



See the descriptions in "Mapping gateway".

Advanced configuration > Protocol > SIP

Phone<300>advanced configuration

Normal Number prefix Routing settings **Protocol** Codec Period control Audio service

H323 **SIP** DTMF

SIP

Reply address Via Port Request address Contact Port

Display Default G723 annexa Auto

G729 negotiation mode G729 G729 annexb Auto

☒ Enable timer protocol ☐ Enable 100rel ☒ Allow t38 ☐ Support privacy

Peer number information

☐ Display Caller

☐ Remote-Party-ID screen None

Privacy None P-Preferred-Identity None P-Asserted-Identity None

Remote ring back mode Passthrough

Call authentication mode Ip

Ok Cancel

See the descriptions in “Mapping gateway”.

Advanced configuration > Protocol > DTMF

Phone<300>advanced configuration

Normal Number prefix Routing settings Protocol Codec Period control Audio service

H323 SIP DTMF

DTMF

Dtmf receive All ☐ Payload 101

☐ Use peer rfc2833 ability

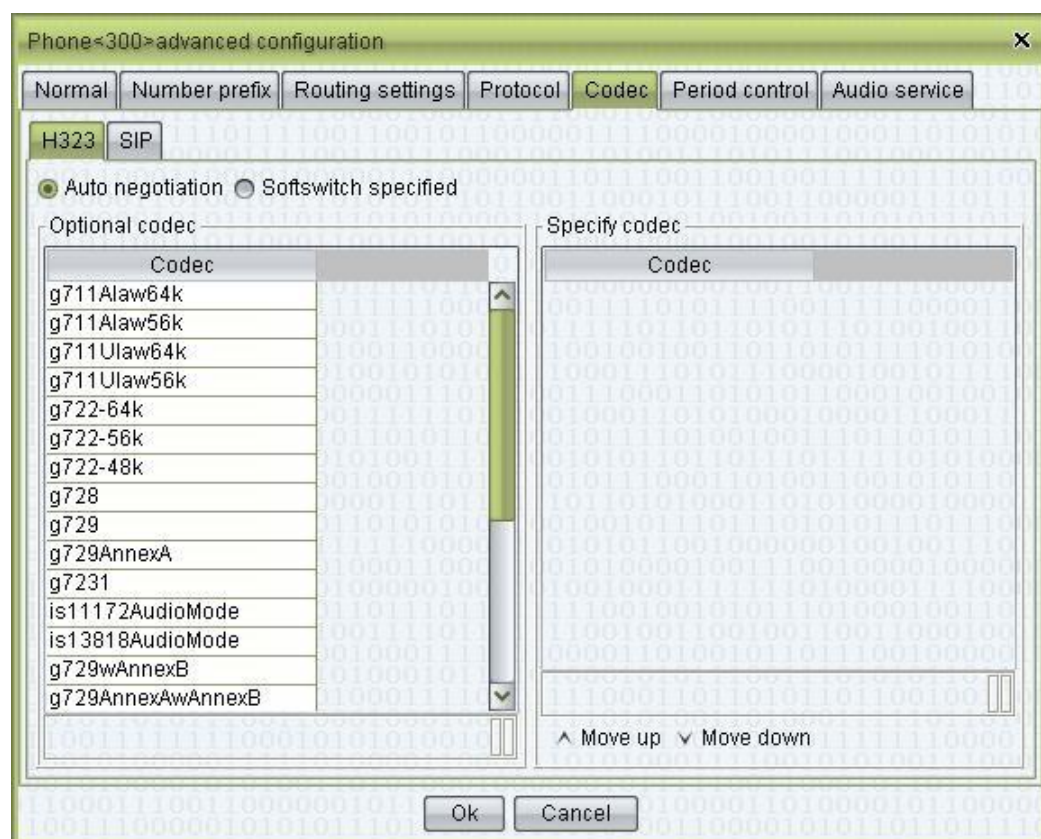
Dtmf send(h323) Auto ☐ Payload 101

Dtmf send(sip) Auto ☐ Payload 101

Ok Cancel

See the descriptions in “Routing gateway”.

Advanced configuration > Protocol > Codec



See the descriptions in "Routing gateway".

Advanced configuration > Period control

Phone<300>advanced configuration

Normal Number prefix Routing settings Protocol Codec **Period control** Audio service

Period call restriction Period capacity Period rewrite rule

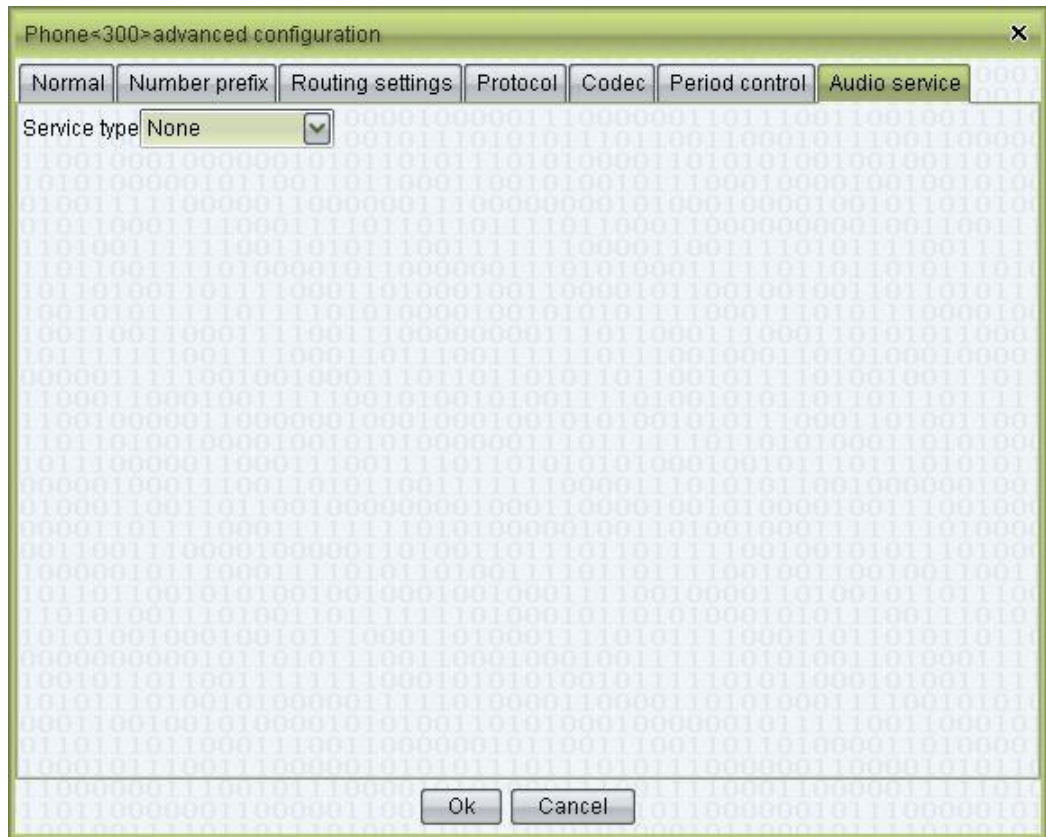
Weekdays	Begin time	End time	Operation	Mode	Number prefix	
						Add
						Delete
						Copy
						Paste

Total: 0 Row(s)

Ok Cancel

See the descriptions in "Routing gateway".

Advanced configuration > Audio service



Other Operations

Double-click the content of "Account name" to open the account management page for this account.

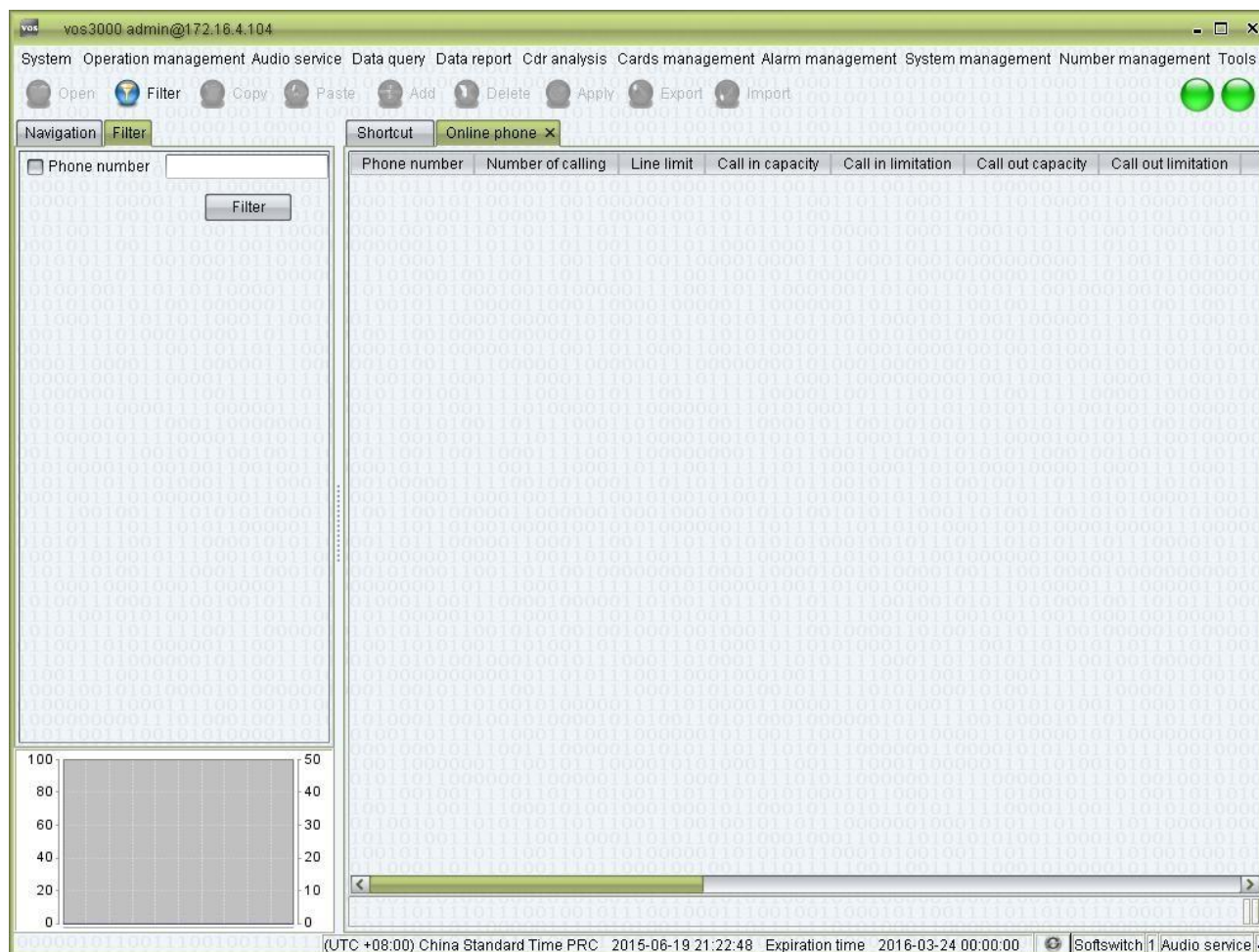
2.5.2.2 Online Phone

This function is used to query online phone.



NOTE

Static phone is not shown.



How to Start

Double-click "Navigation > Operation management > Phone operation > Online phone"

Table Items

Phone number: the number used by the terminal at registration (used as the caller ID and the called number).

Number of calling

Line limit

Call in capacity

Call in limitation

Call out capacity

Call out limitation

Device id: Model of the device.

Protocol: the protocol used at registration (SIP or H323).

Registered ip: the remote address of the terminal used at the registration. If the connection is establish through a firewall, this address may be a local address.

Registration time: the server time of the terminal's most recent registration.

Update time: the time of the most recent confirmation that the terminal is online.

Duration: the time elapsed since the most recent registration.

Encryption type: the type of encryption used by the gateway.

Call tracing

Register tracing

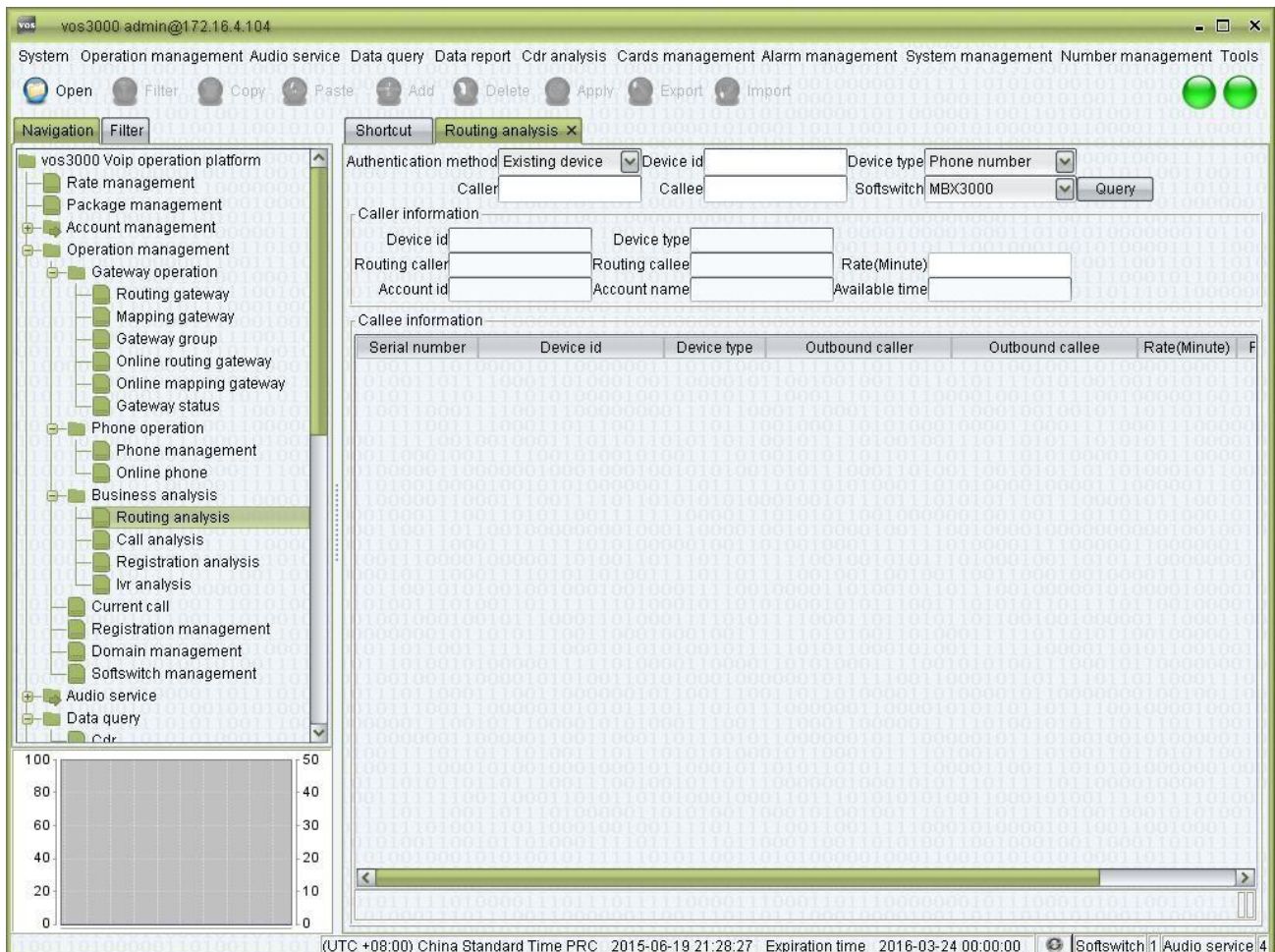
Local ip

Softswitch name

2.5.3 Business Analysis

2.5.3.1 Routing Analysis

This function is used to simulate the routing process of phone or mapping gateway.



How to Start

Double-click "Navigation > Operation management > Business analysis > Routing analysis"

Input Items

Authentication method:

- Existing device: phone or gateway.
- Static ip: IP of mapping gateway.

Device id: phone number or gateway ID depends on device type.

Device type:

- Phone number
- Mapping gateway

Caller

Callee

Softswitch

Output Items

Caller information

Device id: actual caller device.

Device type

Routing Caller: caller after caller device's rewrite rule.

Routing Callee: callee after caller device's rewrite rule.

Rate(Minute): calculate by billing account's fee rate.

Account id: billing account.

Account name

Available time: maximum duration.

Callee information

Serial Number: routing sequence.

Device id

Device type

Outbound caller: caller after callee device's rewrite rule.

Outbound callee: callee after callee device's rewrite rule.

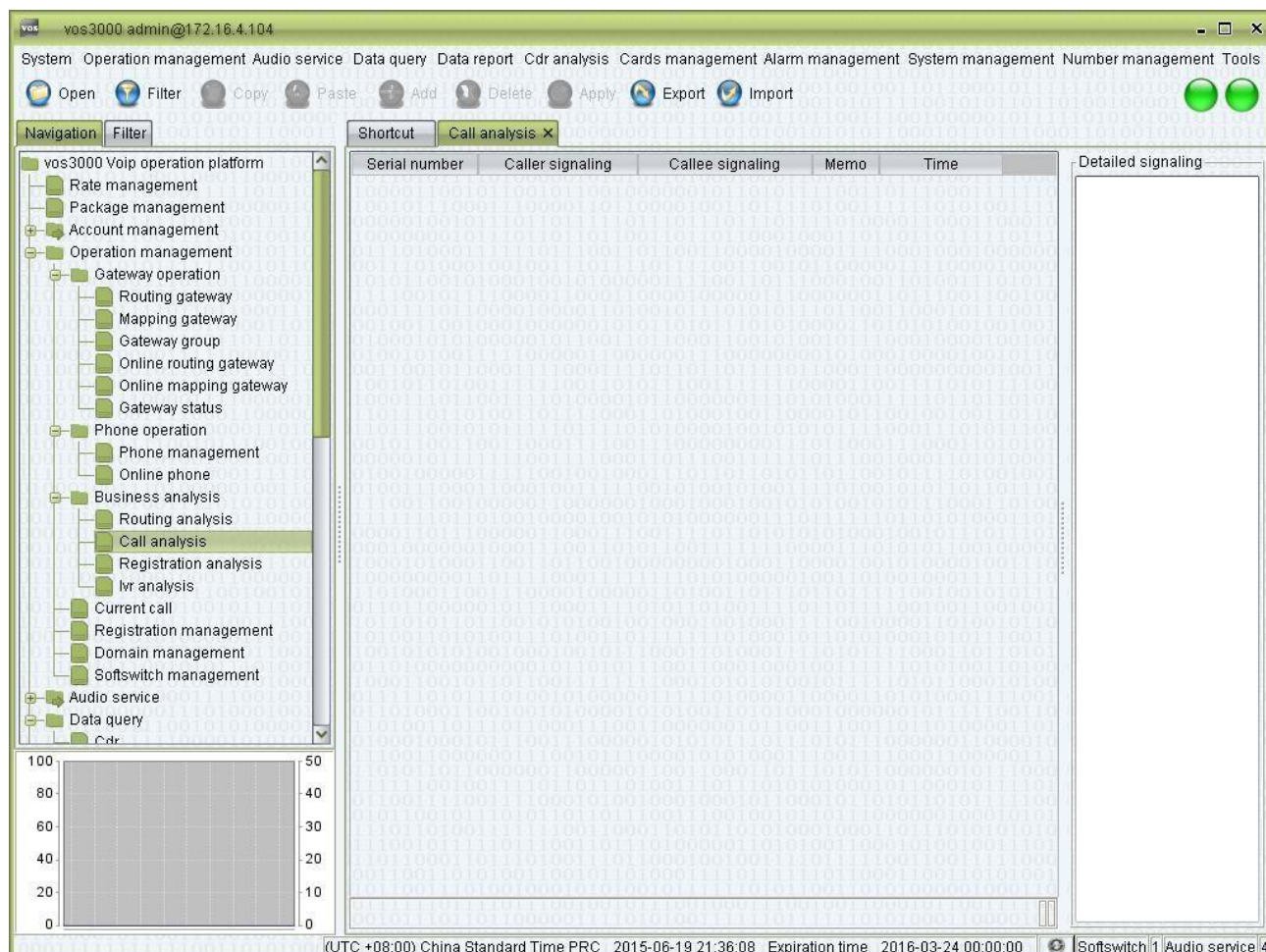
Rate(Minute): calculate by callee device's fee rate.

Rate deviation: difference between caller device's fee rate and callee device's cost.

Detail: like prefix, priority, score and so on.

2.5.3.2 Call Analysis

This function is used to analysis call problem.



How to Start

Double-click “Navigation > Operation management > Business analysis > Call analysis”

Table Items

Serial number: the serial number of signaling interaction.

Caller signaling: content of signaling interaction with caller.

Callee signaling: content of signaling interaction with callee.

Memo: message of softswitch.

Time: time of signaling.

Other Operations

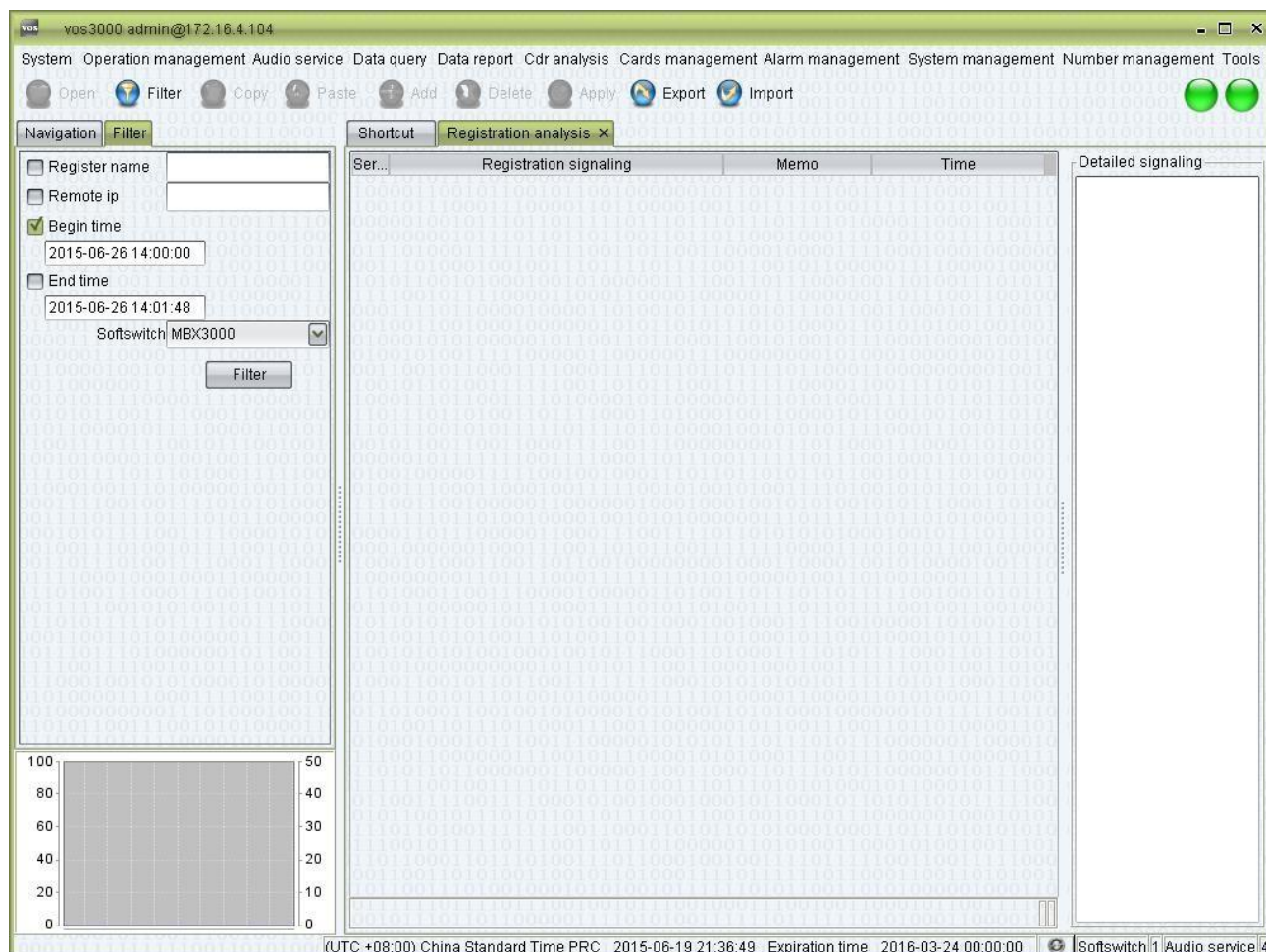
Export: save the signaling as file.

Import: import the signaling file to do analysis.

Downloaded from VOS3000.Com

2.5.3.3 Registration Analysis

This function is used to analysis registration problem.



How to Start

Double-click “Navigation > Operation management > Business analysis > Registration analysis”

Table Items

Serial number: the serial number of signaling interaction.

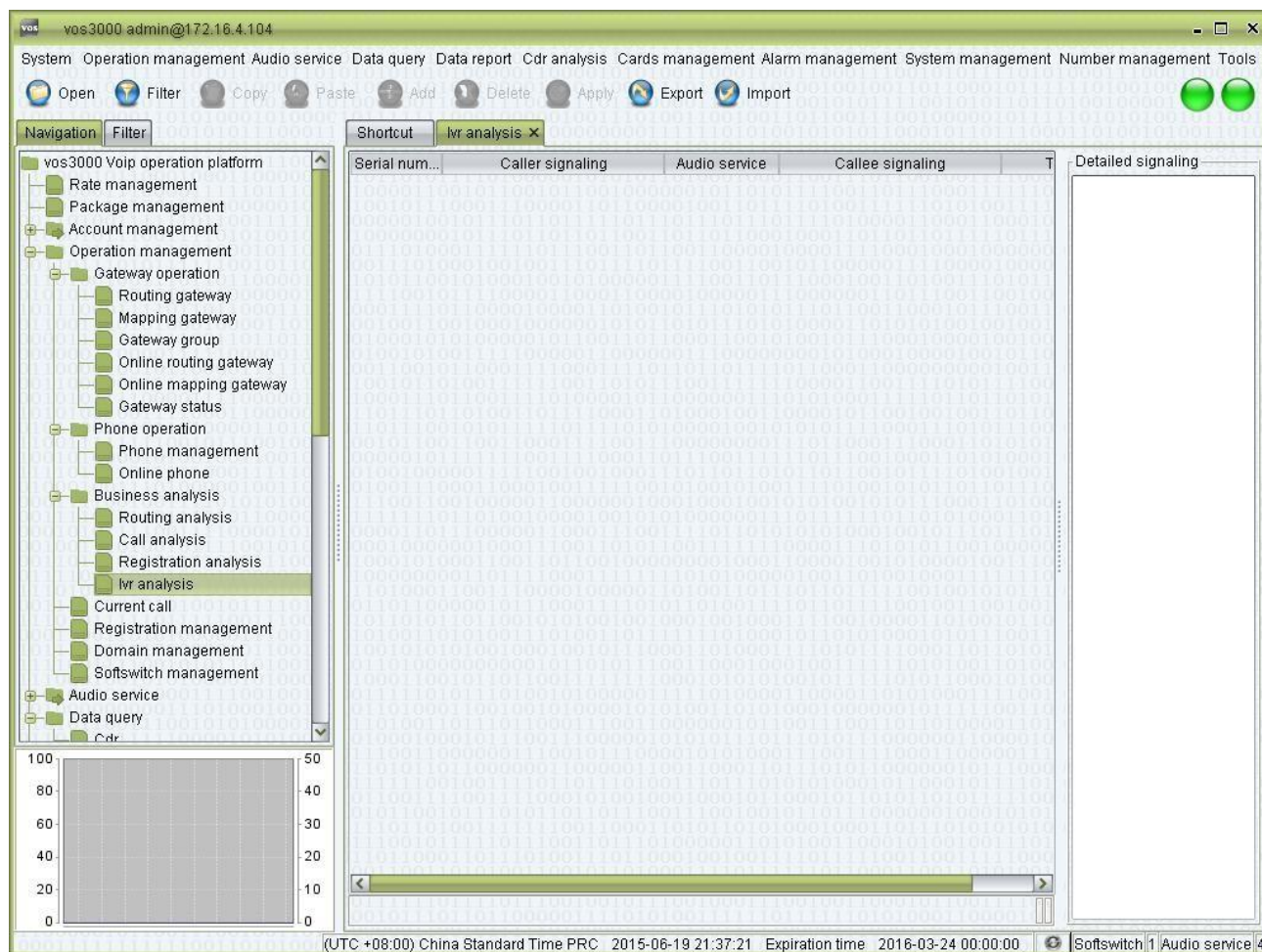
Registration signaling: content of signaling interaction.

Memo: message of softswitch.

Time: time of signaling.

2.5.3.4 Ivr Analysis

This function is used to analysis ivr audio service.



How to Start

Double-click “Navigation > Operation management > Business analysis > Ivr analysis”

Input Items

Serial number: the serial number of signaling interaction.

Caller signaling: content of signaling interaction.

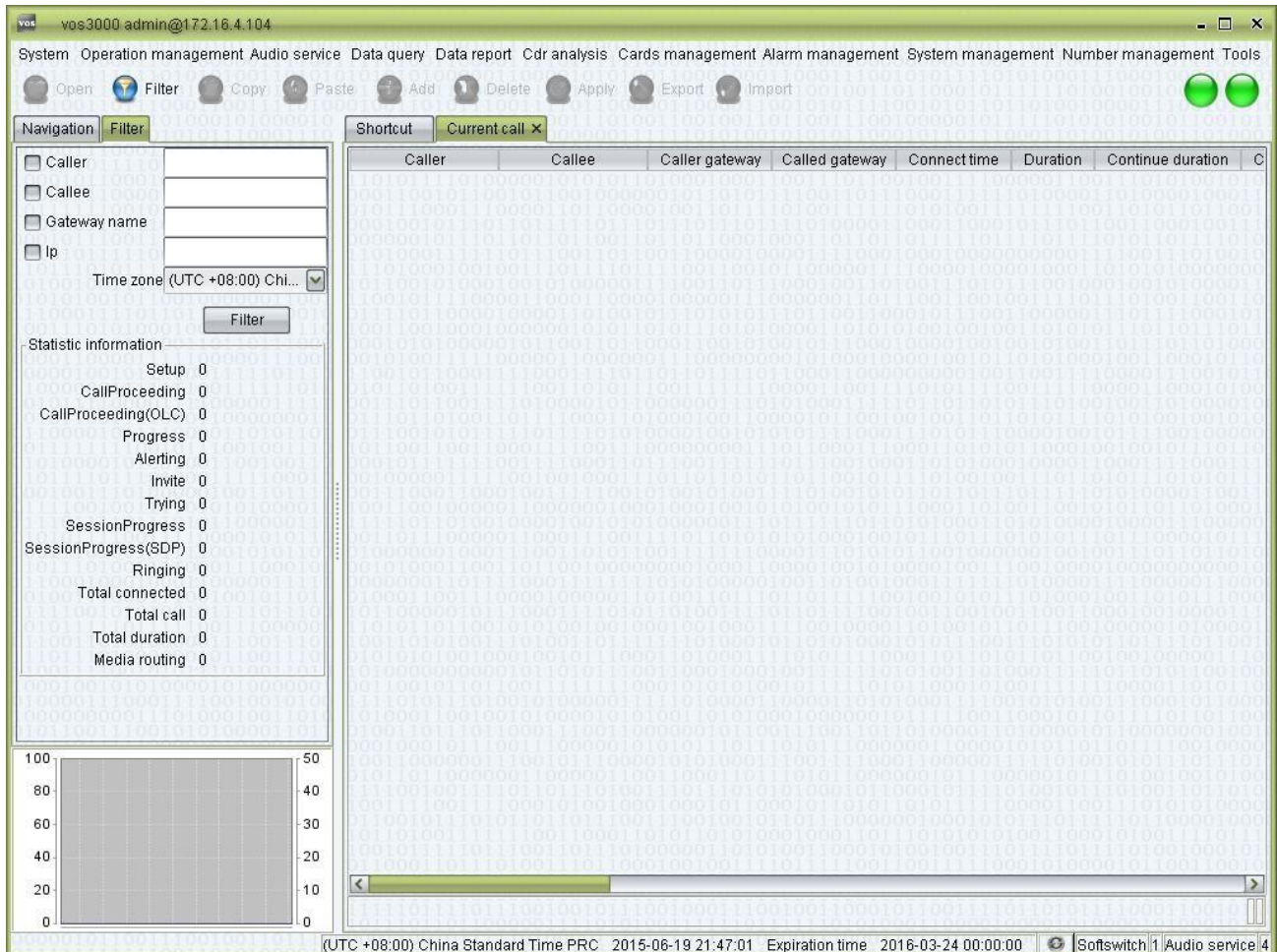
Audio service

Callee signaling

Time: time of signaling.

2.5.4 Current Call

This function is used to query current call.



How to Start

Double-click “Navigation > Operation management > Current call”

Table Items

Caller: the number of the caller.

Callee: the number of the called.

Calling gateway: the gateway between the caller and the softswitch.

Called gateway: the gateway between the called and the softswitch.

Connect time: the time elapsed since the establishment of the connection.

Duration

Continue duration: time elapsed from platform received to call connected.

Connect delay: time elapsed from send call to routing response.

Calling code: the voice encoding used in the session.

Caller rx audio: the voice traffic and package loss of caller.

Caller rx video: the video traffic and package loss of caller

Callee rx audio: the voice traffic and package loss of callee

Callee rx video: the video traffic and package loss of callee

Caller coding: voice encodings supported by the caller (acquired by analyzing the signals).

Callee coding: voice encodings supported by the called (acquired by analyzing the signals).

Caller information: the IP and RTP IP of the caller.

Callee information: the IP and RTP IP of the callee.

Caller dtmf: the DTMF mode of the caller.

Callee dtmf: the DTMF mode of the called.

Media routing: whether the RTP is routed by servers.

Calling device name: the manufacturer of the caller device.

Called device name: the manufacturer of the callee device.

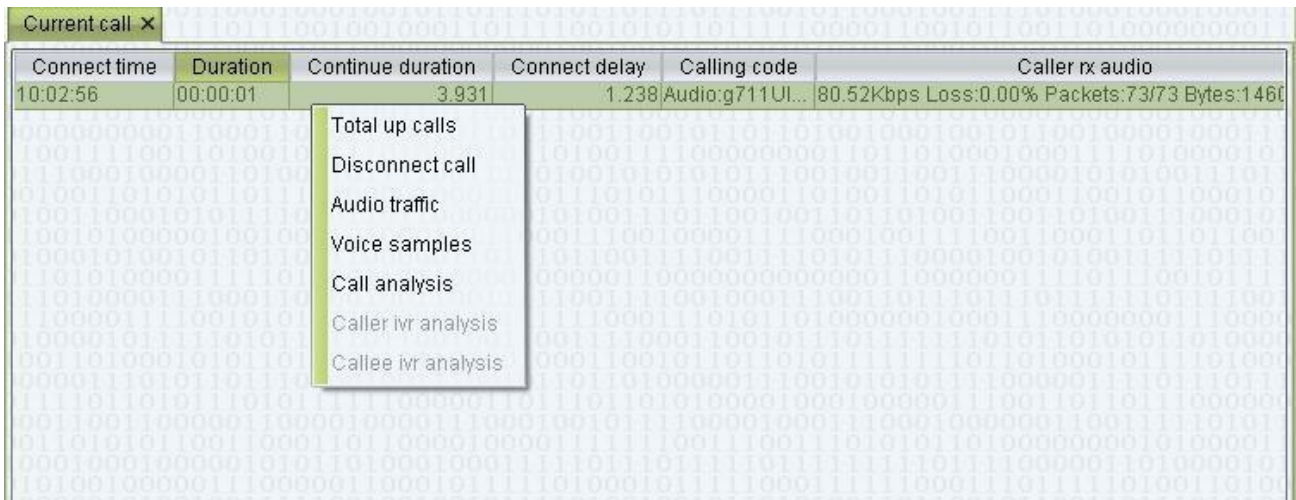
Caller encryption type: the encryption used by the caller.

Callee encryption type: the encryption used by the called.

Softswitch ip: the IP address of the softswitch.

Softswitch name: the name of the softswitch.

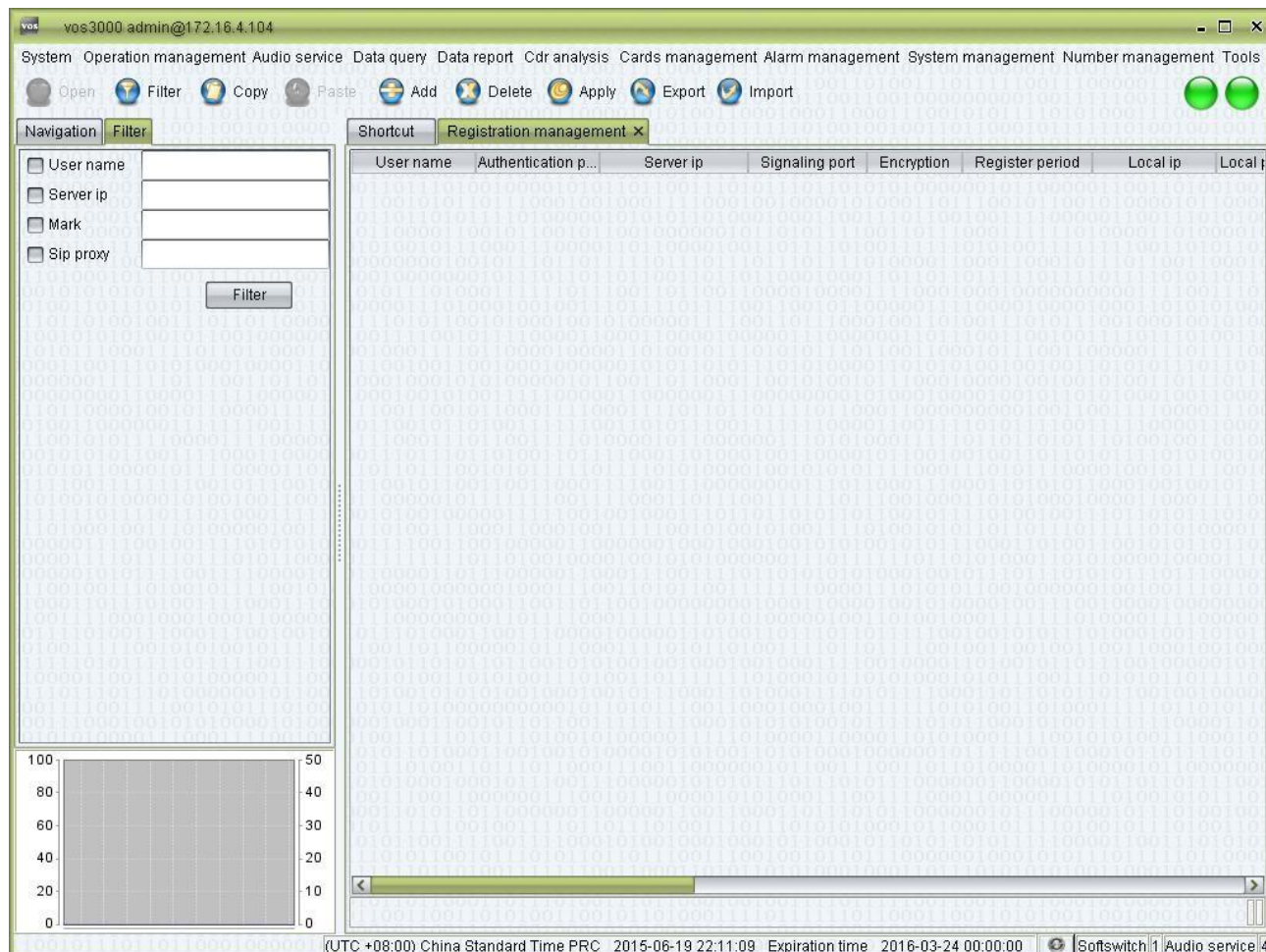
Right-Click Menu



Total up calls
Disconnect call
Audio traffic
Voice samples
Call analysis
Caller ivr analysis
Callee ivr analysis

2.5.5 Registration Management

This function is used to manage registration to other platform.



How to Start

Double-click “Navigation > Operation management > Registration management”

Table Items

User name

Authentication password

Server ip

Signaling port

Encryption

Register period

- Default: set by "Operation management > Softswitch management > Additional settings > System parameter > SS_SIPREGISTEREXPIRE".
- Auto negotiation: decide by server.

Local ip

Local port

Softswitch

Mark: identifier of registration information.



NOTE

When routing gateway type is registration, use mark as registration identification.

Registration time

Update time

Actual register cycle

Last error: error of last time.

Authentication user: username field of SIP 401/407 message, use user name as default.

Host name: server address field of SIP REGISTER/FROM/TO, use server ip as default.

SIP proxy: address of SIP Route.

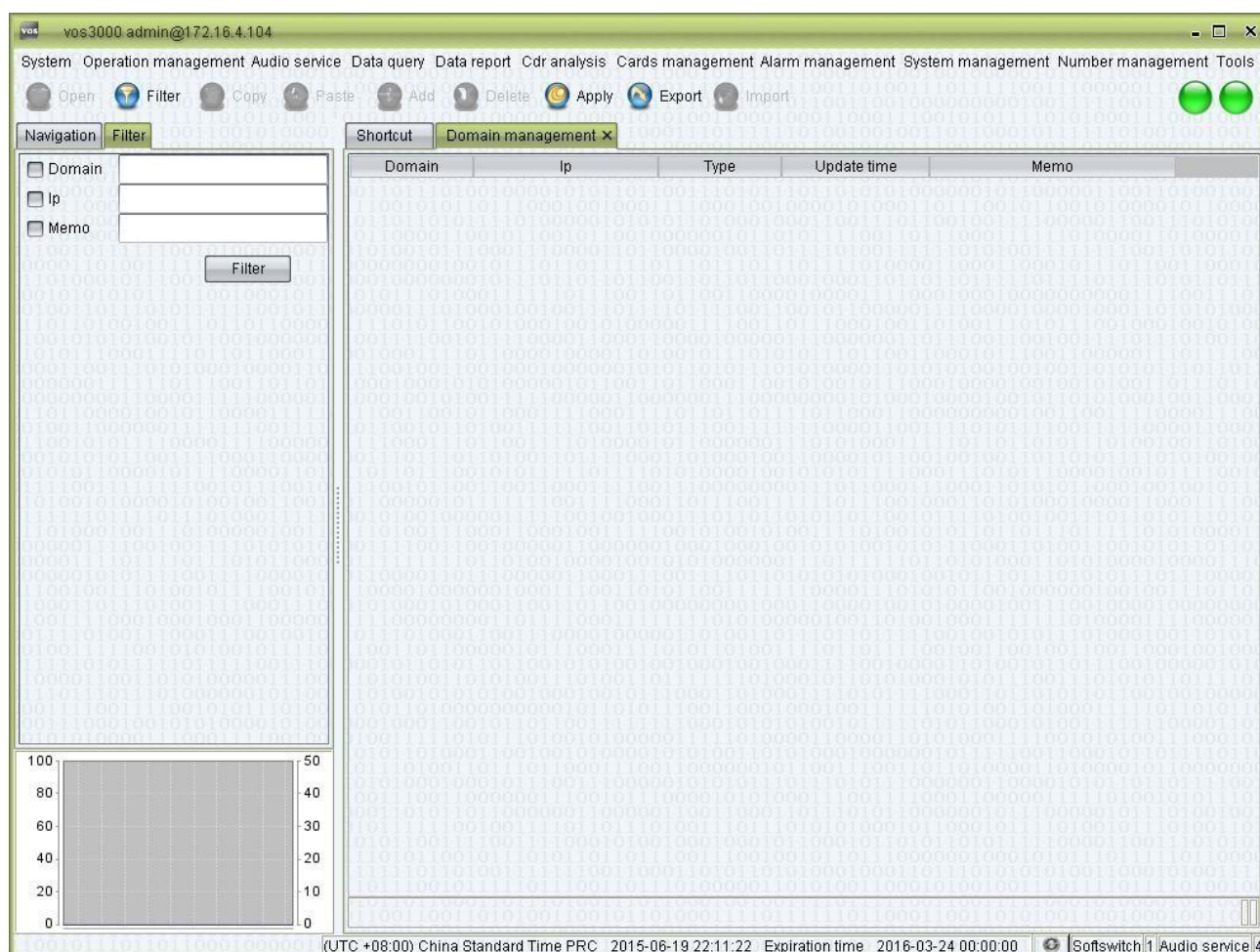
User-Agent: field of SIP User-Agent, use VOS3000 + version as default.

Type

- Third-party: created by third party.
- BLANK: created by system.

2.5.6 Domain Management

This function is used to manage domain, which is used for Routing Gateway and Registration.



How to Start

Double-click “Navigation > Operation management > Domain management”

Table Items

Domain: the domain name of routing gateway or registration.

Ip: Domain's IP.

Type

- Dynamic: update DNS every several minutes, can be set by “System management > System parameter > SERVER_DNS_UPDATE_INTERVAL”.
- Static: won't check DNS.

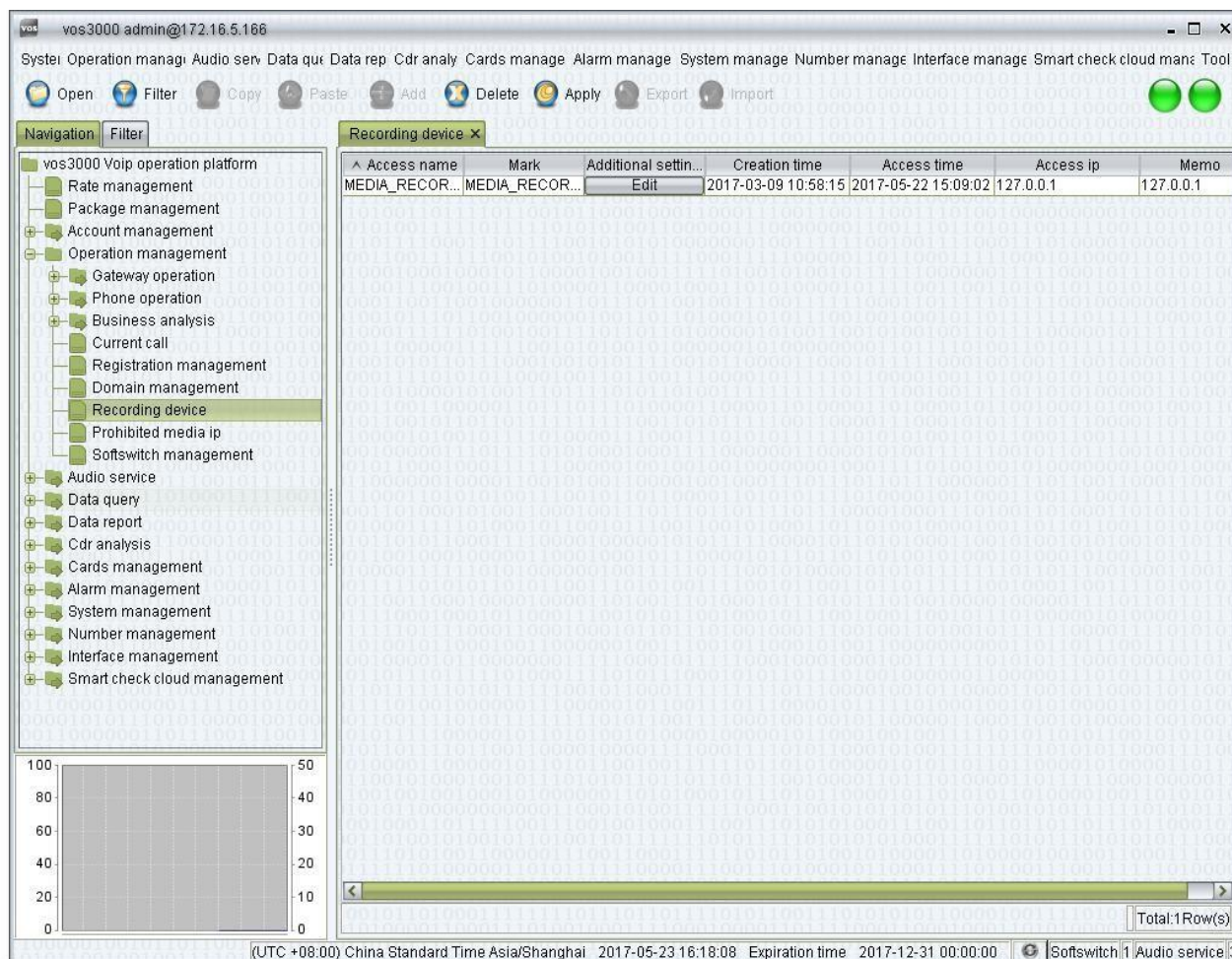
Update time: DNS last update time.

Memo

Downloaded from VOS3000.Com

2.5.7 Recording Device

This function is used to manage recording device



How to Start

Double-click "Navigation > Operation management > Recording device"

Table Items

Access name: the name of the recording device

Mark

Additional settings

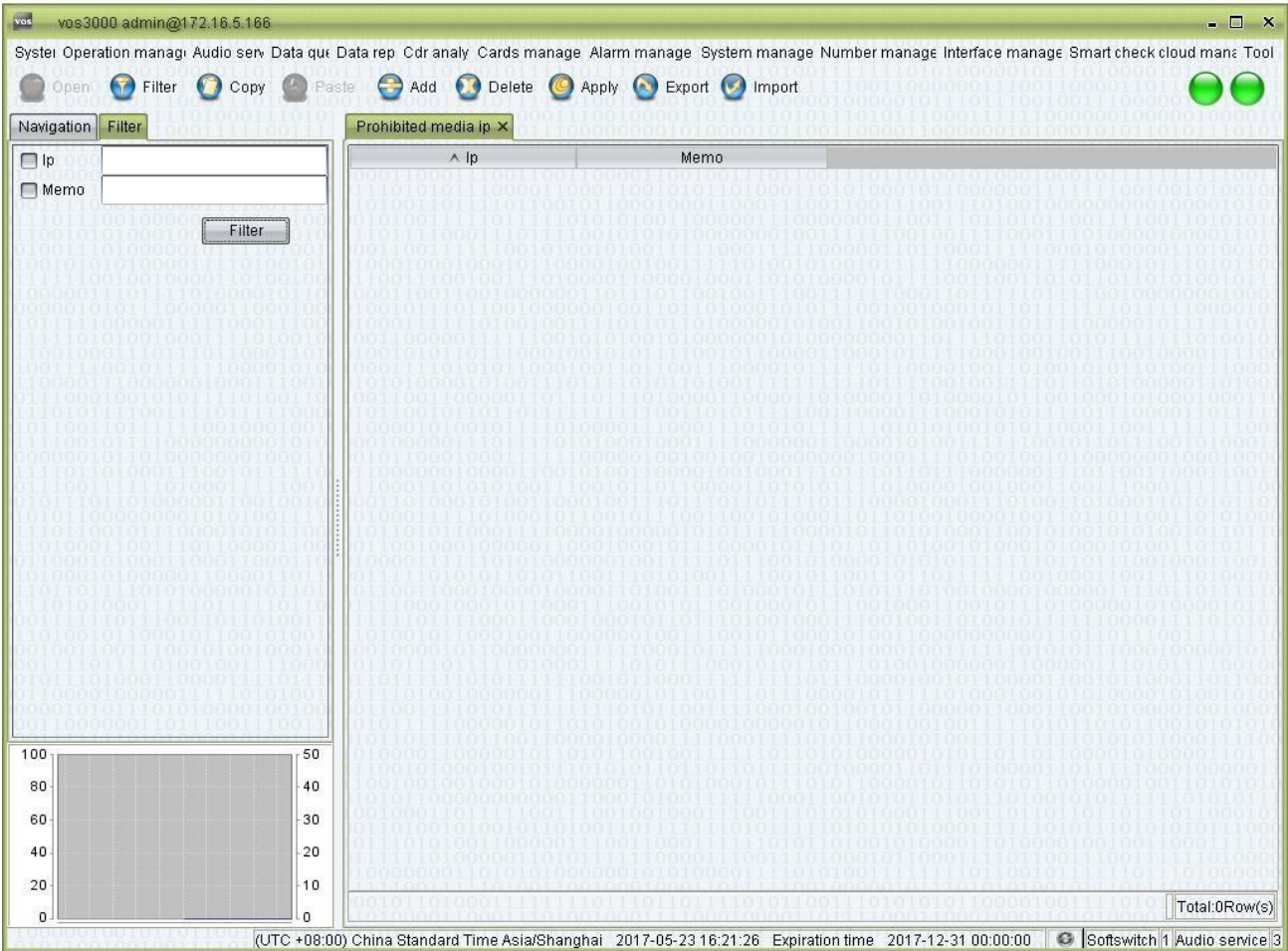
Creation time: the time of first access to the softswitch.

Access time: the most recent access to the softswitch.

Access ip: the IP address of the recording device

2.5.8 Prohibited Media ip

This function is used to prohibit media ip.



How to Start

Double-click "Navigation > Operation management > prohibited media ip"

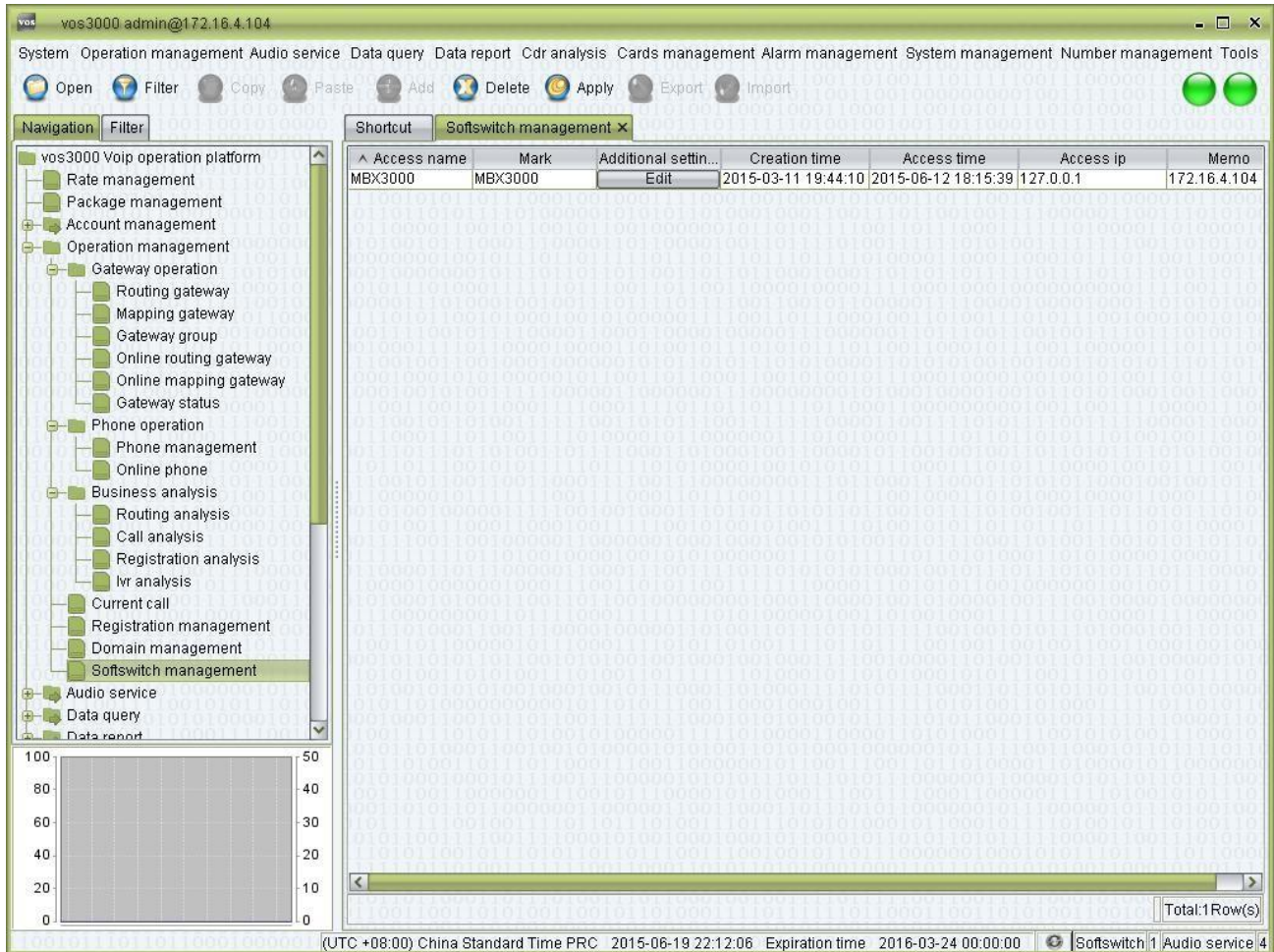
Table Items

Ip: prohibited media ip

Memo

Softswitch Management

This function is used to manage softswitch.



How to Start

Double-click "Navigation > Operation management > Softswitch management"

Table Items

Access name: the name of the Softswitch.

Mark

Additional settings

Creation time: the time of first access to the softswitch.

Access time: the most recent access to the softswitch.

Access ip: the IP address of the softswitch.

Memo: comments on the softswitch.

Right-Click Menu

Synchronize data: synchronize settings of the softswitch with VOS3000.

Current call: current sessions on the softswitch.

Registration track

System information: information about the softswitch.

The screenshot shows the VOS3000 admin interface. The top menu bar includes: System, Operation management, Audio service, Data query, Data report, Cdr analysis, Cards management, Alarm management, System management, Number management, Tools. Below the menu bar is a toolbar with icons for Open, Filter, Copy, Paste, Add, Delete, Apply, Export, and Import. The left sidebar contains a navigation tree with categories like Rate management, Package management, Account management, Operation management, Gateway operation, Phone operation, Business analysis, Routing analysis, Call analysis, Registration analysis, Ivr analysis, Current call, Registration management, Domain management, Softswitch management, Audio service, Data query, and Data report. The main window displays the 'Softswitch-MBX3000>system information' data. The data is organized into sections: License information, Product information, Operating parameters, and Synchronize state. The table below lists the parameters and their values.

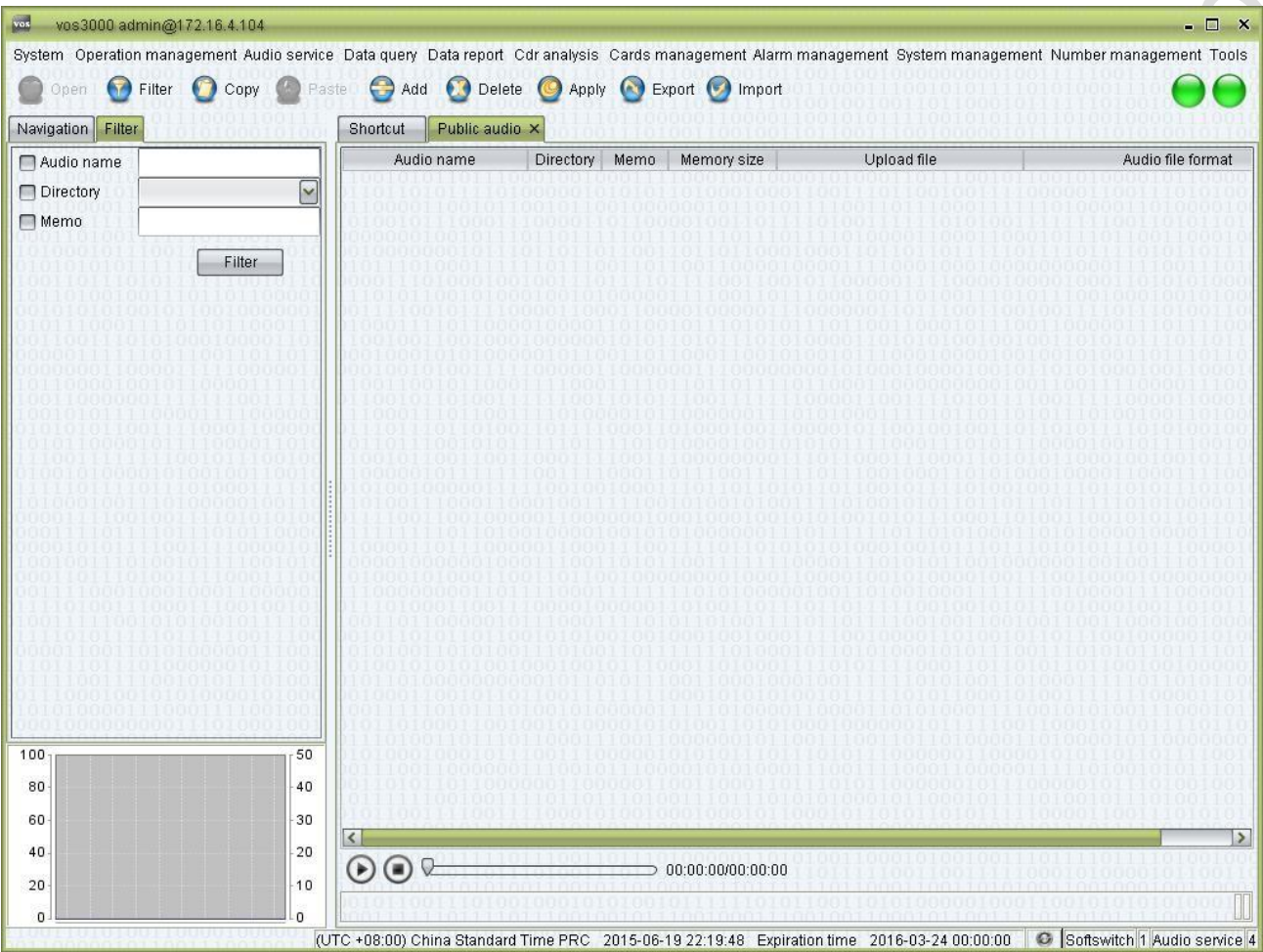
Information name	Information value
License information	
Product information	H323/SIP
Terminal limit	No limit
Concurrent call limit	5000
Expiry date	2016-03-24 00:00:00
Additional information	Routing settings
Operating parameters	
COMPILE_TIME	Apr 7 2015 18:41:49
BIND_IP	172.16.4.104
H323_RAS_PORT	1719
H323_RC4_RAS_PORT	3719
H323_SIGNAL_PORT	1720
H323_RC4_SIGNAL_PORT	3720
SIP_PORT	5060,6060,5066
SIP_RC4_PORT	5070
SIP_STATIC_RC4_PORT	5078
SIP_STATIC_AES128_PORT	5079
SIP_STATIC_AES256_PORT	5080
SIP_DYNAMIC_XOR_RC4_PORT	5071
SIP_DYNAMIC_XOR_AES128_PORT	5072
SIP_DYNAMIC_XOR_AES256_PORT	5073
SIP_DYNAMIC_RC4_PORT	5074
SIP_DYNAMIC_AES128_PORT	5076
SIP_DYNAMIC_AES256_PORT	5077
SS_TRACE_CALL	ON
SS_TRACE_CALL_START	2015-06-09 18:22:45
SS_TRACE_CALL_MAX_SIZE	16MB
SS_TRACE_REGISTER	OFF
Synchronize state	

At the bottom of the window, there is a status bar showing: (UTC +08:00) China Standard Time PRC 2015-06-19 22:12:56 Expiration time 2016-03-24 00:00:00 | Softswitch 1 Audio service 4

2.6 Audio Service

2.6.1 Public Audio

This function is used to manage common audio file.



How to Start

Double-click “Navigation > Audio service > Public audio”

Table Items

Audio name



NOTE

Special characters like “*?<>|!#%^&” and “..” are not allowed for the name.

Directory: the language catalogue of the audio.

Memo: comments on the voice file.

Memory size: display of the size of the voice file. ("0" indicates the file has not been uploaded yet.)

Upload file: specify the local path of the voice file for upload.

Audio file format: display the format of the uploaded voice file.

Toolbar

Filter: filter the items.

Copy: copy the selected audio information.



NOTE

The actual audio file data will not be copied.

Add: add contents to the service.

Delete: delete selected contents.

Import: import voice files in batch mode. The voices will be automatically named according to their file names.

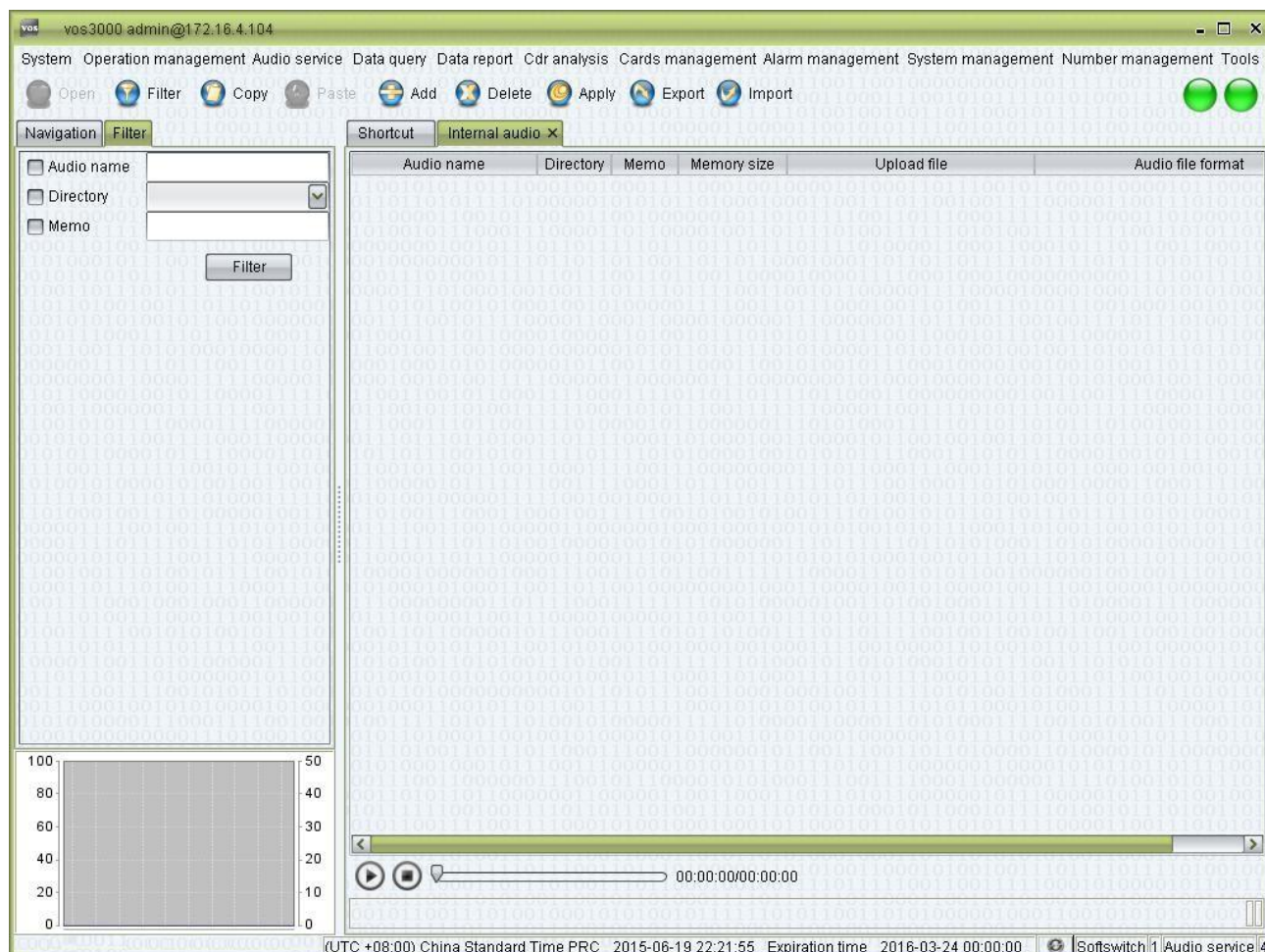


CAUTION

Supported formats include WAV files (8 KHz, 16 bit, monophonic).

2.6.2 Internal Audio

This function is used to manage error prompt audio file.



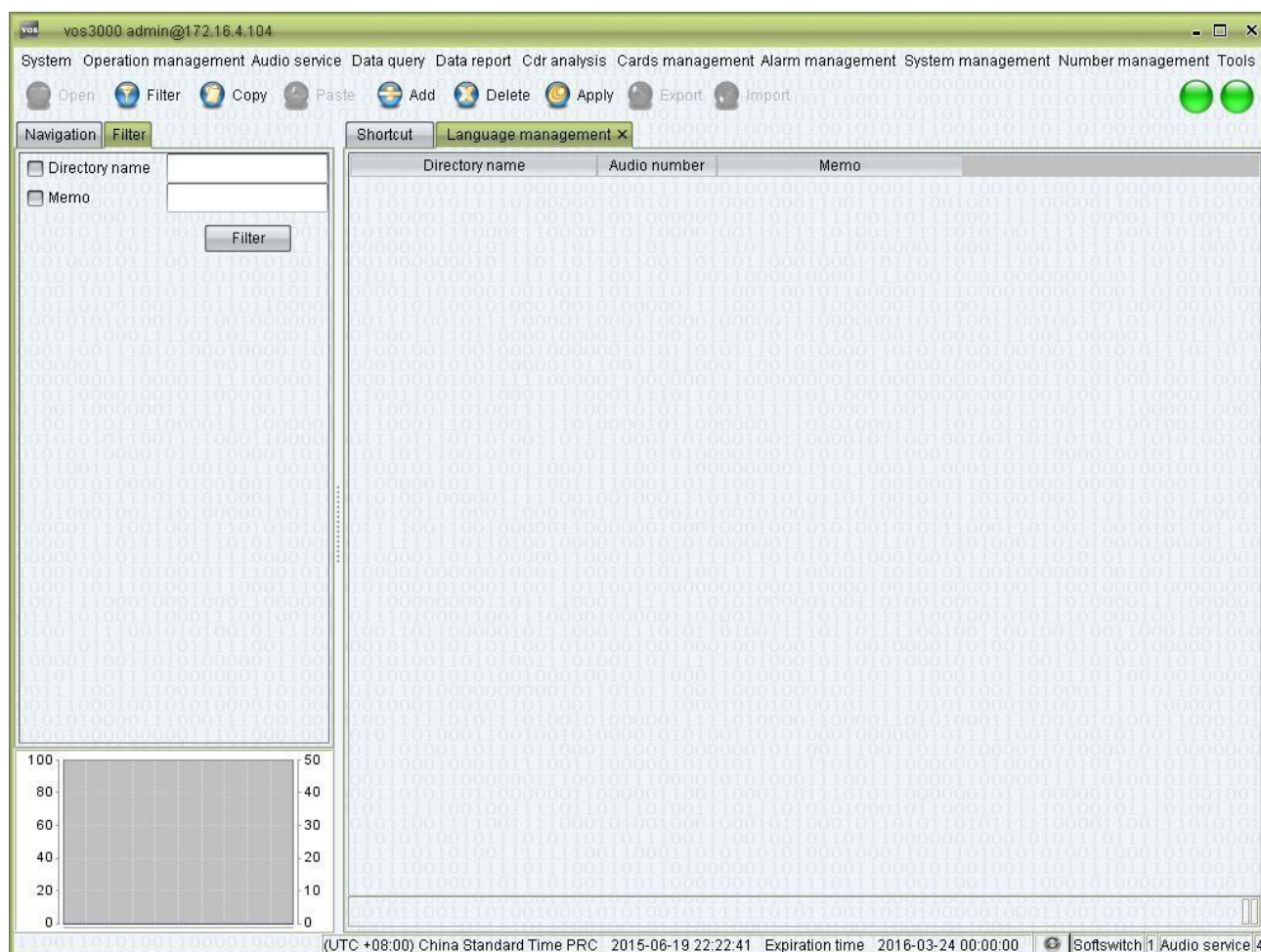
How to Start

Double-click “Navigation > Audio service > Internal audio”

Table Items

See the descriptions in “Public audio”.

2.6.3 Language Management



How to Start

Double-click "Navigation > Audio service > Language management"

Table Items

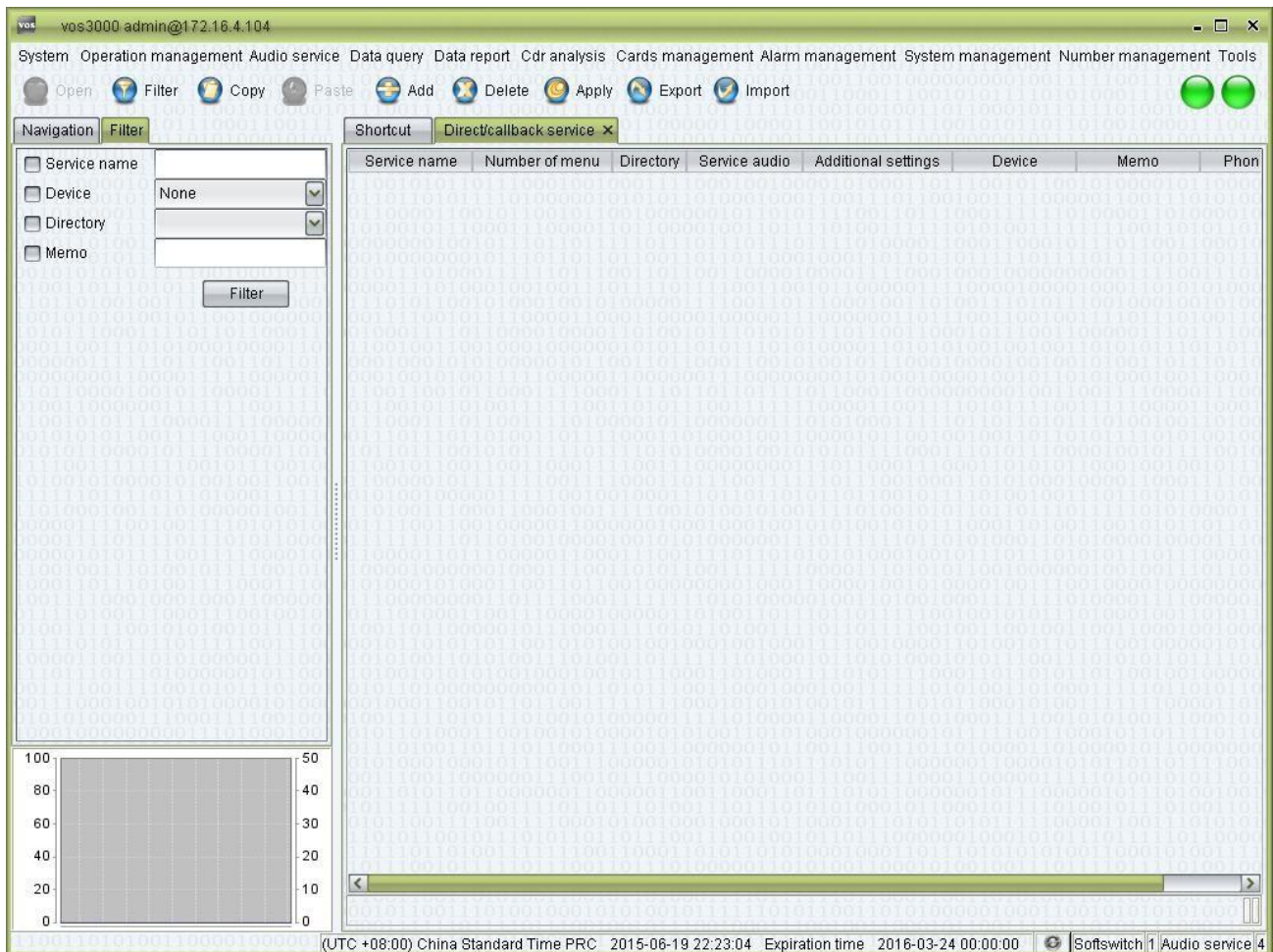
Directory name: the name of the language catalogue.

Memo: information related to the language catalogue.

2.6.4 Direct/Callback

2.6.4.1 Direct/Callback Service

This function is used to define direct/callback service and billing.



How to Start

Double-click “Navigation > Audio service > Direct/Callback > Direct/Callback service”

Table Items

Service name: name of audio service, used in “Phone management > Advanced configuration > Audio service”

Number of menu: number of IVR menu.

Directory: service’s language.

Service audio: service’s private audio.

Additional settings: see below.

Device: service's devices.

Memo

Phone number: the number of phone which uses this audio service.

Caller number group of record: record the call if caller is in the number group when second line is connected.

Callee number group of record: record the call if callee is in the number group when second line is connected.

Audio service<webcallback>additional settings

Normal

Call out display number: Caller

Call quick display number: Caller

Callout callee rewrite rule:

Ringing control

First time audio delay: 2

Access alerting time: None

Alerting end action: Hangup

Callback

Callback delay: 1

☐ Replace last call

☒ Call both side at the same time

Access decline signaling: Busy Here(486)

Retry times: 1

Retry interval: 10

Callback rewrite rule:

Ok Cancel

Call out display number: "Ivr number" or "Caller".

Call quick display number: "Ivr number" or "Caller" number is sent the called for quick calls.

Call out callee rewrite rule: rewrite rules of the called number for calls initiated by the IVR.

Ringing control

First time audio delay: set the delay after which the voice will be played upon access to the IVR (Unit: second).

Access alerting time: maximum time for call-in ringing (Unit: second).

Alerting end action: "Connect" or "Hang up".

Callback

Callback delay: time waited before calling back (Unit: second).

Replace last call

Call both side at the same time

Access decline signaling: specify the rejection signal for IVR callback.

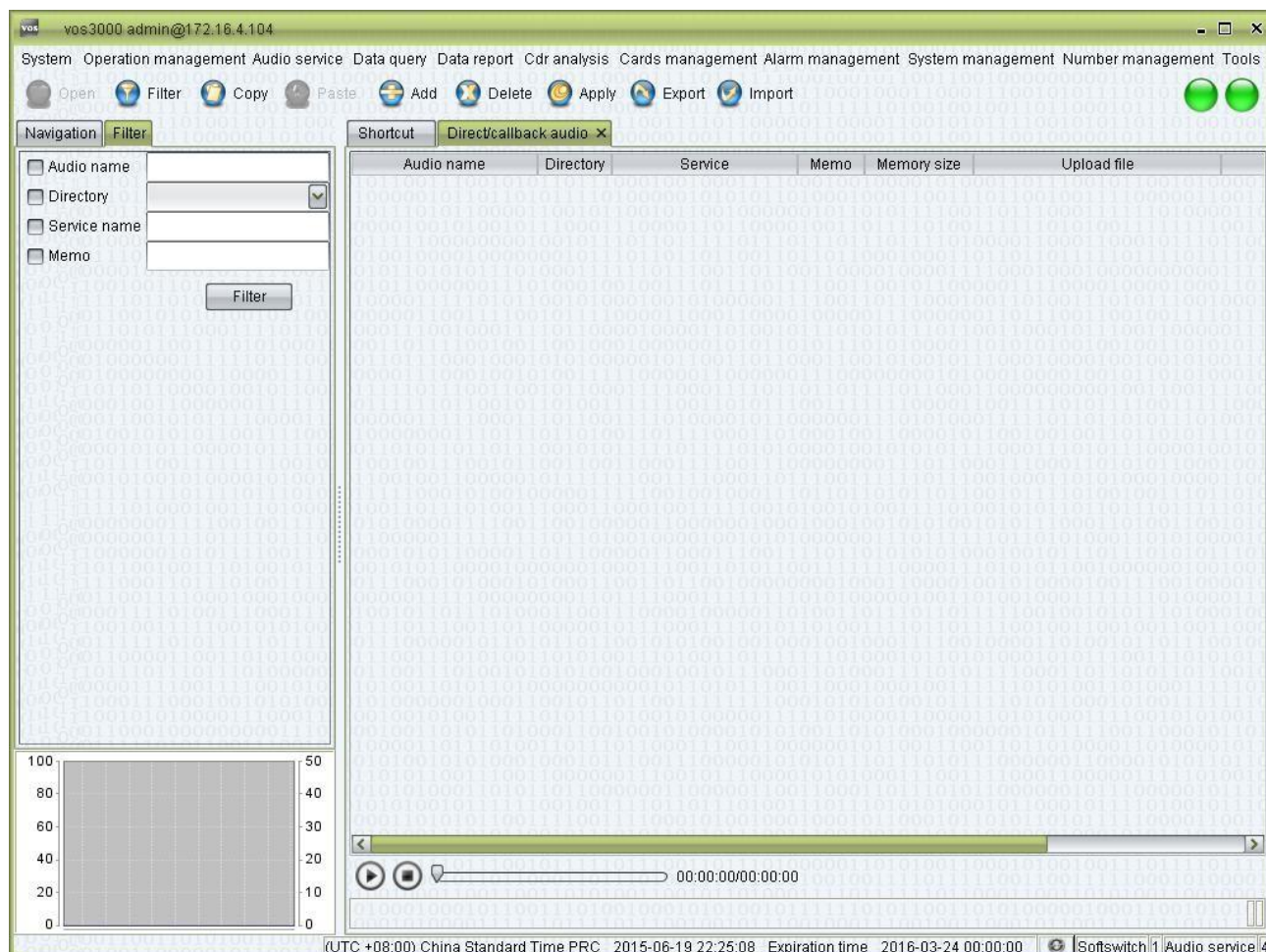
Retry times: the number of retries for unsuccessful callback.

Retry interval: the time interval between two retries (Unit: second).

Callback rewrite rule: the rewrite rules applied to the called number during callback.

2.6.4.2 Direct/Callback Audio

This function is used to manage direct/callback's audio.



How to Start

Double-click "Navigation > Audio service > Direct/Callback > Direct/Callback audio"

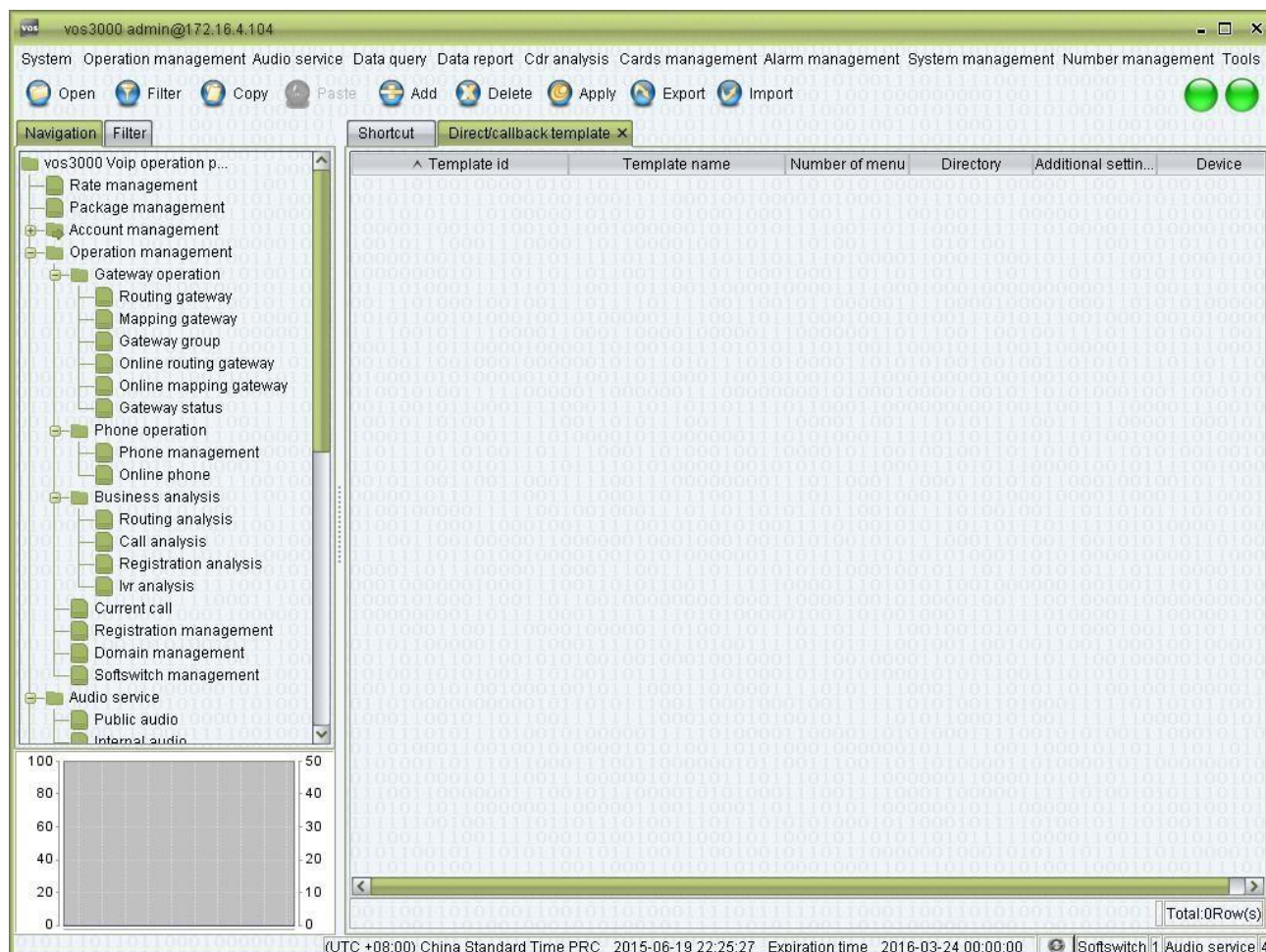
Table Items

See "Public audio".

Service: audio belong to which service, blank means all services can use this audio.

2.6.4.3 Direct/Callback Template

This function is used to define direct/callback's template.



How to Start

Double-click "Navigation > Audio service > Direct/Callback > Direct/Callback template"

Table Items

Template id
Template name
Number of menu
Directory
Additional settings
Device

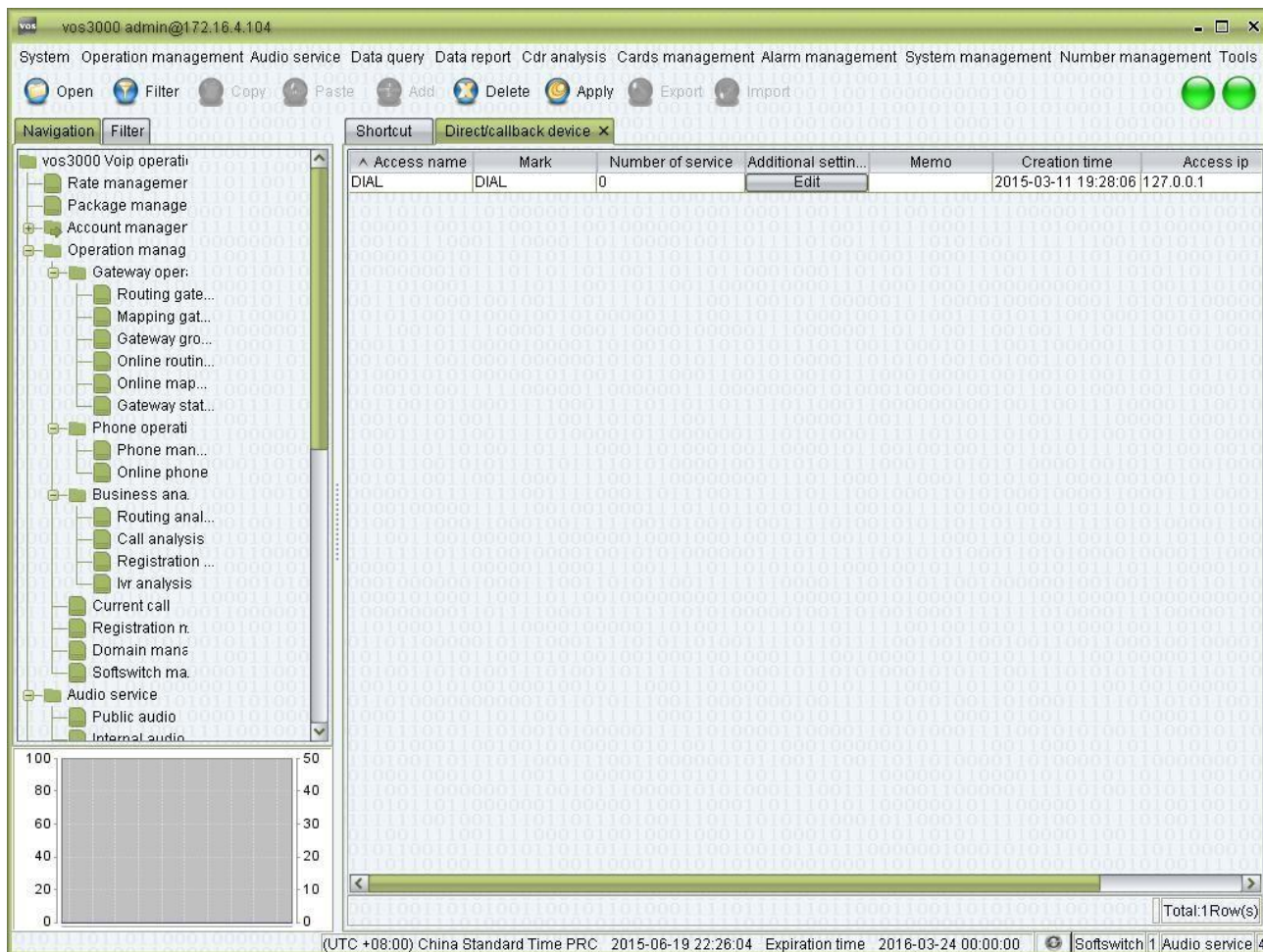
Others

Template is used to configure service, right click to select template and apply to service.

Downloaded from VOS3000.Com

2.6.4.4 Direct/Callback Device

This function is used to manage direct/callback device.



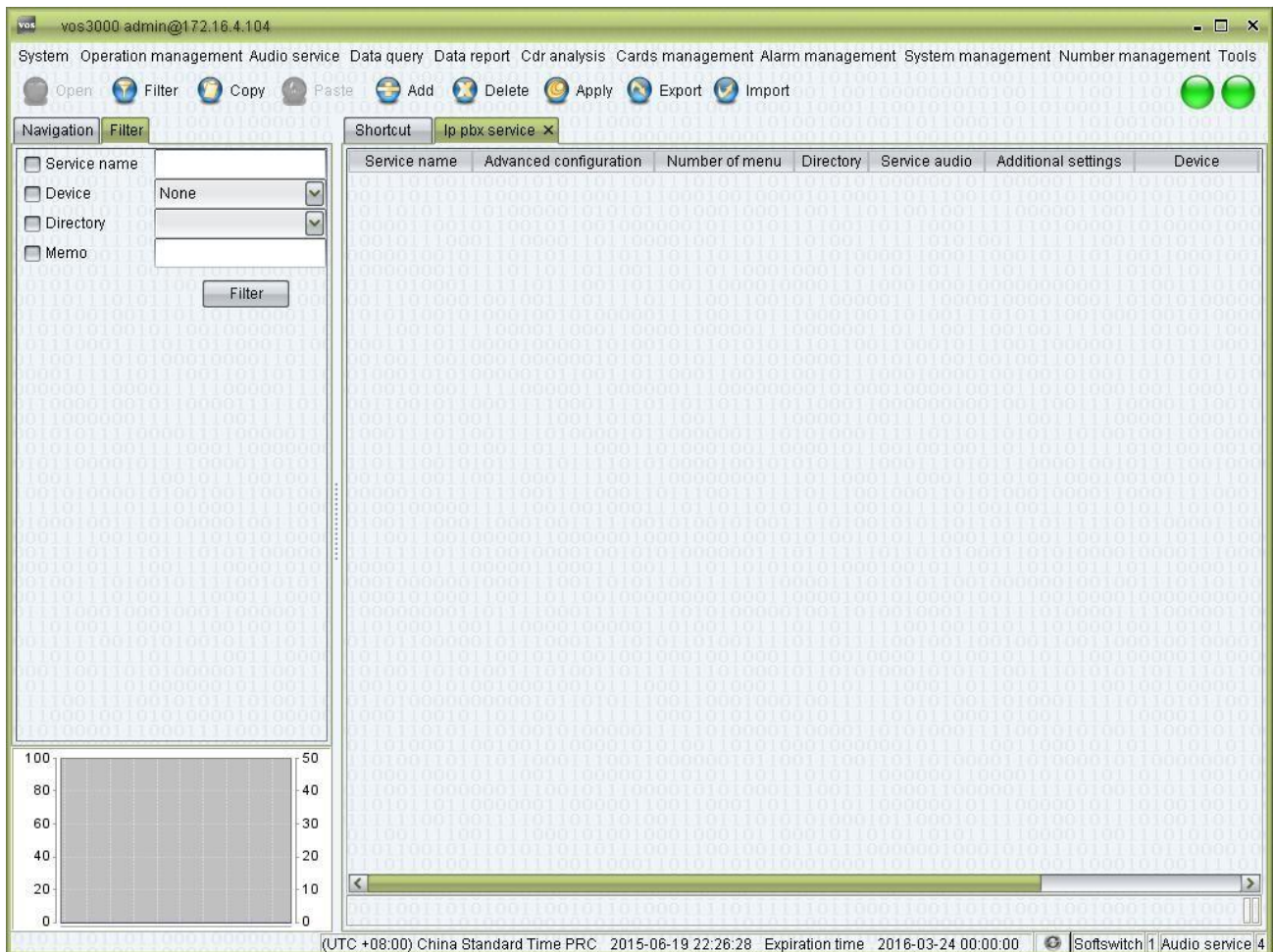
How to Start

Double-click "Navigation > Audio service > Direct/Callback > Direct/Callback device"

2.6.5 IP PBX

2.6.5.1 IP PBX Service

This function is used to define IP PBX service and billing.



How to Start

Double-click "Navigation > Audio service > Ip pbx > Ip pbx service"

Table Items

See "Direct/Callback service".

Advanced configuration

Advanced configuration

Audio service

Weekdays	Begin time	End time	Audio file	Audio s
1,2,3,4,5	00:00:00	09:00:00	\$non-working-h...	Chod
6,0	00:00:00	24:00:00	\$non-working-h...	Chod
1,2,3,4,5	09:00:00	12:00:00	\$morning,\$pbx...	Chod
1,2,3,4,5	12:00:00	17:00:00	\$afternoon,\$pbx...	Chod
1,2,3,4,5	17:00:00	24:00:00	\$non-working-h...	Chod

Total:5Row(s)

Key	Phone number	Memo
0	901	Directory assistance

Total:1Row(s)

Waiting audio

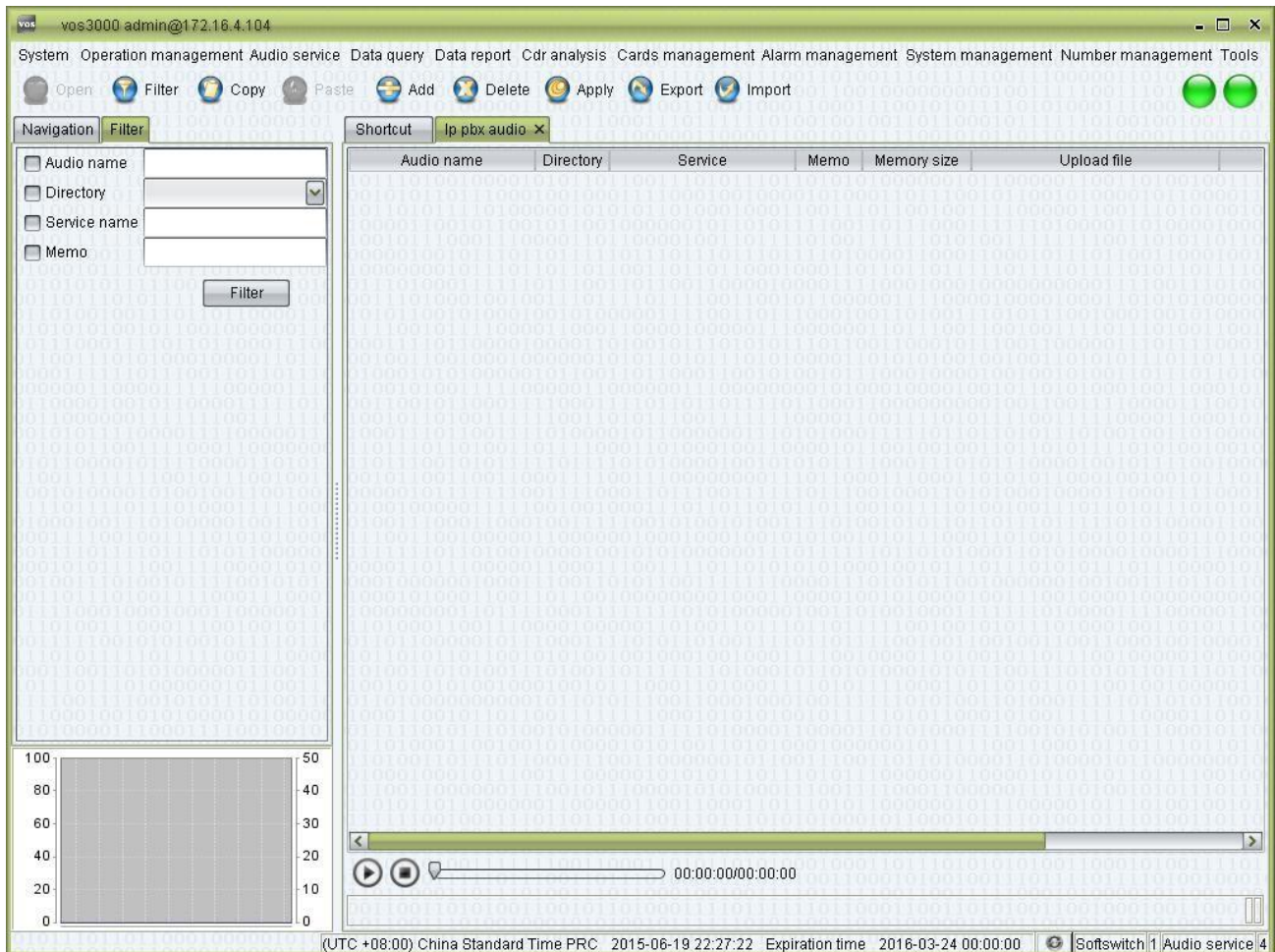
Template Apply

Ok Cancel

Set period audio and operations.

2.6.5.2 IP PBX Audio

This function is used to manage IP PBX's audio.



How to Start

Double-click "Navigation > Audio service > Ip pbx > Ip pbx audio"

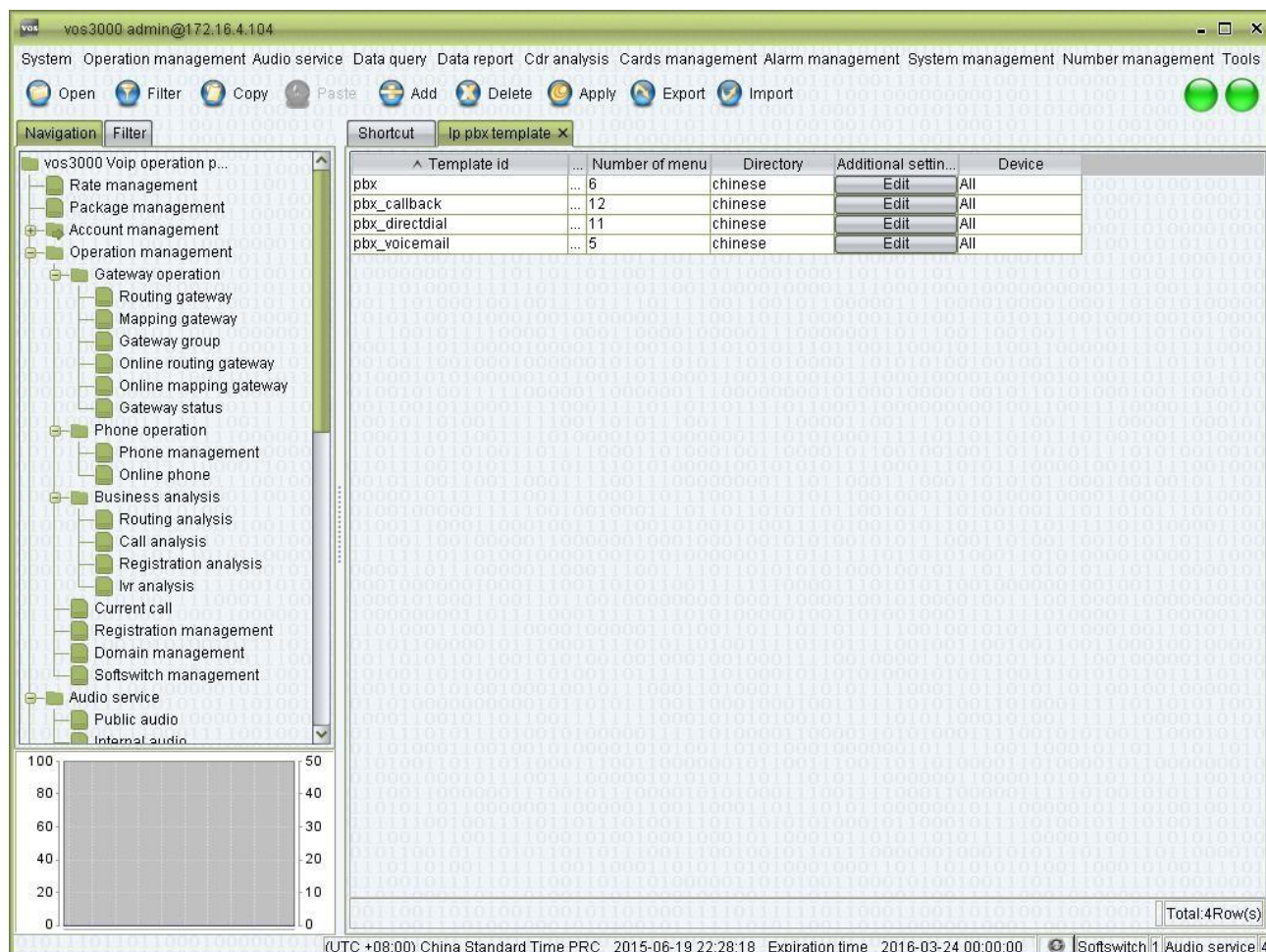
Table Items

See "Public audio".

Service: audio belong to which service, blank means all services can use this audio.

2.6.5.3 IP PBX Template

This function is used to define IP PBX's template.



How to Start

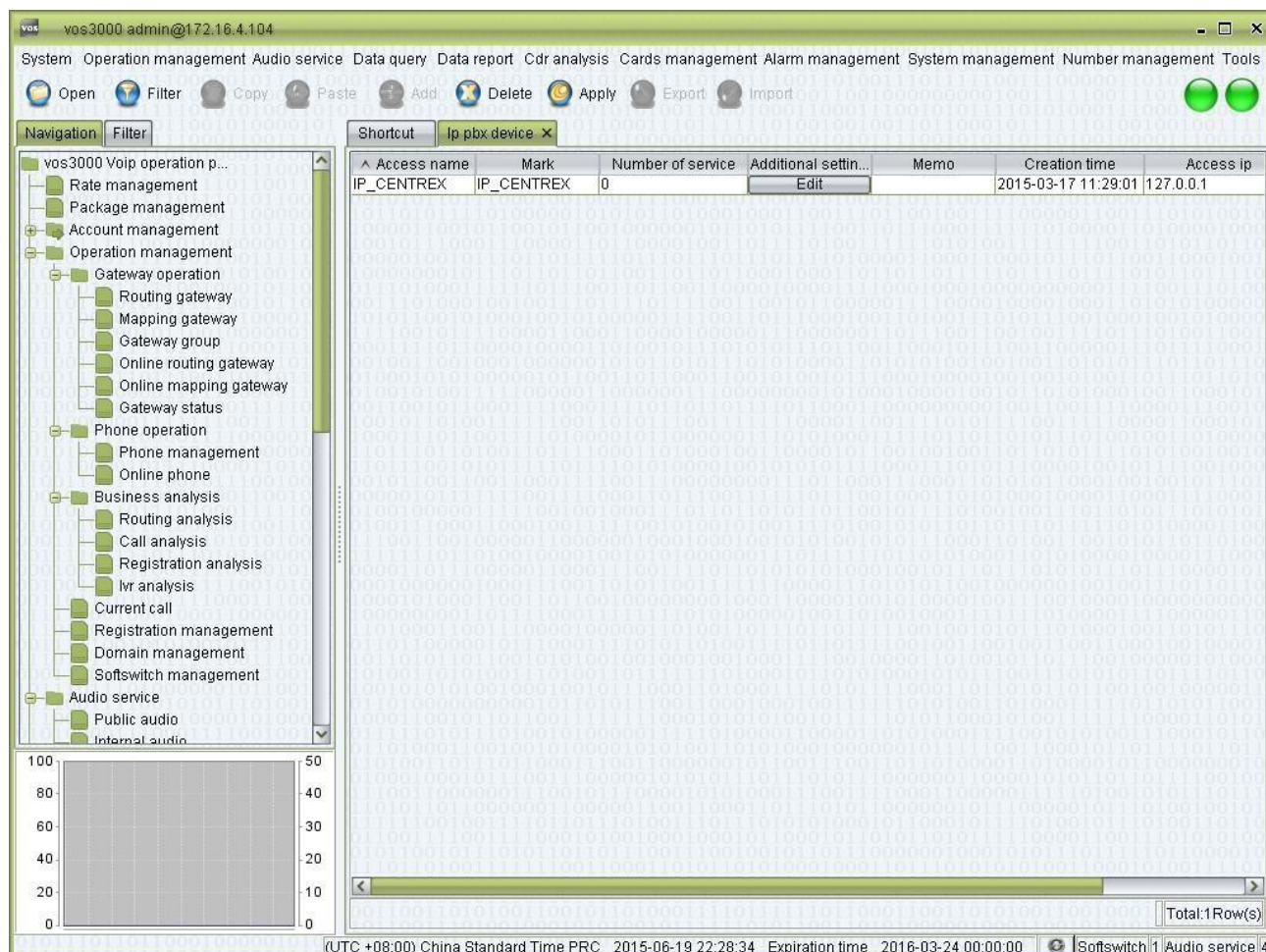
Double-click "Navigation > Audio service > Ip pbx > Ip pbx template"

Table Items

See "Direct/Callback template".

2.6.5.4 IP PBX Device

This function is used to manage IP PBX device.



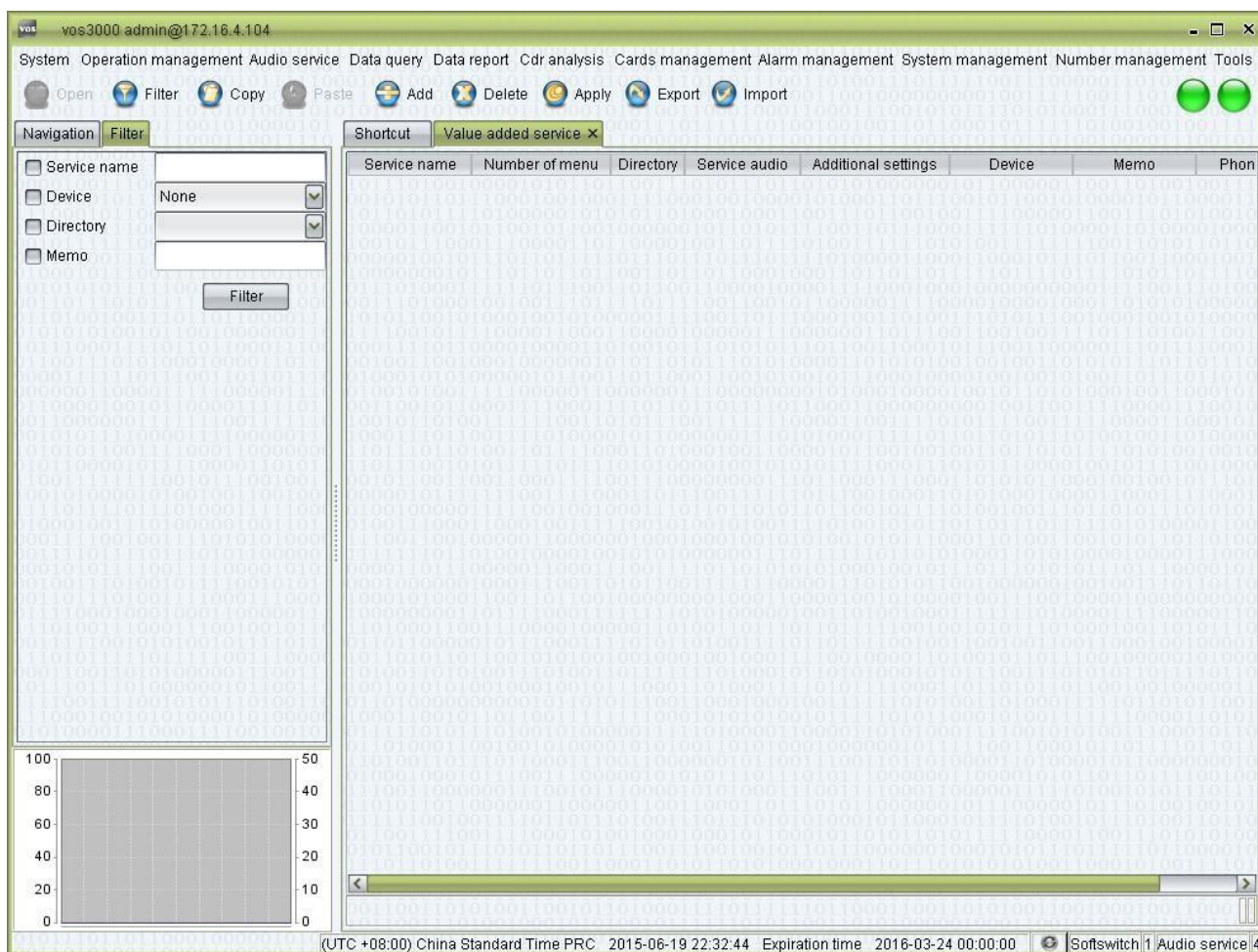
How to Start

Double-click "Navigation > Audio service > Ip pbx > Ip pbx Device"

2.6.6 Value Added

2.6.6.1 Value Added Service

This function is used to define Value Added service and billing.



How to Start

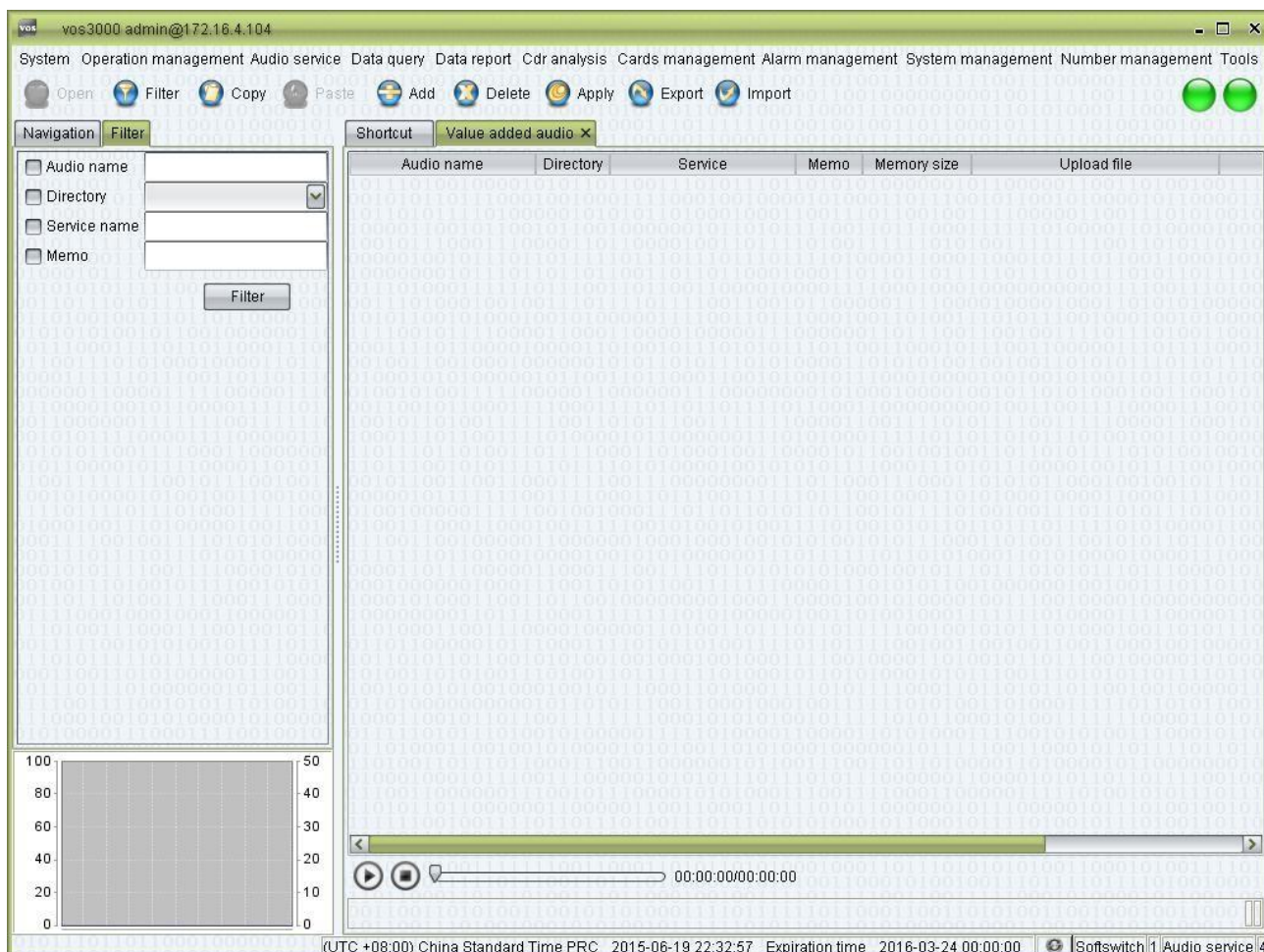
Double-click "Navigation > Audio service > Value added > Value added service"

Table Items

See "Direct/Callback Service".

2.6.6.2 Value Added Audio

This function is used to manage Value Added audio.



How to Start

Double-click “Navigation > Audio service > Value added > Value added audio”

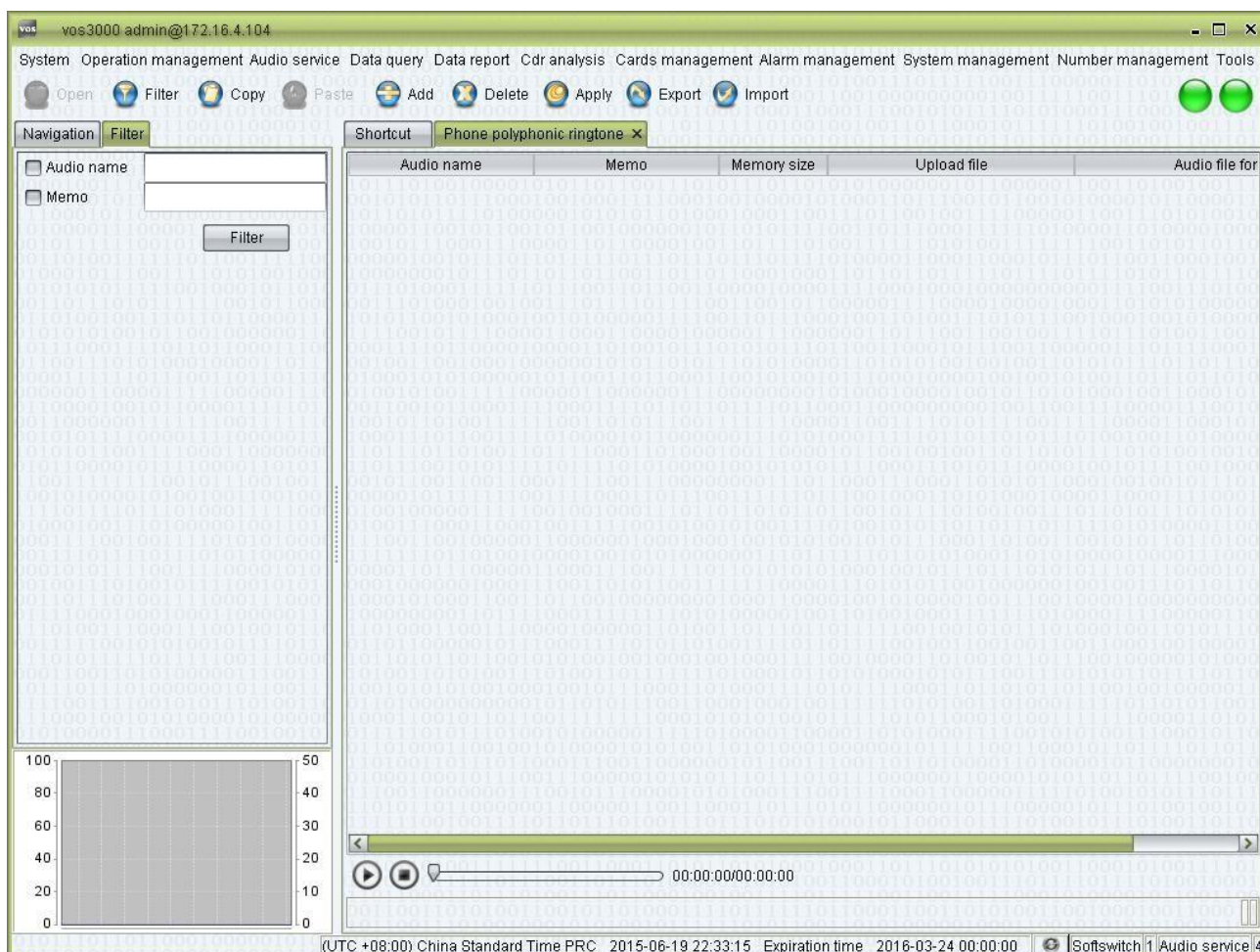
Table Items

See “Public audio”.

Service: audio belong to which service, blank means all services can use this audio.

2.6.6.3 Phone Polyphonic Ringtone

This function is used to manage polyphonic ringtone audio, which is used in “Phone management > Supplementary service”.



How to Start

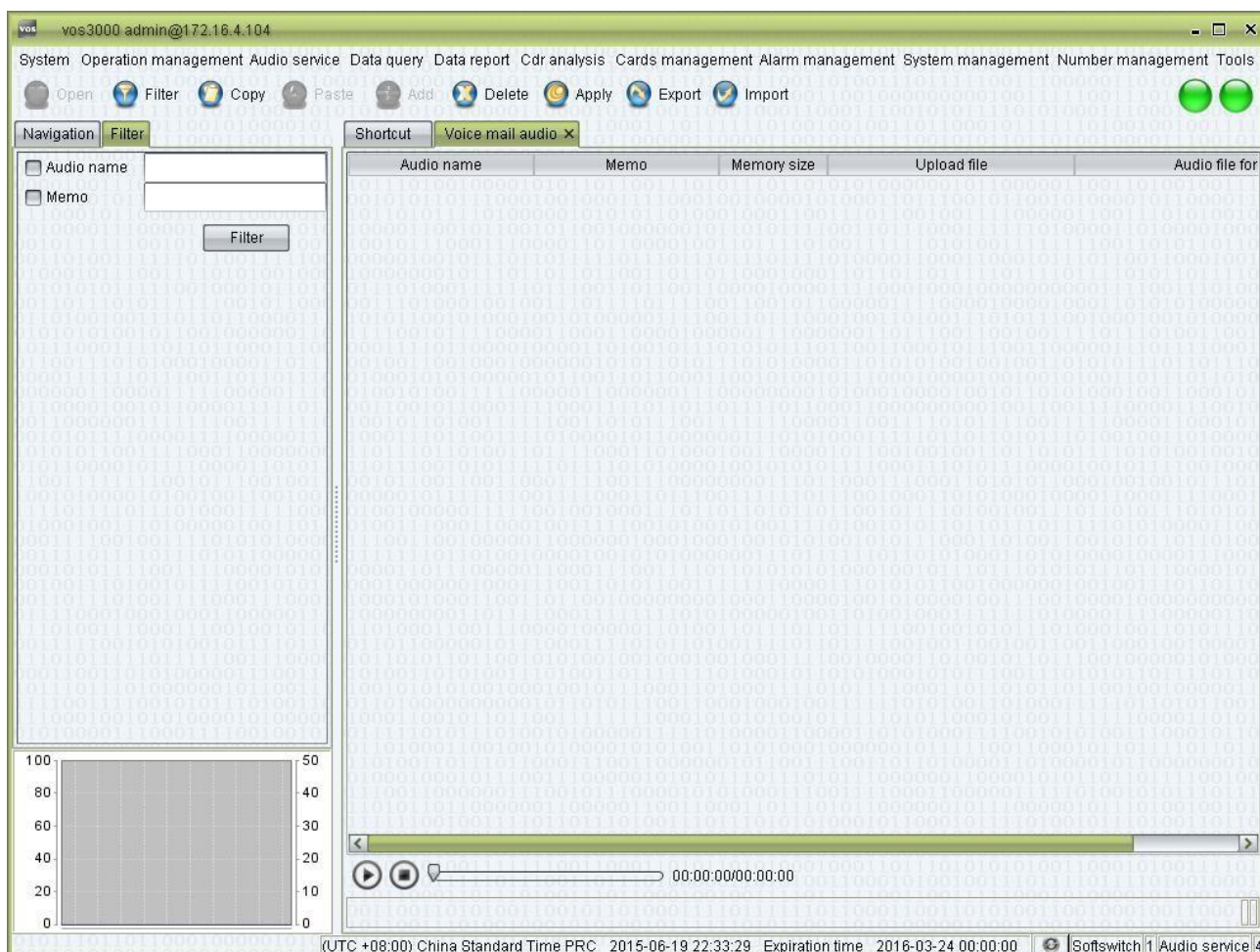
Double-click “Navigation > Audio service > Value added > Phone polyphonic ringtone”

Table Items

See “Public Audio”.

2.6.6.4 Voice Mail Audio

This function is used to manage voice mail audio, which is used in “Phone management > Supplementary service”.



How to Start

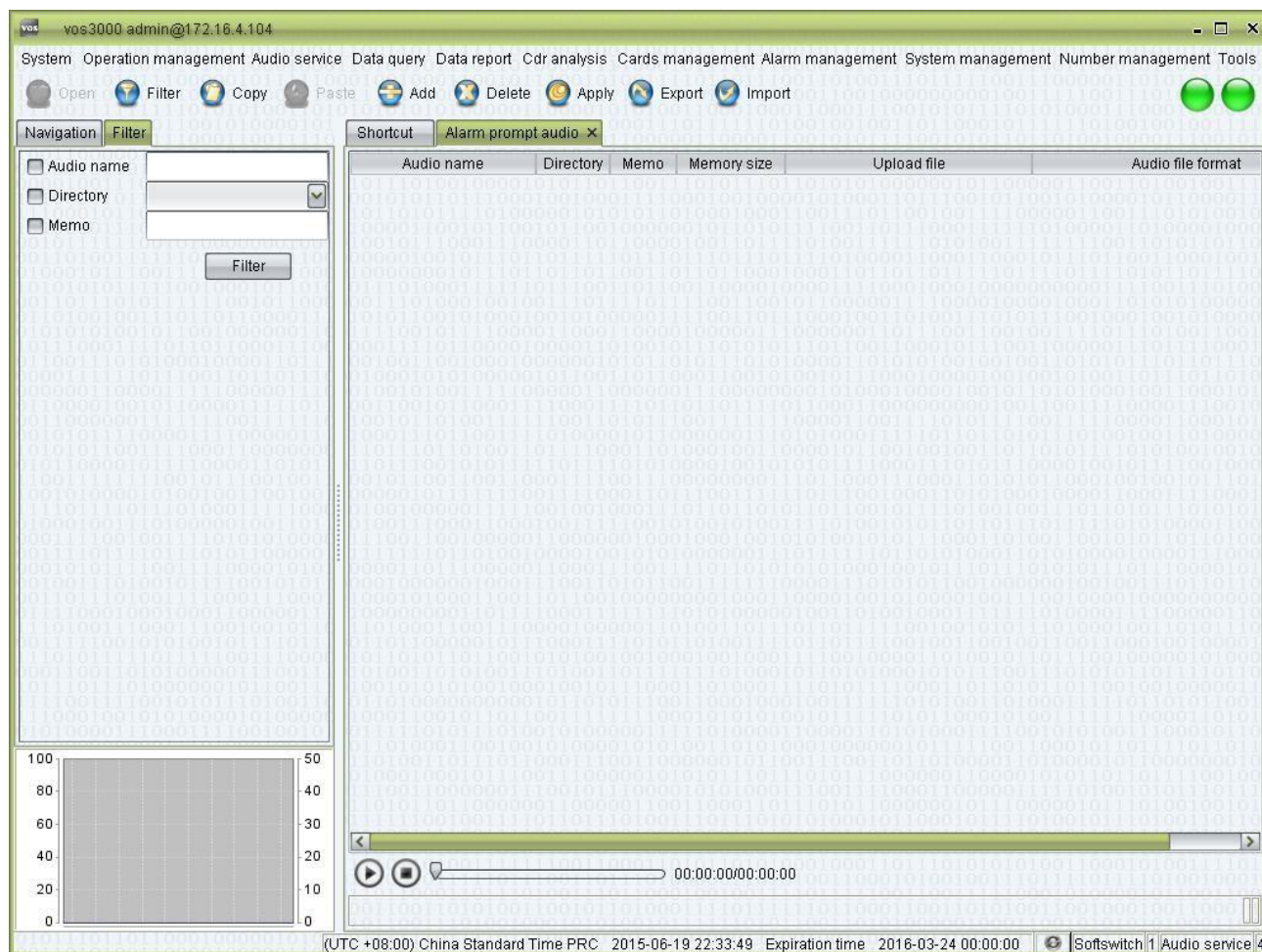
Double-click “Navigation > Audio service > Value added > Voice mail audio”

Table Items

See “Public audio”.

2.6.6.5 Alarm Prompt Audio

This function is used to manage alarm prompt audio.



How to Start

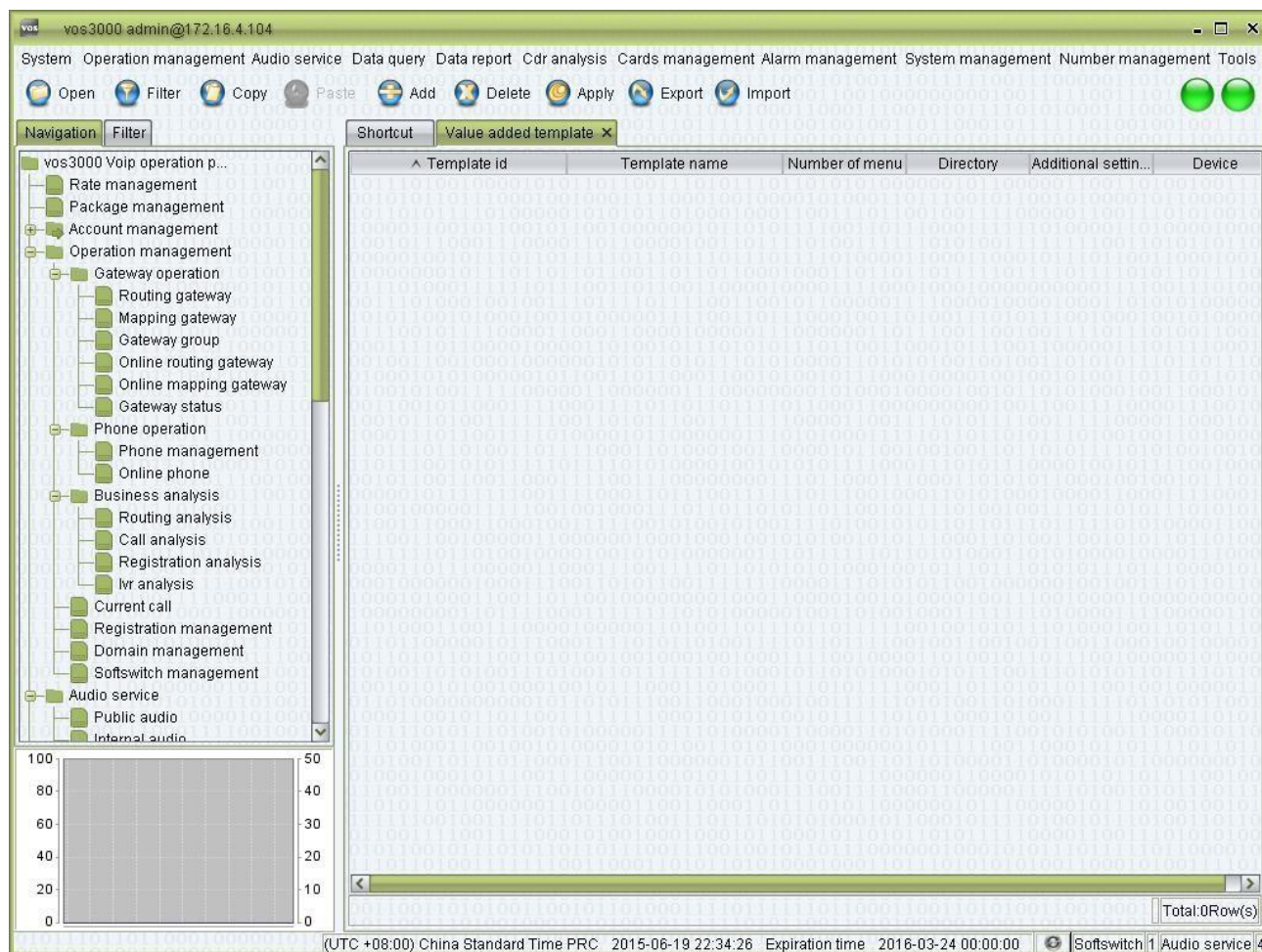
Double-click "Navigation > Audio service > Value added > Alarm prompt audio"

Table Items

See "Public audio".

2.6.6.6 Value Added Template

This function is used to define Value Added template.



How to Start

Double-click “Navigation > Audio service > Value added > Value added template”

Table Items

See “Direct/Callback template”.

2.6.6.7 Value Added Device

This function is used to manage Value Added device.

The screenshot displays the VOS3000 web interface. The top menu bar includes: System, Operation management, Audio service, Data query, Data report, Cdr analysis, Cards management, Alarm management, System management, Number management, and Tools. Below the menu bar are icons for Open, Filter, Copy, Paste, Add, Delete, Apply, Export, and Import. The left navigation pane shows a tree structure with 'Audio service' expanded, containing 'Public audio' and 'Internal audio'. The main content area is titled 'Value added device' and contains a table with the following data:

Access name	Mark	Number of service	Additional settin...	Memo	Creation time	Access ip
VALUE_ADDED	VALUE_ADDED	0	Edit		2015-03-17 15:32:02	127.0.0.1

At the bottom of the interface, there is a status bar showing: (UTC +08:00) China Standard Time PRC, 2015-06-19 22:34:42, Expiration time 2016-03-24 00:00:00, and Softswitch 1 Audio service 4.

How to Start

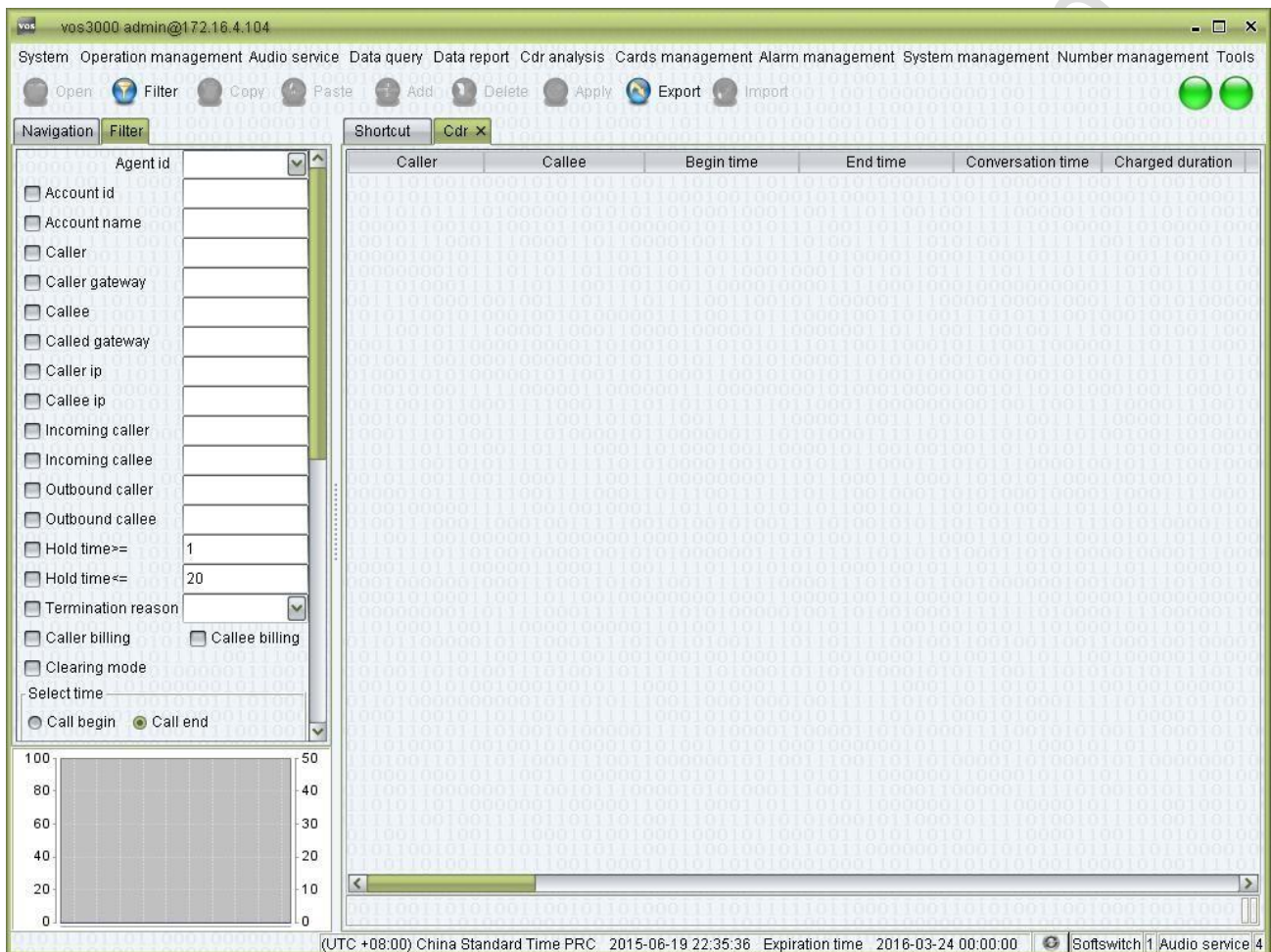
Double-click "Navigation > Audio service > Value added > Value added device"

2.7 Data Query

Historical data is independent, won't be affected by configurations, e.g. delete accounts won't remove account's CDR.

2.7.1 CDR

This function is used to query CDR.



How to Start

Double-click "Navigation > Data query > Cdr"

Table Items

By default, the system displays 1000 records every page. This number can be changed in "System parameter > SERVER_QUERY_ONE_PAGE_SIZE".

Caller: the phone number of the caller.

Callee: the phone number of the caller.

Begin time: the time when the call is initiated.

End time: the time the call lasts.



NOTE

In Normal Mode, End Time is begin time + conversation time.

In Clearing Mode, End Time is the actual stop time.

Accuracy of billing time makes the difference, e.g. call from 201304018:13:34:40.002 to 201304018:13:34:41.003, billing duration 1.001 second, according to accuracy should treat as 2 second, and display 201304018:13:34:42.002.

Conversation time: Actual duration.

Charged duration: the time used for billing, which is calculated according the Billing Cycle specified in rate policies.

Call charges: the fee charged for this call.

Talk taxes

Call expense: the cost of delivering this call.

Cost taxes

Termination reason: see Appendix for details.

Hangup side

Caller gateway: the ID of the gateway between the caller and the Softswitch.

Called gateway: the ID of the gateway between the called and the Softswitch.

Caller ip: the IP address of the caller.

Callee ip: the IP address of the called.

Account name: the name of the account used for billing this call.

Account id: the number of the account used for billing this call.

Agent id

Call type: "Network", "Local", "Domestic" or "International".



NOTE

Call type depends on the matched fee rate's Rate Type.

Area prefix: the prefix used for billing this call.

Area name: obtain the name of the prefix according to the area prefix.

Incoming caller: original caller number sent to server.

Incoming callee: original callee number sent to server.

Outbound caller: the caller number sent to the called after the application of rewrite rules.

Outbound callee: the called number sent to the called after the application of rewrite rules.

Calling device name

Called device name

Package duration

Package charges

Billing method

- By caller: billing on caller account.
- By callee: billing on callee account.

Charge mode

Continue duration: Time elapsed from platform received to call connected.

Calling Call-Id

Called Call-Id

Reason: Vos will add "reason" filed in hangup signaling, used to transparent peer's "reason" filed which includes hangup reason.

Serial number

Connect delay (PDD): Time elapsed from send call to routing response.



NOTE

PDD is defined as the time from send call to callee to receive signal (switch routing gateway will recalculate), see detail below:

Callee is SIP

1. Receive callee's 180/200
2. Receive callee's 486/600
3. If callee is phone, callee's signal contains SDP
4. If callee is routing gateway and enabled "Stop Switch Gateway After Receive SDP", callee's signal contains SDP

Callee is H323

1. Receive callee's Progress/Alerting/Connect

2. Receive callee's ReleaseComplete (UserBusy)
3. Callee is phone, callee's Q931 contains faststart or H245 contains openlogicalchannel
4. Callee is routing gateway and enabled "Stop Switch Gateway After OLC", callee's Q931 contains faststart or H245 contains openlogicalchannel

Media Proxy Enabled

1. Callee is phone, receive callee's RTP
2. Callee is routing gateway and enabled "Stop Switch Gateway when RTP Start", receive callee's RTP

Right-Click Menu

Corrected time: Correction of the start time. Only integers are supported (Unit: second)

Call analysis: Open the "Call analysis" page.

2.7.2 Payment Record

This function is used to query payment.

vos3000 admin@172.16.4.104

System Operation management Audio service Data query Data report Cdr analysis Cards management Alarm management System management Number management Tools

Open Filter Copy Paste Add Delete Apply Export Import

Navigation Filter

Agent id

☐ Account id

☐ Account name

☐ Type Payment

☒ Begin time 2015-06-26 00:00:00

☐ End time 2015-06-27 00:00:00

☐ Query all

Mode

☐ Standard ☐ Phone card

Filter

Download

Payment statistic

Total money amount 0

100 80 60 40 20 0

50 40 30 20 10 0

Account id Account name Money amount Account bala... Type Payment time Mode Payment user

(UTC +08:00) China Standard Time PRC 2015-06-19 22:43:23 Expiration time 2016-03-24 00:00:00 Softswitch 1 Audio service 4

How to Start

Double-click "Navigation > Data query > Payment record"

Table Items

Account id

Account name

Payment amount: the amount.

Account balance: the balance after payment.

Type: "Create Account", "Credit" or "Payment".

Payment time: Time of the payment.

Mode: Method of the payment.

Payment user: the name of the user that fulfills this payment.

Memo: Comments on the payment.

Agent id

Agent name

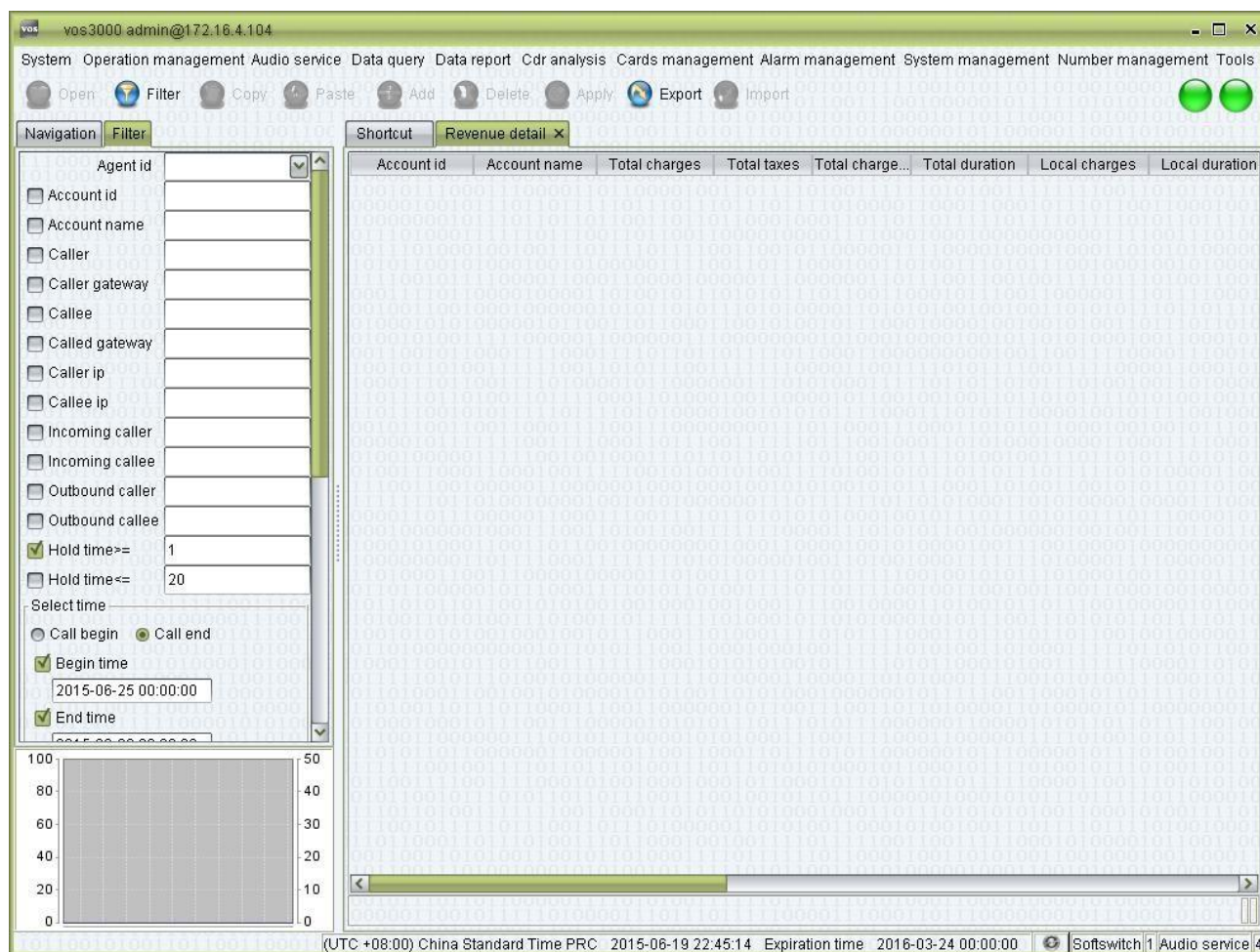
Serial number

Downloaded from VOS3000.Com

2.7.3 Bill Query

2.7.3.1 Revenue Details

This function is used to query account's consumption.



How to Start

Double-click "Navigation > Data query > Bill query > Revenue details"

Table Items

Account id
Account name
Total charges
Total taxes
Total charges & taxes

Total duration

Local charges: total charge of CDRs which call type is Local.

Local duration: total duration of CDRs which call type is Local.

Domestic charges: total charge of CDRs which call type is Domestic.

Domestic duration: total duration of CDRs which call type is Domestic.

International charges: total charge of CDRs which call type is International.

International duration: total duration of CDRs which call type is International.

Intranet charges: total charge of CDRs which call type is Net.

Intranet duration: total duration of CDRs which call type is Net.

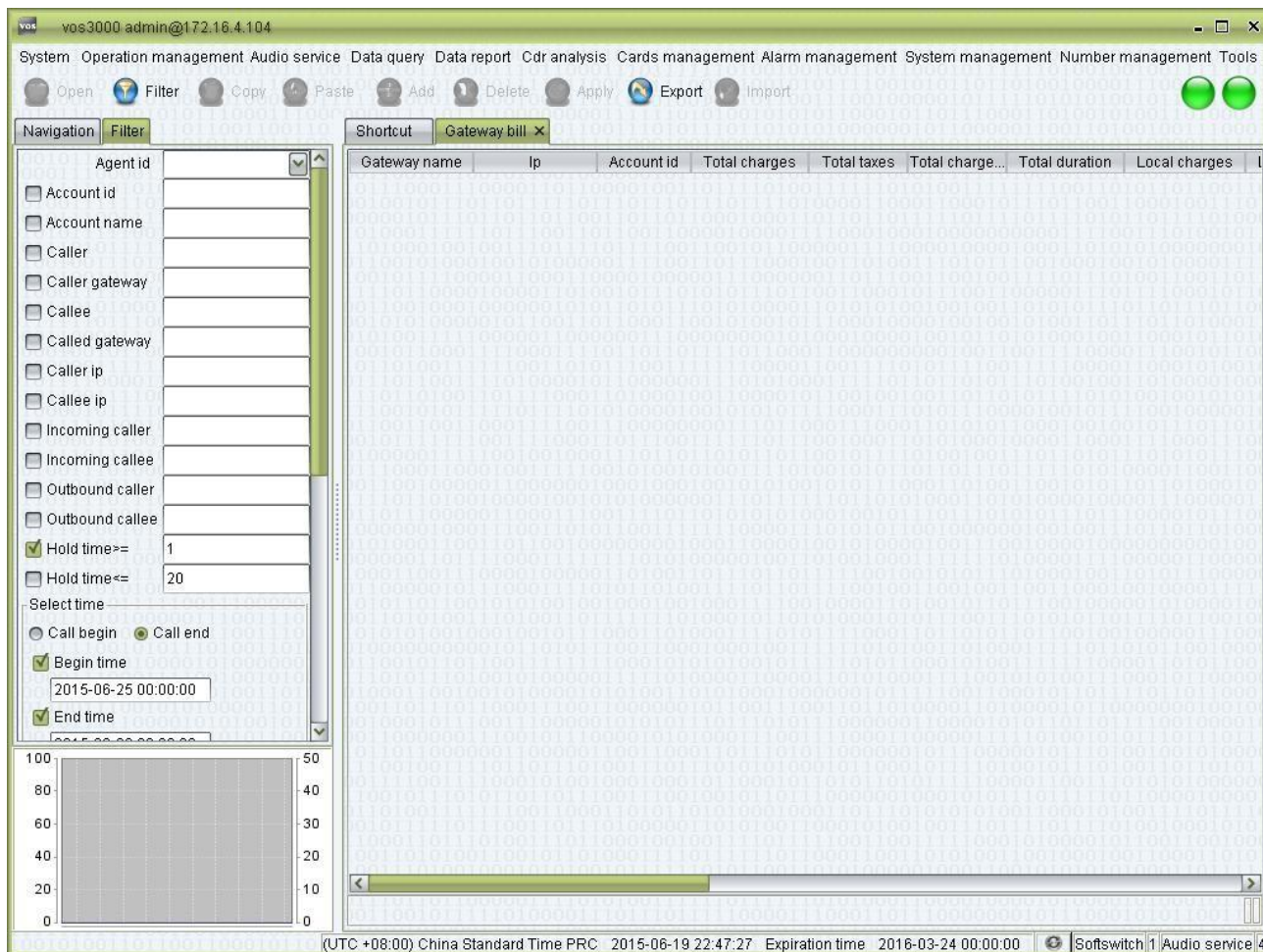
Total package: total charge of package's free money amount.

Package duration: total duration of package's free duration.

Number of CDR

2.7.3.2 Gateway Bill

This function is used to query mapping gateway's consumption.



How to Start

Double-click "Navigation > Data query > Bill query > Gateway bill"

2.7.3.3 Phone Bill

This function is used to query phone's consumption.

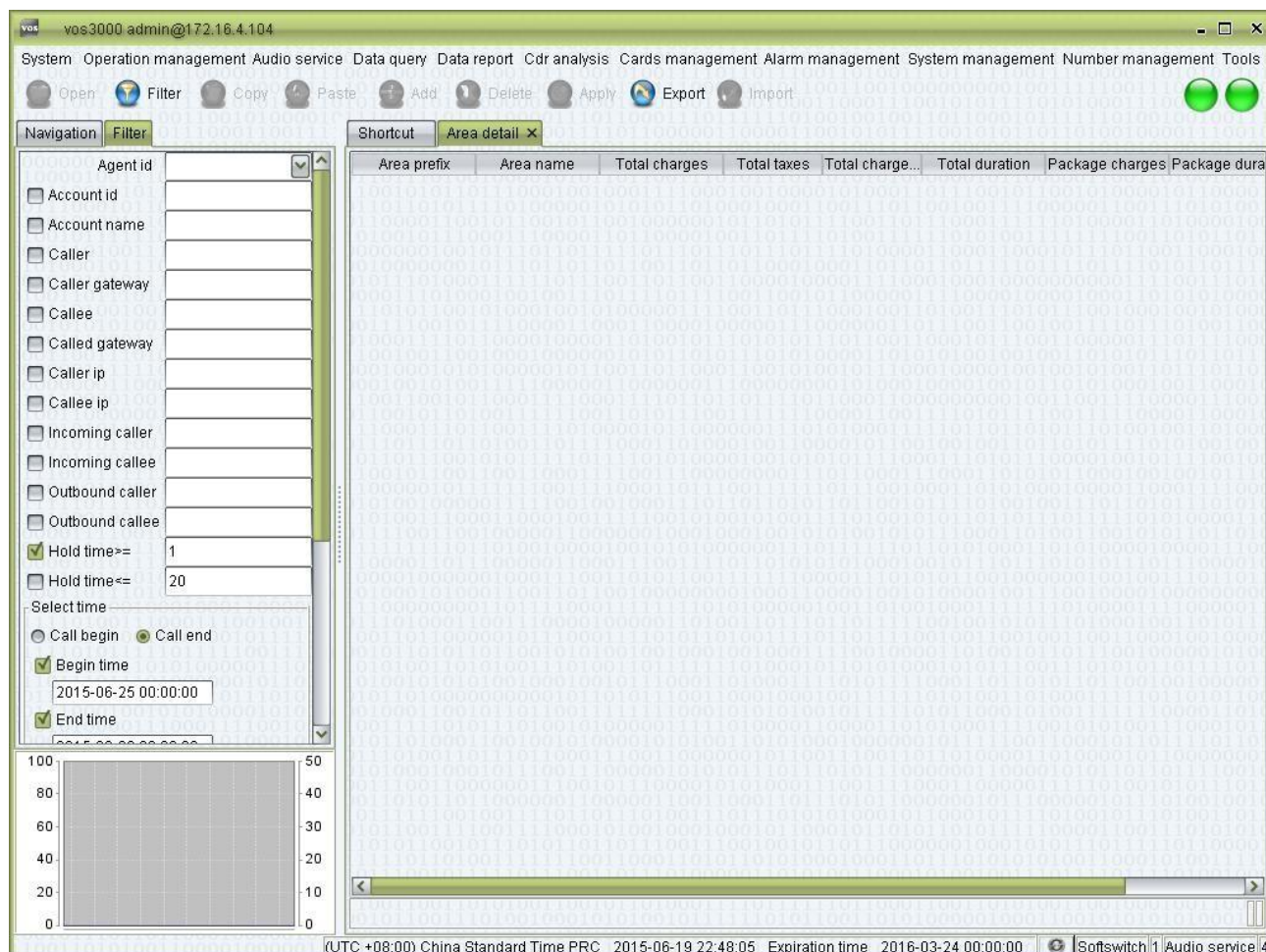
The screenshot displays the VOS3000 Phone Bill query interface. The window title is "vos3000 admin@172.16.4.104". The menu bar includes "System", "Operation management", "Audio service", "Data query", "Data report", "Cdr analysis", "Cards management", "Alarm management", "System management", "Number management", and "Tools". The toolbar has buttons for "Open", "Filter", "Copy", "Paste", "Add", "Delete", "Apply", "Export", and "Import". The left sidebar shows a "Navigation" tree with "Filter" selected. The main area is titled "Phone bill" and contains a table with columns: "Phone number", "Account id", "Billing method", "Total charges", "Total taxes", "Total charge...", "Total duration", and "Local charges". The table is filled with binary data. Below the table is a status bar showing "(UTC +08:00) China Standard Time PRC 2015-06-19 22:47:49", "Expiration time 2016-03-24 00:00:00", and "Softswitch 1 Audio service 4".

How to Start

Double-click "Navigation > Data query > Bill query > Phone bill"

2.7.3.4 Area Details

This function is used to query details of each area.

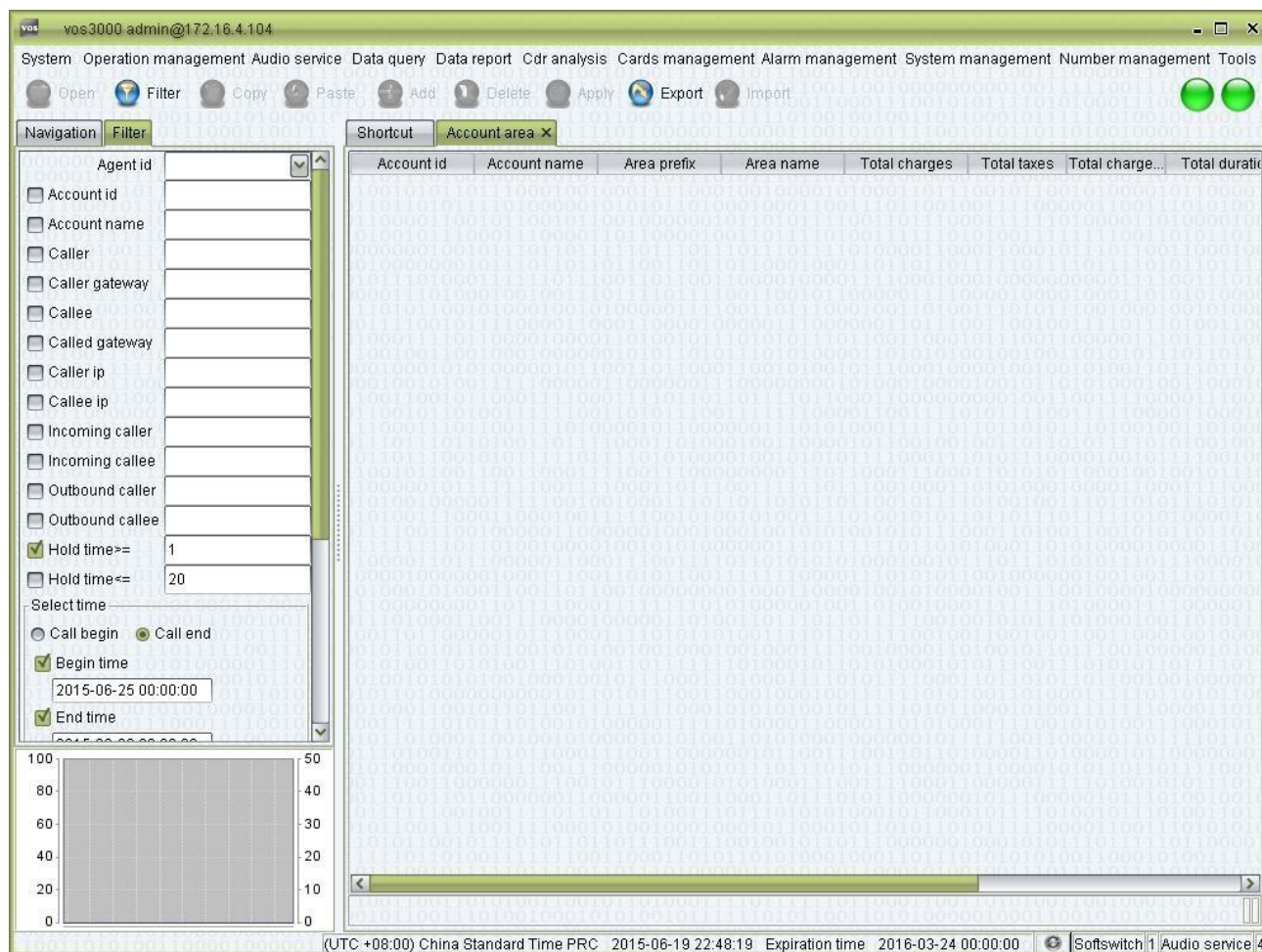


How to Start

Double-click "Navigation > Data query > Bill query > Area details"

2.7.3.5 Account Area

This function is used to query area consumption of account.



How to Start

Double-click "Navigation > Data query > Bill query > Account area"

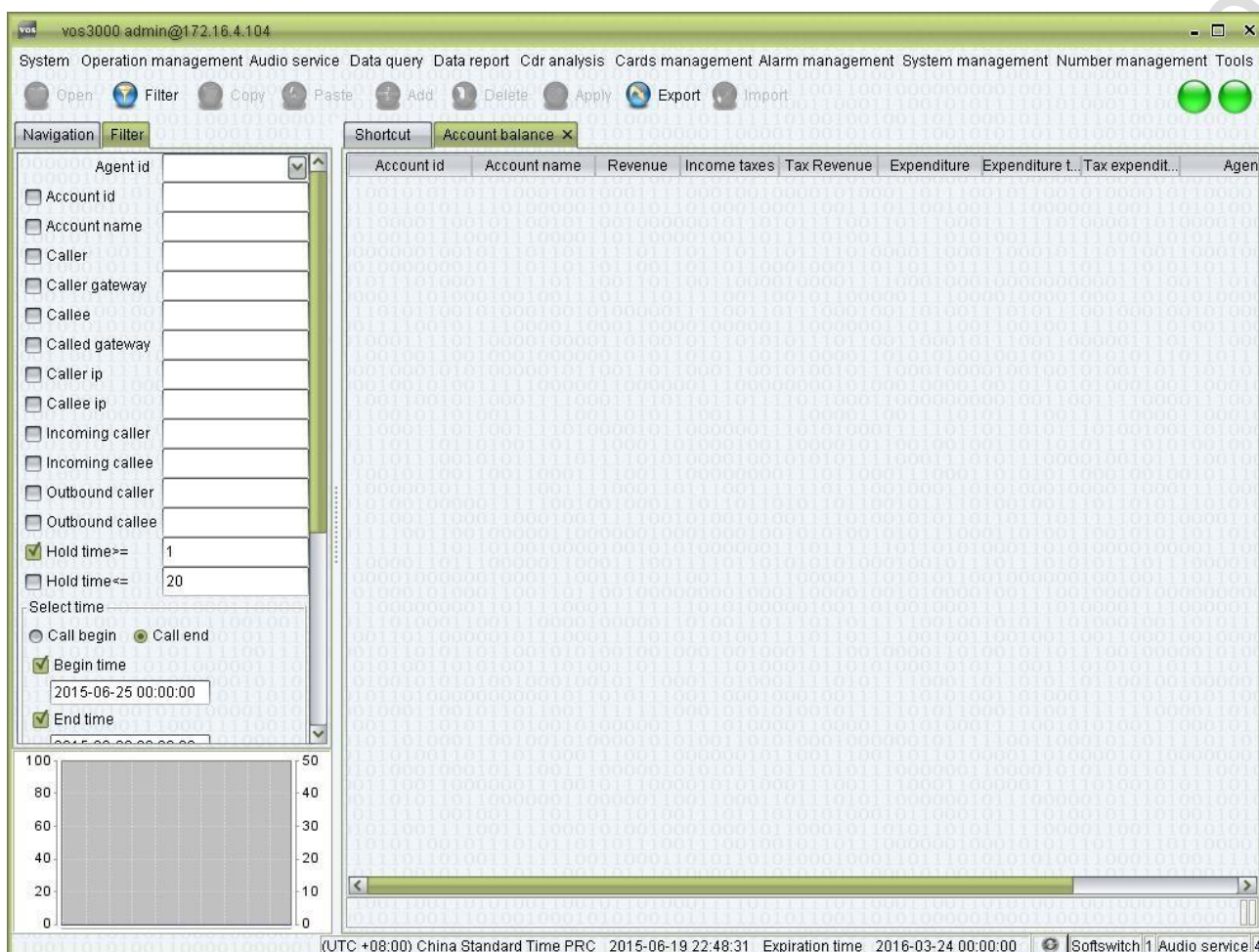
2.7.3.6 Account Balance

This function is used to query account and agent's revenue and expenditure.

NOTE

Revenue is only for agent, which is the total expenditure of sub accounts.

For ordinary accounts, there will only be expenditures in the table.



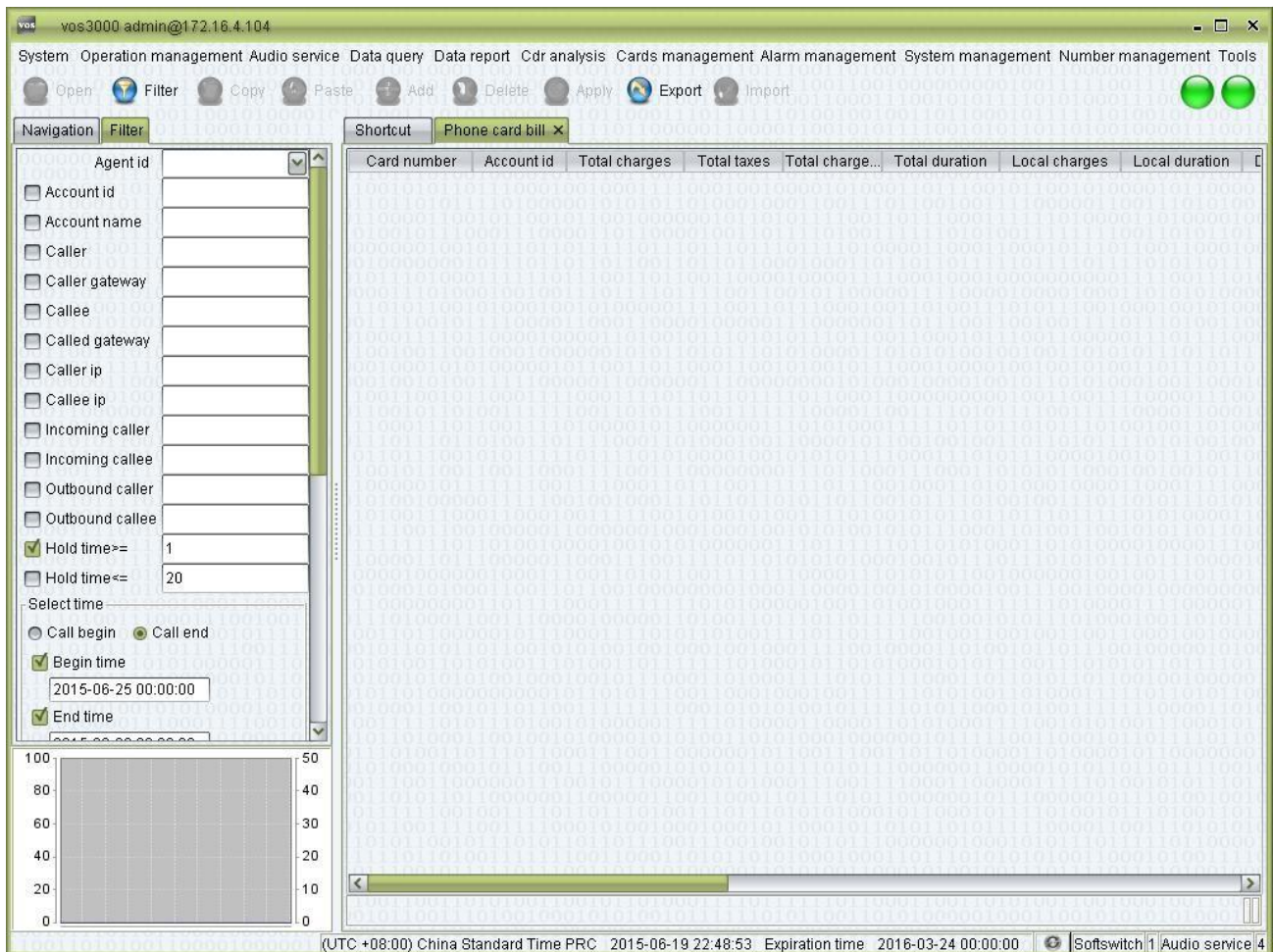
How to Start

Double-click "Navigation > Data query > Bill query > Account balance"

2.7.4 Cards Query

2.7.4.1 Phone Card Bill

This function is used to query phone card bill.

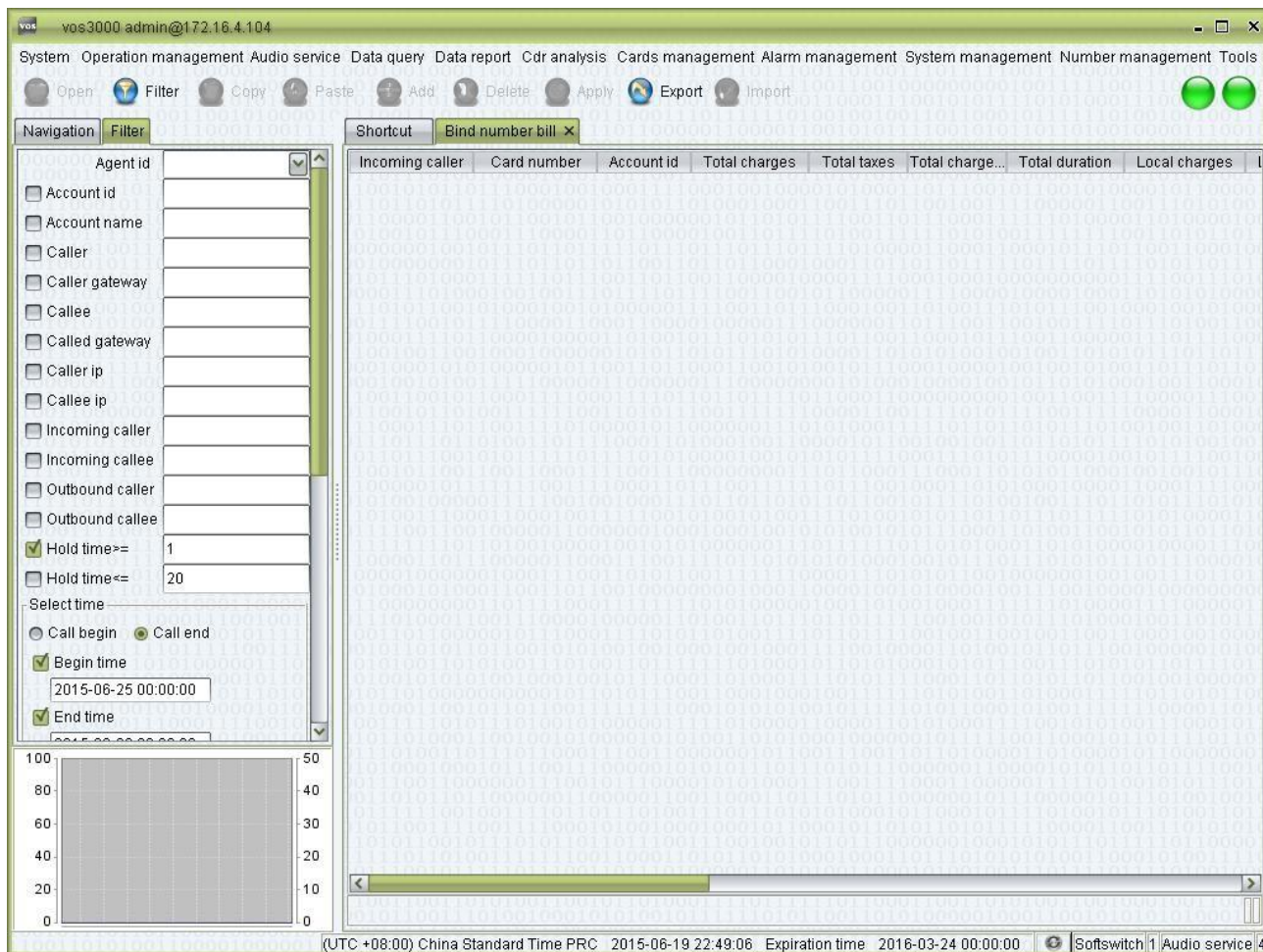


How to Start

Double-click "Navigation > Data query > Cards query > Phone card bill"

2.7.4.2 Bind Number Bill

This function is used to query bind number bill.



How to Start

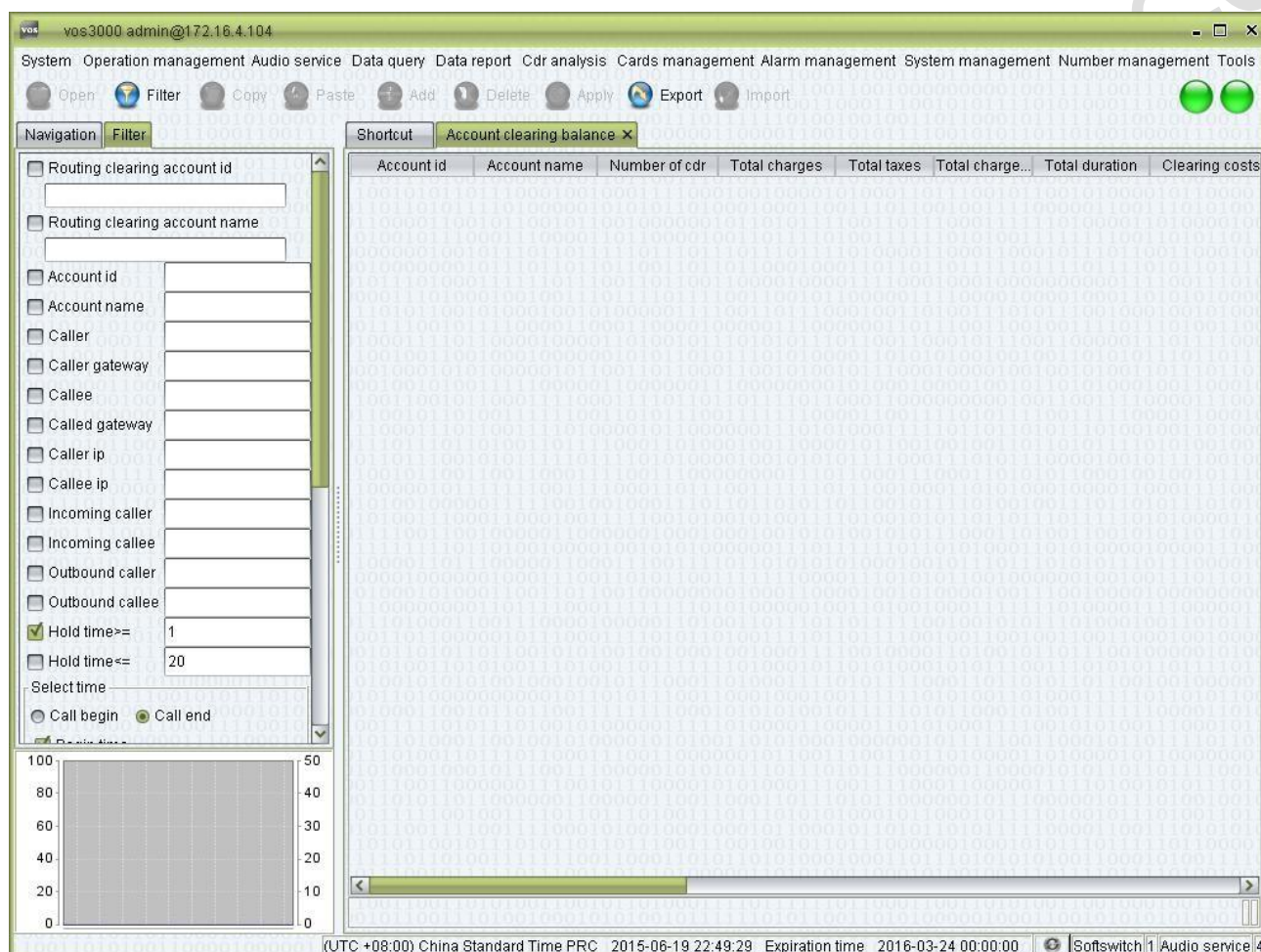
Double-click "Navigation > Data query > Cards query > Bind number bill"

2.7.5 Clearing Query

2.7.5.1 Account Clearing Balance

This function is used to query account's clearing consumption.

Display the expenditures on each account on different clearing account in order to quickly calculate operating profits.



How to Start

Double-click "Navigation > Data query > Clearing query > Account clearing balance"

Table Items

Clearing costs: total charge of clearing account.

Clearing duration: total duration of clearing account.

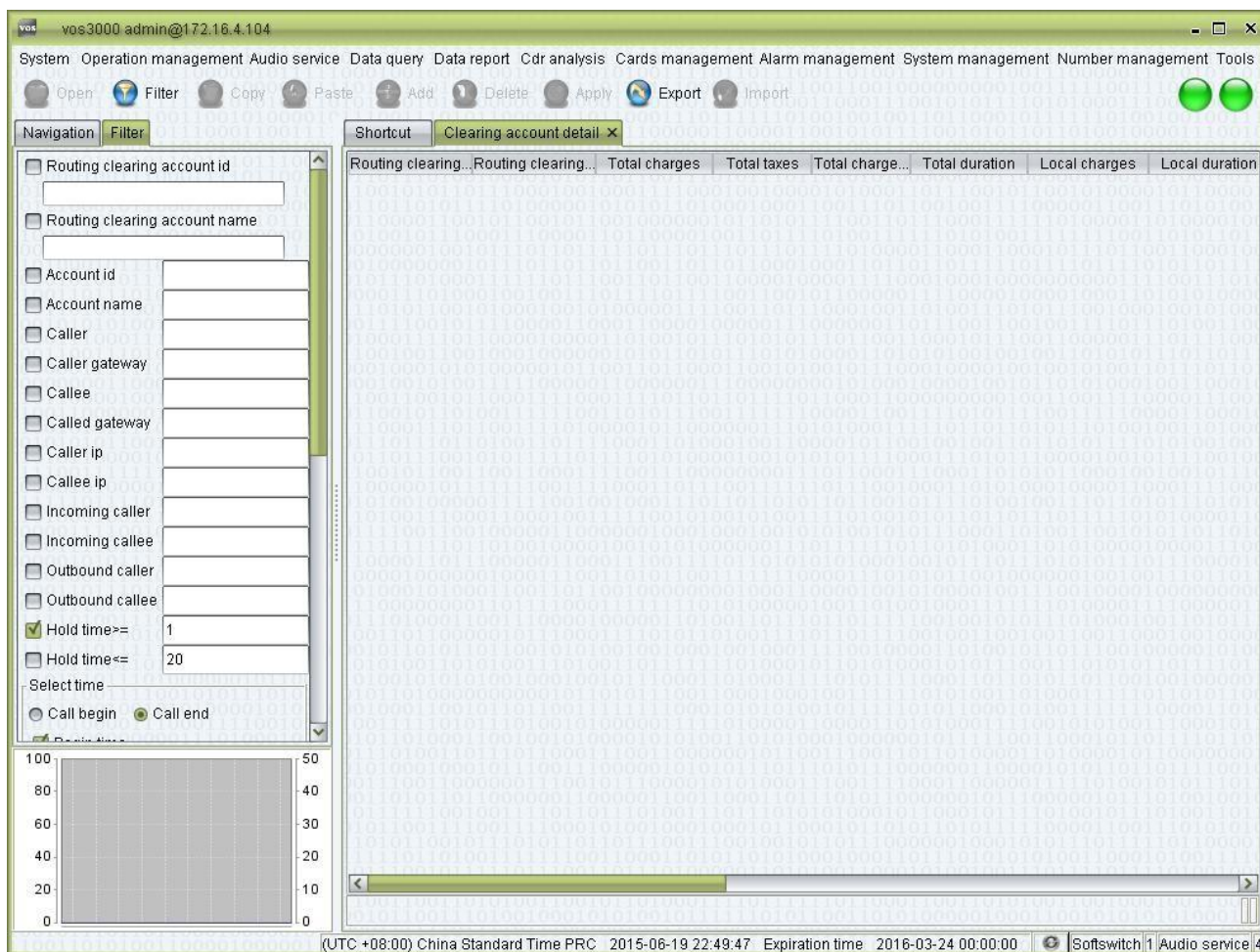
Clearing package amount: total package amount of clearing account.

Clearing package duration: total package duration of clearing account.

Downloaded from VOS3000.Com

2.7.5.2 Clearing Account Detail

This function is used to query clearing account's consumption.



How to Start

Double-click "Navigation > Data query > Clearing query > Clearing account detail"

2.7.5.3 Clearing Gateway Details

This function is used to query routing gateway's consumption.

The screenshot displays the VOS3000 admin interface. The top menu bar includes System, Operation management, Audio service, Data query, Data report, Cdr analysis, Cards management, Alarm management, System management, Number management, and Tools. Below the menu is a toolbar with icons for Open, Filter, Copy, Paste, Add, Delete, Apply, Export, and Import. The left sidebar has a 'Navigation' tab and a 'Filter' tab. The 'Filter' tab is active, showing a list of filters: Routing clearing account id, Routing clearing account name, Account id, Account name, Caller, Caller gateway, Callee, Called gateway, Caller ip, Callee ip, Incoming caller, Incoming callee, Outbound caller, Outbound callee, Hold time >= 1, and Hold time <= 20. The 'Hold time' section is expanded, showing a checkbox for 'Hold time >=' and a value of 1. The 'Select time' section is also expanded, showing radio buttons for 'Call begin' and 'Call end'. The main table area has a 'Shortcut' tab and a 'Clearing gateway detail' tab. The table has columns: Gateway name, Ip, Total charges, Total taxes, Total charge..., Total duration, Package charges, and Package dt. The table is filled with binary data. The bottom status bar shows the time as (UTC +08:00) China Standard Time PRC 2015-06-19 22:50:17 and the expiration time as 2016-03-24 00:00:00. The status bar also shows 'Softswitch 1 Audio service 4'.

How to Start

Double-click "Navigation > Data query > Clearing query > Clearing gateway detail"

2.8 Data Report

Users can specify in the “System parameter” whether to generate certain data report.

SERVER_REPORT_CLEARING_CUSTOMER_FEE	Off	Auto generate clearing account detail report
SERVER_REPORT_CLEARING_CUSTOMER_IO	Off	Auto generate account clearing balance report
SERVER_REPORT_CLEARING_CUSTOMER_LOCATION_FEE	On	Auto generate clearing account area detail report
SERVER_REPORT_CLEARING_GATEWAY_FEE	Off	Auto generate clearing gateway detail report
SERVER_REPORT_CUSTOMER_FEE	On	Auto generate revenue detail report
SERVER_REPORT_CUSTOMER_IO	Off	Auto generate account balance report
SERVER_REPORT_CUSTOMER_LOCATION_FEE	On	Auto generate account area detail report
SERVER_REPORT_GATEWAY_FEE	On	Auto generate gateway bill report
SERVER_REPORT_GATEWAY_MAPPING_ASR_ACD	Off	Auto generate mapping gateway connect analysis report
SERVER_REPORT_GATEWAY_ROUTING_ASR_ACD	Off	Auto generate routing gateway connect analysis report
SERVER_REPORT_PHONE_CARD_E164_FEE	Off	Auto generate bind number bill report
SERVER_REPORT_PHONE_CARD_FEE	Off	Auto generate phone card bill report
SERVER_REPORT_PHONE_FEE	On	Auto generate phone bill report



NOTE

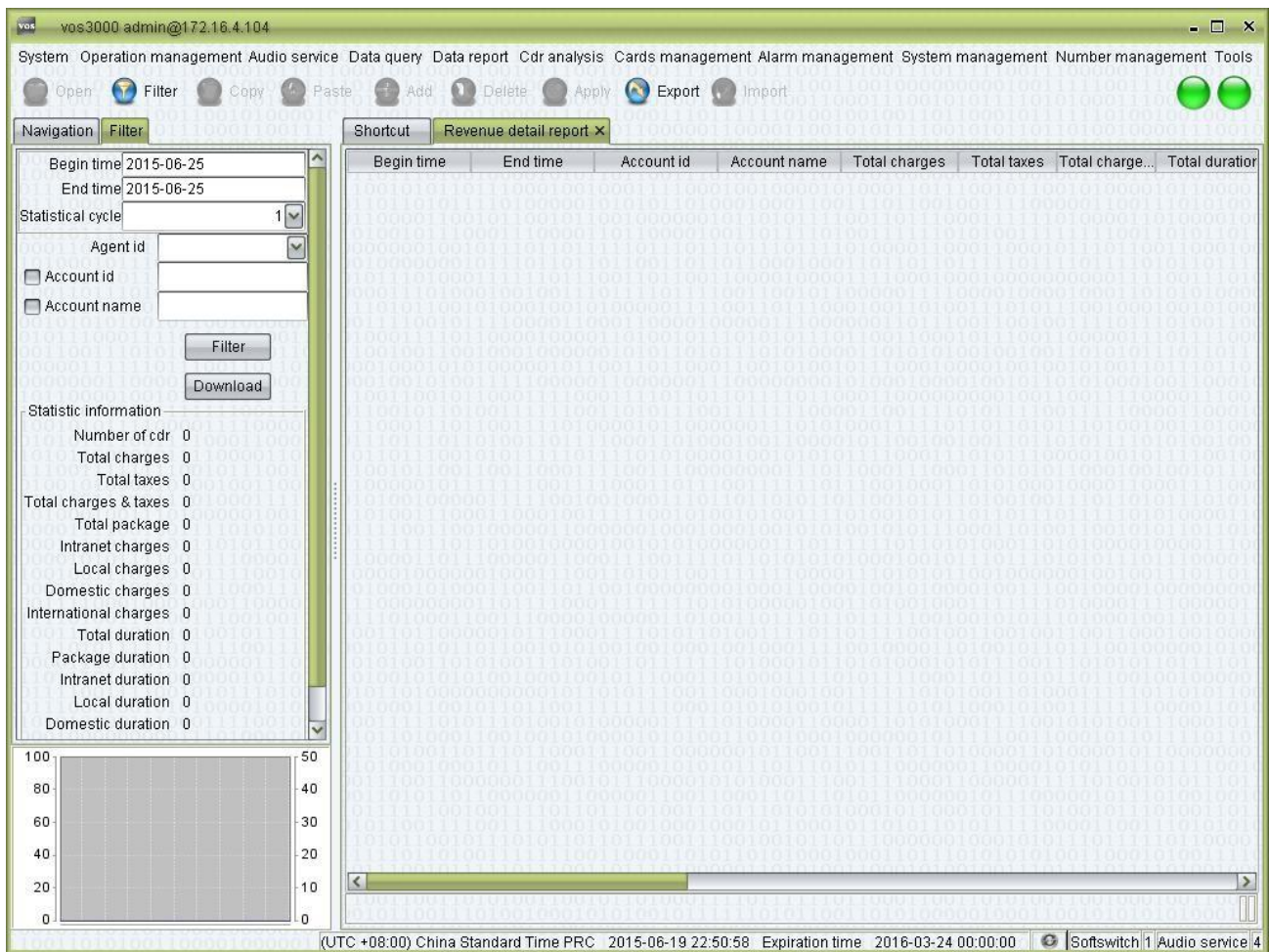
The generation of reports will begin at 1:00 A.M every day.

The time of completion depends on the capacity of the server and the amount of data.

2.8.1 Bill Report

2.8.1.1 Revenue Details Report

This function is used to query account's consumption report.



How to Start

Double-click “Navigation > Data report > Bill report > Revenue detail report”

Table Items

Begin time

End time

Account id: the number of the account being displayed.

Account name: the name of the account being displayed.

Total charges: the total amount of charges.

Total taxes

Total charges & taxes

Total duration: the total amount of session time for all calls.

Local charges: the amount charged for local calls.

Local duration: the amount of session time for local calls.

National charges: the amount charged for national calls.

National duration: the amount of session time for national calls.

International charges: the amount charged for international calls.

International duration: the amount of session time for international calls.

Intranet charges: the amount charged for net calls.

Intranet duration: the amount of session time for net calls.

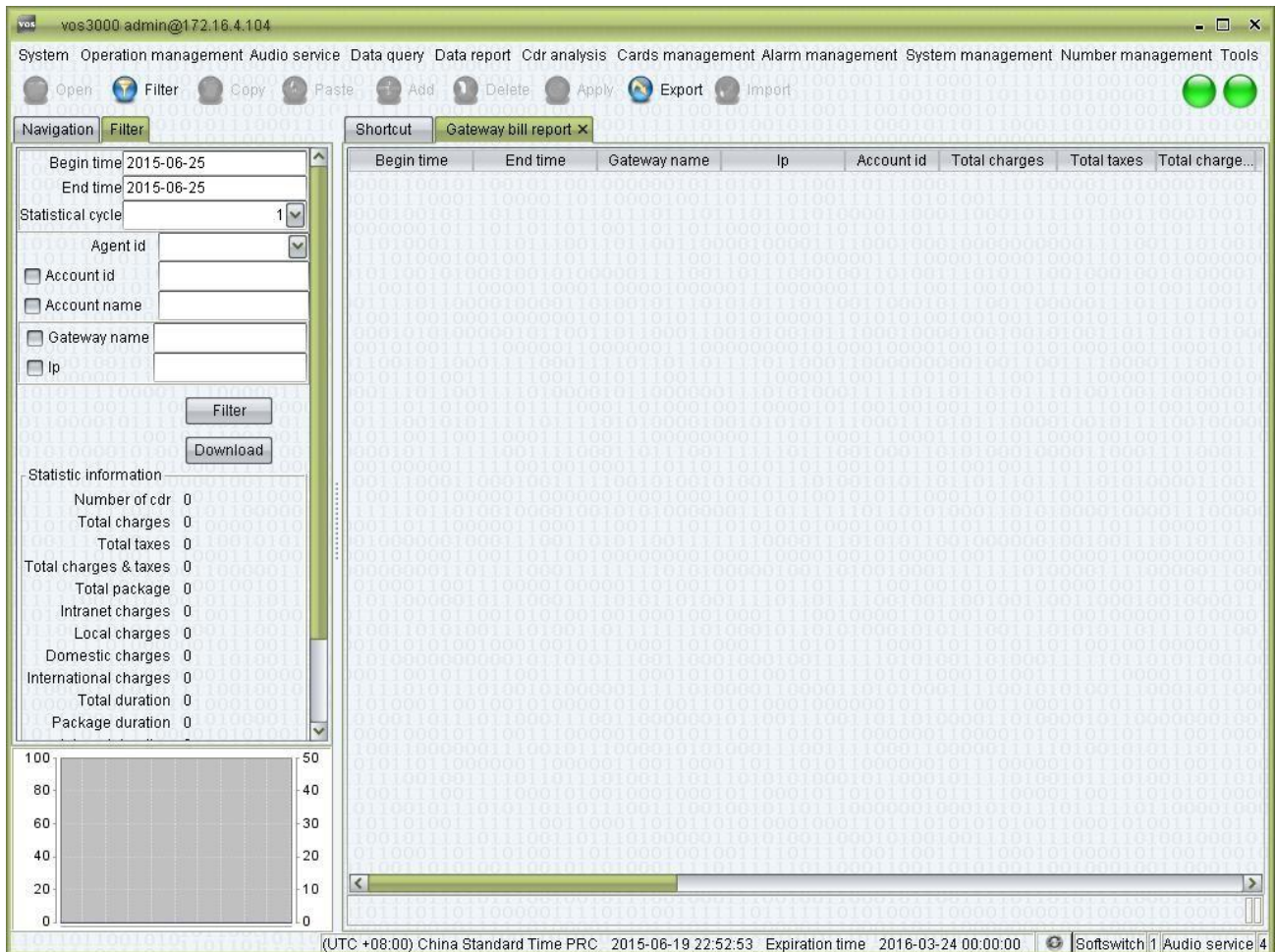
Total package: the total consumption of gift amount.

Package duration: the total consumption of free duration.

Number of CDR: the total number of phone records.

2.8.1.2 Gateway Bill Report

This function is used to query mapping gateway's consumption report.



How to Start

Double-click “Navigation > Data report > Bill report > Gateway bill report”

Table Items

Gateway id: the unique id of the device, used for the authentication of dynamic gateways. For static gateways (usually relay gateways), the only requirement is that their ids do not conflict with one another.

IP: the IP address of the gateway.

Please refer to the descriptions in “Revenue detail report” for further instructions.

2.8.1.3 Phone Bill Report

This function is used to query phone's consumption report.

The screenshot shows the VOS3000 Phone Bill Report interface. The window title is 'vos3000 admin@172.16.4.104'. The menu bar includes System, Operation management, Audio service, Data query, Data report, Cdr analysis, Cards management, Alarm management, System management, Number management, and Tools. The toolbar has buttons for Open, Filter, Copy, Paste, Add, Delete, Apply, Export, and Import. The left sidebar shows a navigation tree with 'Phone bill report' selected. The main area displays a table with columns: Begin time, End time, Phone number, Account id, Billing method, Total charges, Total taxes, and Total charge... The table is currently empty. Below the table is a status bar showing '(UTC +08:00) China Standard Time PRC 2015-06-19 22:53:24 Expiration time 2016-03-24 00:00:00' and 'Softswitch 1 Audio service 4'.

How to Start

Double-click "Navigation > Data report > Bill report > Phone bill report"

Table Items

Phone number: the number used as caller id and the called number for the terminal.

Account id: the number of the account that the phone belongs to.

Billing type: whether the caller or the called is charged.

Please refer to the descriptions in "Revenue detail report" for further instructions.

2.8.1.4 Account Area Report

This function is used to query area consumption of account report.

The screenshot shows the VOS3000 web interface for the 'Account area report'. The top navigation bar includes 'System', 'Operation management', 'Audio service', 'Data query', 'Data report', 'Cdr analysis', 'Cards management', 'Alarm management', 'System management', 'Number management', and 'Tools'. The 'Data report' menu is expanded, showing 'Open', 'Filter', 'Copy', 'Paste', 'Add', 'Delete', 'Apply', 'Export', and 'Import'. The 'Filter' sub-menu is active, displaying a form with the following fields:

- Begin time: 2015-06-25
- End time: 2015-06-25
- Statistical cycle: 1
- Agent id: [dropdown]
- ☐ Account id
- ☐ Account name
-
-

Below the filter form is the 'Area detail statistics' section, which lists the following metrics:

- Total charges: 0
- Total taxes: 0
- Total charges & taxes: 0
- Package charges: 0
- Total duration: 0
- Package duration: 0
- Number of cdr: 0

The main data table has the following columns: 'Begin time', 'End time', 'Account id', 'Account name', 'Area prefix', 'Area name', 'Total charges', and 'Total'. The table contains a large number of rows, each representing a record of area consumption. At the bottom of the interface, there is a status bar showing the current time (2015-06-19 22:54:08), expiration time (2016-03-24 00:00:00), and the user's session information (Softswitch 1 Audio service 4).

How to Start

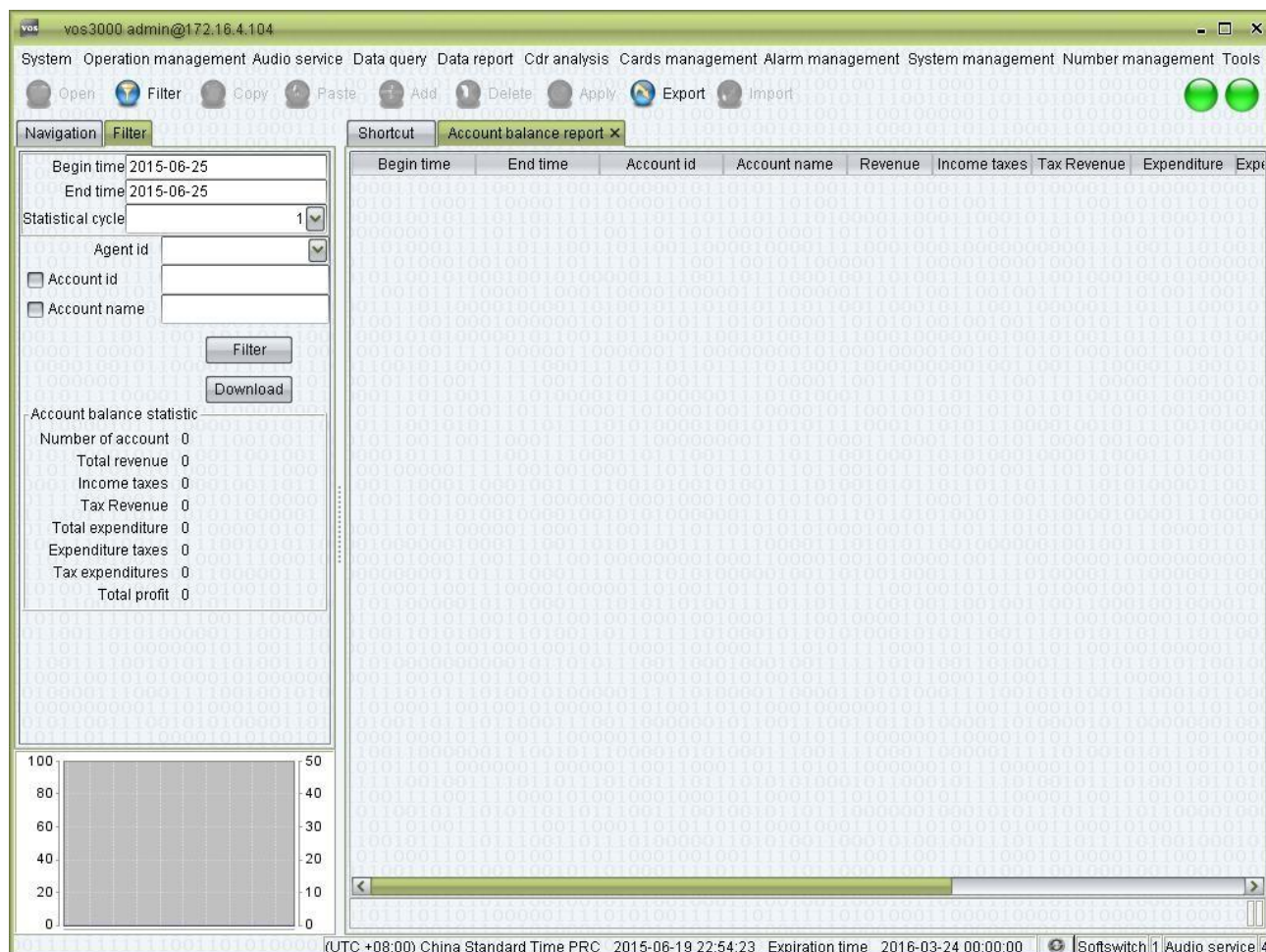
Double-click "Navigation > Data report > Bill report > Account area report"

Table Items

See the descriptions in "Revenue detail report".

2.8.1.5 Account Balance Report

This function is used to query account and agent's revenue and expenditure report.



How to Start

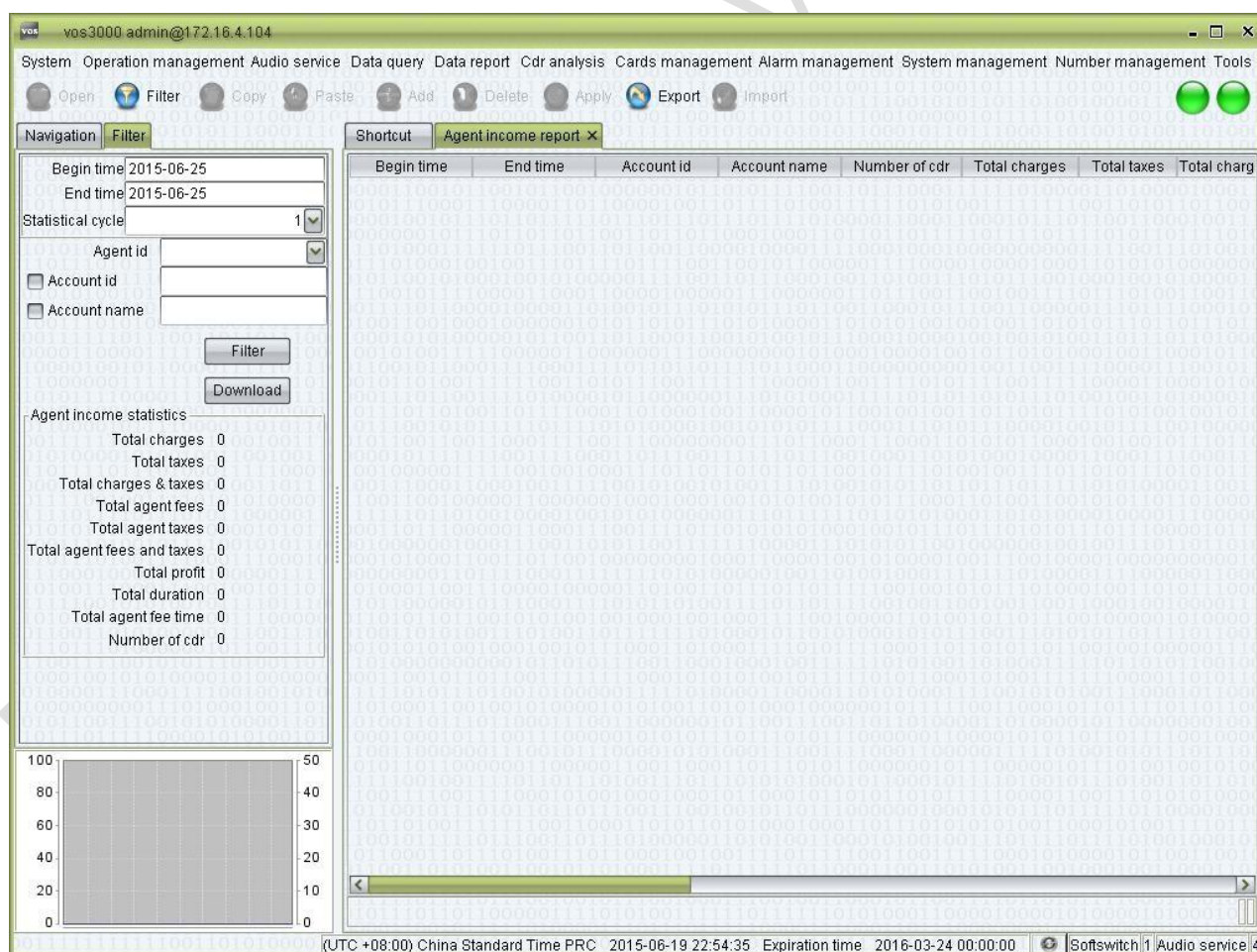
Double-click "Navigation > Data report > Bill report > Account balance report"

Table Items

See the descriptions in "Revenue detail report".

2.8.1.6 Agent Income Report

This function is used to display each sub-account under the agents account make profits for the agents. However Account Balance Report only display agents account income and expenses overall.



How to Start

Double-click “Navigation > Data report > Bill report > Agent income report”

Table Items

Total charges: The sum of the expenses generated by the agents sub-account.

Total agent fees: The sum of agent cost for this sub-account's consumption.

Downloaded from VOS3000.Com

2.8.2 Cards Report

2.8.2.1 Phone Card Bill Report

This function is used to query phone card report.

The screenshot displays the VOS3000 software interface. The title bar shows 'vos3000 admin@172.16.4.104'. The menu bar includes: System, Operation management, Audio service, Data query, Data report, Cdr analysis, Cards management, Alarm management, System management, Number management, Tools. The toolbar contains icons for Open, Filter, Copy, Paste, Add, Delete, Apply, Export, and Import. The navigation pane on the left shows a tree structure with 'Navigation' and 'Filter' selected. The 'Filter' pane on the right contains fields for 'Begin time' (2015-06-25), 'End time' (2015-06-25), 'Statistical cycle' (1), 'Agent id', 'Account id', 'Account name', and 'Card number'. Below these fields are 'Filter' and 'Download' buttons. The 'Statistic information' section lists various metrics with values of 0: Number of cdr, Total charges, Total taxes, Total charges & taxes, Total package, Intranet charges, Local charges, Domestic charges, International charges, Total duration, Package duration, Intranet duration, and Local duration. The main data table has columns: Begin time, End time, Card number, Account id, Total charges, Total taxes, Total charge..., and Total duration. The table contains a large number of rows of data. The status bar at the bottom shows: (UTC +08:00) China Standard Time PRC, 2015-06-19 22:54:58, Expiration time, 2016-03-24 00:00:00, Softswitch 1, Audio service 4.

How to Start

Double-click "Navigation > Data report > Cards report > Phone card bill report"

2.8.2.2 Bind Number Bill Report

This function is used to query bind number report.

The screenshot shows the VOS3000 Bind Number Bill Report interface. The window title is "vos3000 admin@172.16.4.104". The menu bar includes "System", "Operation management", "Audio service", "Data query", "Data report", "Cdr analysis", "Cards management", "Alarm management", "System management", "Number management", and "Tools". The toolbar has buttons for "Open", "Filter", "Copy", "Paste", "Add", "Delete", "Apply", "Export", and "Import". The left sidebar shows a navigation tree with "Bind number bill report" selected. The main area displays a table with columns: "Begin time", "End time", "Incoming caller", "Card number", "Account id", "Total charges", "Total taxes", and "Total charge...". The table is currently empty. Below the table is a status bar showing "(UTC +08:00) China Standard Time PRC 2015-06-19 22:55:14 Expiration time 2016-03-24 00:00:00" and "Softswitch 1 Audio service 4".

How to Start

Double-click "Navigation > Data report > Cards report > Bind number bill report"

2.8.3 Clearing Report

2.8.3.1 Clearing Account Detail Report

This function is used to query clearing account's consumption report.

The screenshot shows the VOS3000 software interface. The title bar indicates the user is 'vos3000 admin@172.16.4.104'. The menu bar includes 'System', 'Operation management', 'Audio service', 'Data query', 'Data report', 'Cdr analysis', 'Cards management', 'Alarm management', 'System management', 'Number management', and 'Tools'. The toolbar contains icons for 'Open', 'Filter', 'Copy', 'Paste', 'Add', 'Delete', 'Apply', 'Export', and 'Import'. The 'Navigation' pane on the left shows the 'Filter' tab selected. It contains fields for 'Begin time' (2015-06-25), 'End time' (2015-06-25), and 'Statistical cycle' (1). There are checkboxes for 'Routing clearing account id' and 'Routing clearing account name', both of which are empty. Below these are 'Filter' and 'Download' buttons. A 'Statistic information' section lists various metrics with their values: Number of cdr (0), Total charges (0), Total taxes (0), Total charges & taxes (0), Total package (0), Intranet charges (0), Local charges (0), Domestic charges (0), International charges (0), Total duration (0), Package duration (0), Intranet duration (0), and Local duration (0). At the bottom left is a small empty graph. The main window displays a table with the following columns: 'Begin time', 'End time', 'Routing clearing...', 'Total charges', 'Total taxes', 'Total charge...', and 'Total duration'. The table contains many rows of data. The status bar at the bottom shows '(UTC +08:00) China Standard Time PRC 2015-06-19 22:55:33 Expiration time 2016-03-24 00:00:00' and a 'Softswitch 1 Audio service 4' indicator.

How to Start

Double-click "Navigation > Data report > Clearing report > Clearing account detail report"

Table Items

See the descriptions in "Revenue detail report".

2.8.3.2 Clearing Account Area Report

This function is used to query clearing account's area consumption report.

The screenshot shows the VOS3000 software interface. The top menu bar includes: System, Operation management, Audio service, Data query, Data report, Cdr analysis, Cards management, Alarm management, System management, Number management, and Tools. Below the menu bar is a toolbar with icons for Open, Filter, Copy, Paste, Add, Delete, Apply, Export, and Import. The main window is titled 'Clearing account area report'. On the left, there is a 'Navigation' pane with a 'Filter' button. Below it, there are input fields for 'Begin time' (2015-06-25), 'End time' (2015-06-25), and 'Statistical cycle' (1). There are also checkboxes for 'Routing clearing account id' and 'Routing clearing account name'. Below these are 'Filter' and 'Download' buttons. The main area displays a table with the following columns: Begin time, End time, Account id, Account name, Area prefix, Area name, Total charges, and Total. The table is currently empty. At the bottom, there is a status bar showing the date and time: (UTC +08:00) China Standard Time PRC 2015-06-19 22:55:54. The expiration time is 2016-03-24 00:00:00. The bottom right corner shows 'Softswitch 1 Audio service 4'.

How to Start

Double-click "Navigation > Data report > Clearing report > Clearing account area report"

Table Items

See the descriptions in "Revenue detail report".

2.8.3.3 Clearing Gateway Detail Report

This function is used to query routing gateway's consumption report.

The screenshot displays the VOS3000 web interface. The top navigation bar includes links for System, Operation management, Audio service, Data query, Data report, Cdr analysis, Cards management, Alarm management, System management, Number management, and Tools. Below this is a toolbar with icons for Open, Filter, Copy, Paste, Add, Delete, Apply, Export, and Import. The main content area is divided into two panels. The left panel, titled 'Navigation', contains a 'Filter' section with input fields for 'Begin time' (2015-06-25), 'End time' (2015-06-25), and 'Statistical cycle' (1). It also has checkboxes for 'Routing clearing account id', 'Routing clearing account name', 'Gateway name', and 'Ip'. Below these are 'Filter' and 'Download' buttons. A 'Statistic information' table is also present. The right panel, titled 'Clearing gateway detail report', shows a large table with columns: 'Begin time', 'End time', 'Gateway name', 'Ip', 'Total charges', 'Total taxes', 'Total charge...', and 'Total duration'. The table is filled with data rows. At the bottom of the interface, a status bar displays '(UTC +08:00) China Standard Time PRC 2015-06-19 22:56:09', 'Expiration time 2016-03-24 00:00:00', and 'Softswitch 1 Audio service 4'.

How to Start

Double-click "Navigation > Data report > Clearing report > Clearing gateway detail report"

Table Items

See the descriptions in "Revenue detail report".

2.8.3.4 Account Clearing Balance Report

This function is used to query account's clearing consumption report.

The screenshot displays the VOS3000 web interface. The top navigation bar includes links for System, Operation management, Audio service, Data query, Data report, Cdr analysis, Cards management, Alarm management, System management, Number management, and Tools. Below this is a toolbar with icons for Open, Filter, Copy, Paste, Add, Delete, Apply, Export, and Import. The main content area is divided into two sections. On the left, there is a 'Filter' section with input fields for 'Begin time' (2015-06-25) and 'End time' (2015-06-25), a 'Statistical cycle' dropdown (set to 1), and checkboxes for 'Routing clearing account id' and 'Routing clearing account name'. Below these are 'Filter' and 'Download' buttons. A 'Clearing balance statistic' table is also present, showing various metrics like Total charges, Total taxes, Total charges & taxes, Clearing costs, Settled taxes, Cost Total, Total profit, Total duration, Clearing duration, and Number of cdr, all with values of 0. On the right, there is a large table titled 'Account clearing balance report' with columns: Begin time, End time, Account id, Account name, Number of cdr, Total charges, Total taxes, and Total charge. The table contains multiple rows of data. At the bottom, a status bar shows the current time (2015-06-19 22:56:27) and system information (Softswitch 1 Audio service 4).

How to Start

Double-click "Navigation > Data report > Clearing report > Account clearing balance report"

Table Items

See the descriptions in "Revenue detail report".

2.8.4 Analysis Report

Total calls: total unconnected and total connected.

Total unconnected: unconnected calls.

Total connected: calls with connect/busy/no answer/ringing signaling.



NOTE

Ringing: callee sent SIP 180 or H323 Alerting.

If callee is phone, sent SIP 183 with SDP or H323 CallProceeding(OLC).

If callee is routing gateway, sent SIP 183 with SDP or H323 CallProceeding(OLC) and enable SIP "Stop switch gateway after receive sdp" or H323 "Stop switch gateway after olc".

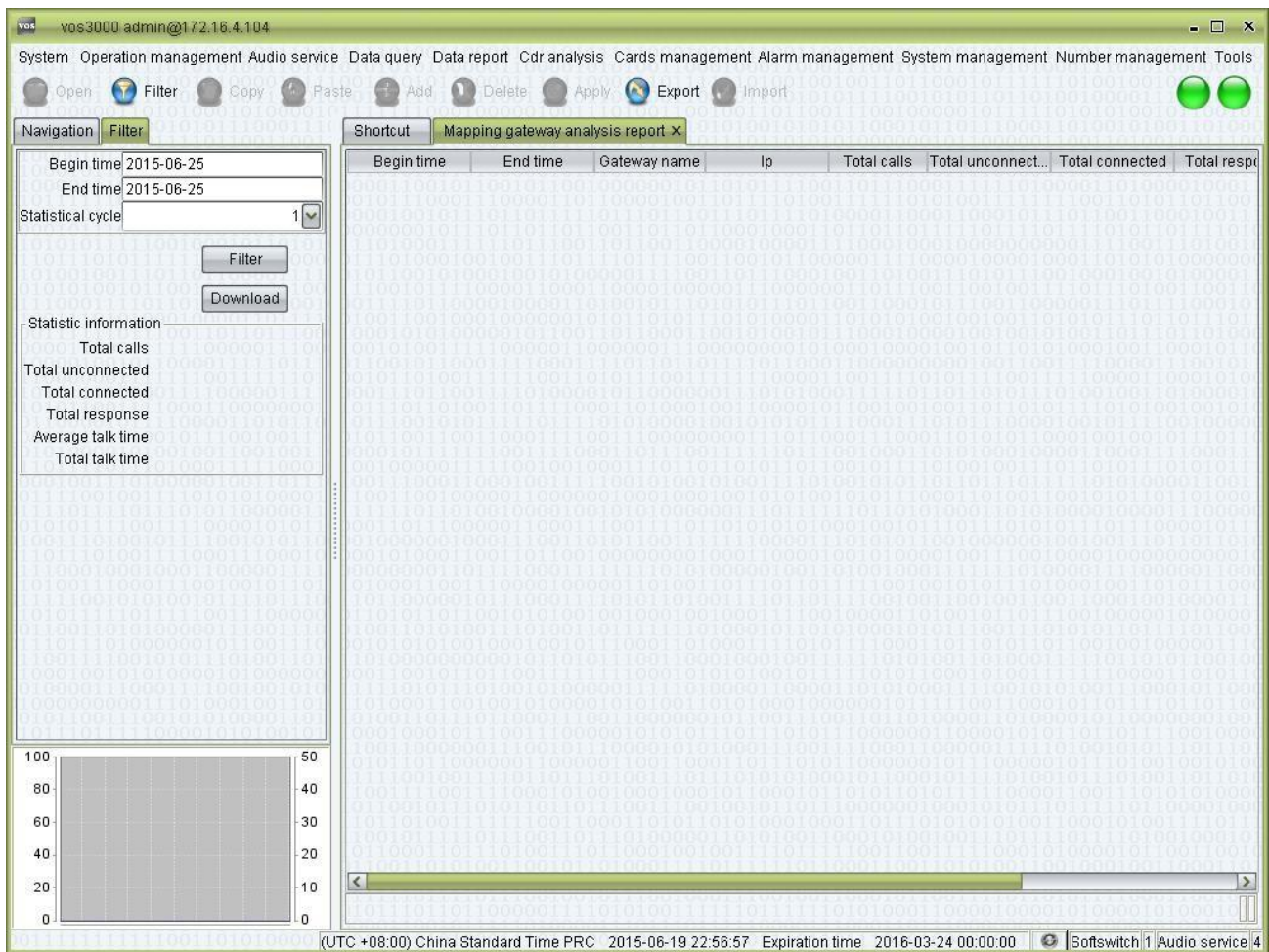
Total response: calls with connect signaling.

Average talk time: average call duration.

Total talk time: total duration.

2.8.4.1 Mapping Gateway Analysis Report

This function is used to analysis mapping gateway.

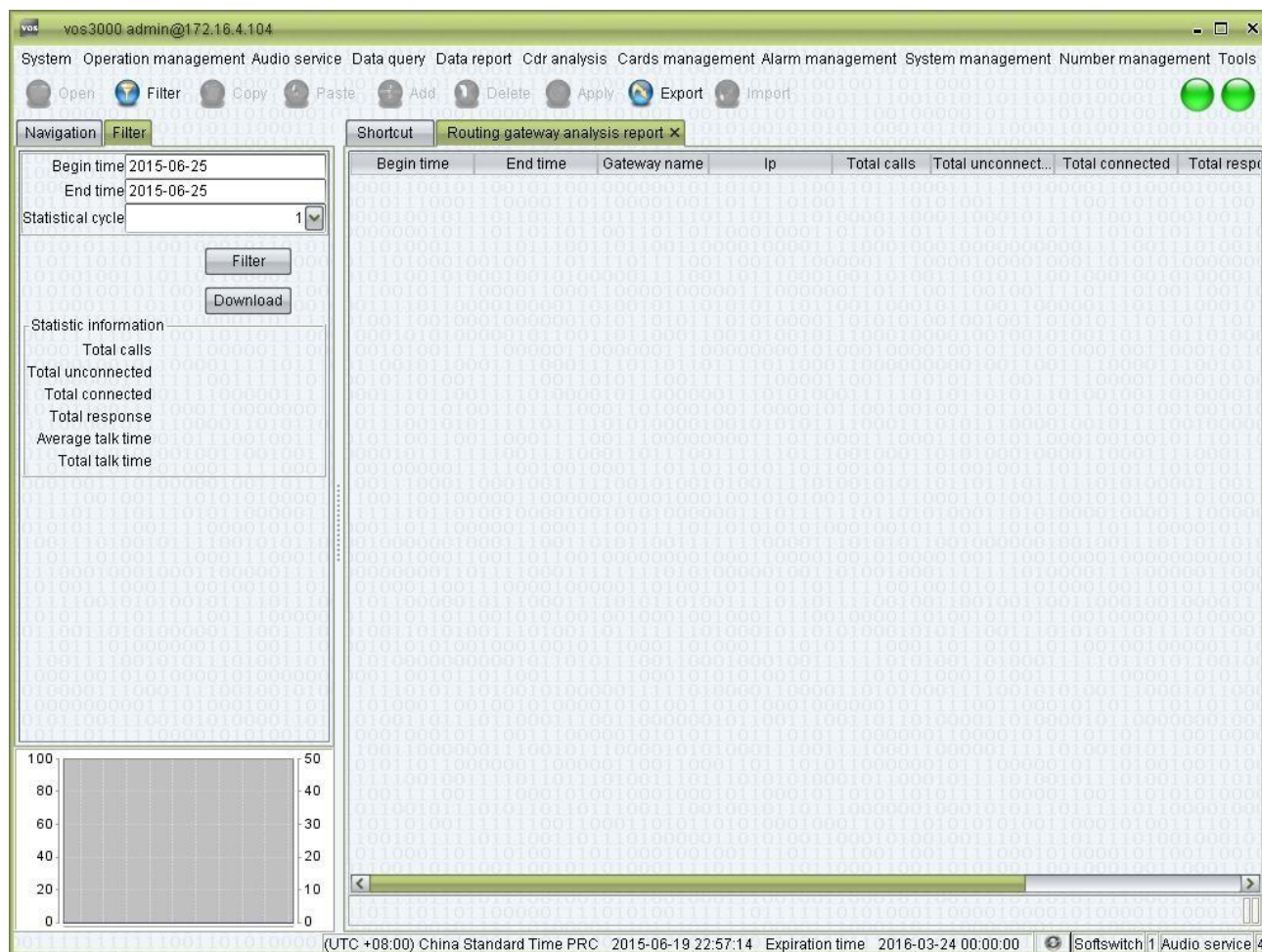


How to Start

Double-click "Navigation > Data report > Analysis report > Mapping gateway analysis report"

2.8.4.2 Routing Gateway Analysis Report

This function is used to analysis routing gateway.

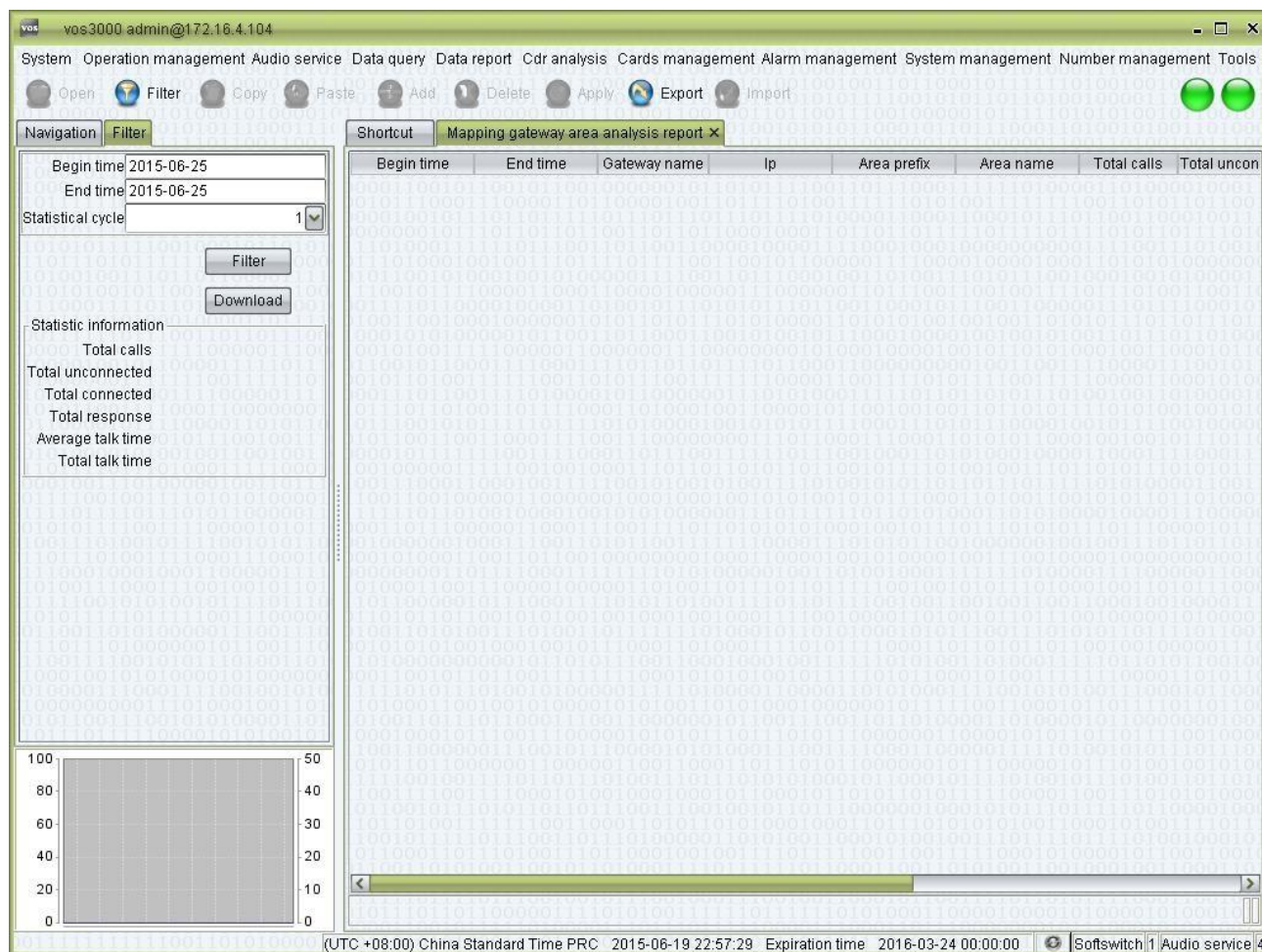


How to Start

Double-click “Navigation > Data report > Analysis report > Routing gateway analysis report”

2.8.4.3 Mapping Gateway Location Analysis Report

This function is used to analysis mapping gateway location.

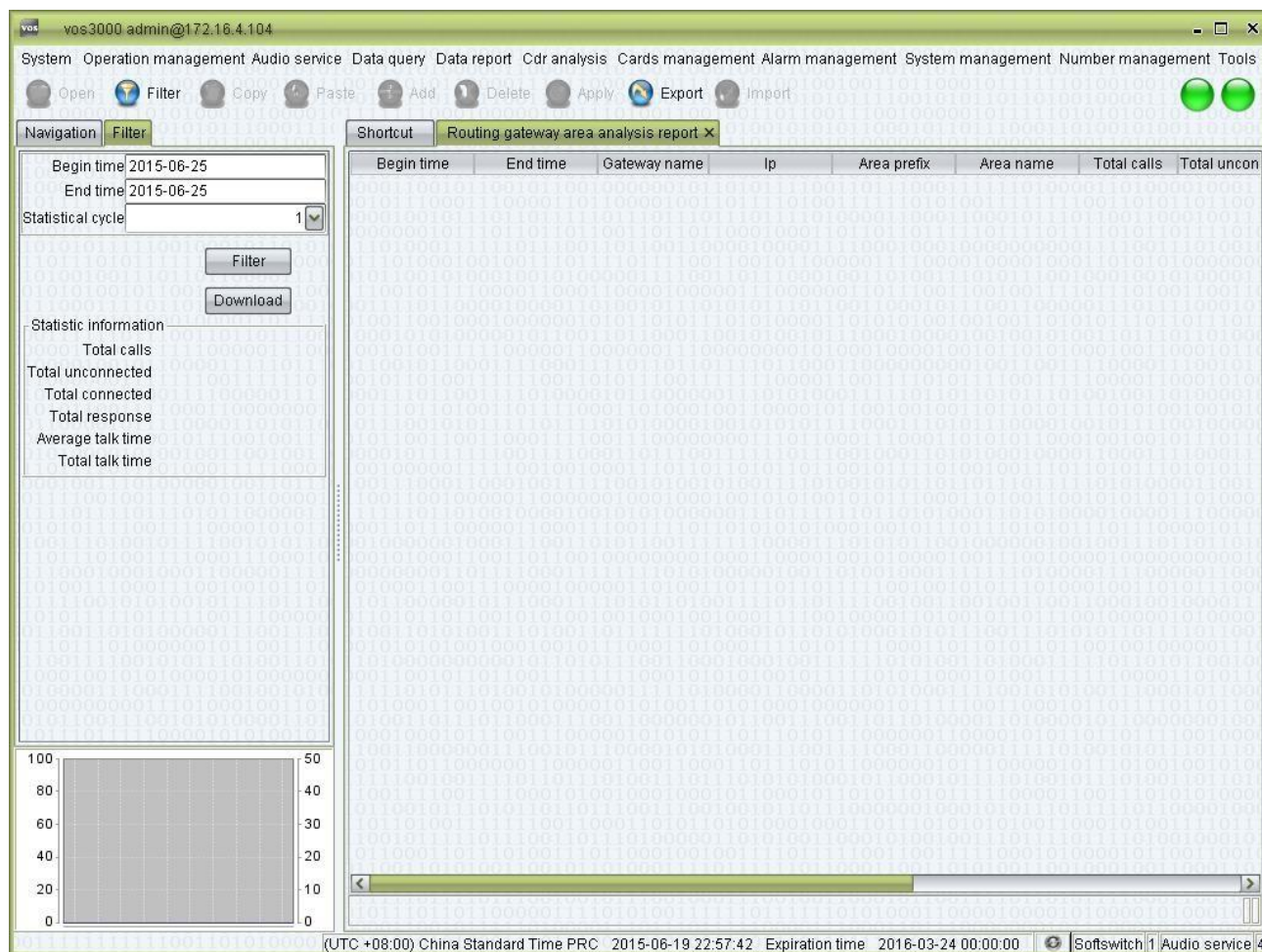


How to Start

Double-click “Navigation > Data report > Analysis report > Mapping gateway location analysis report”

2.8.4.4 Routing Gateway Location Analysis Report

This function is used to analysis routing gateway location.

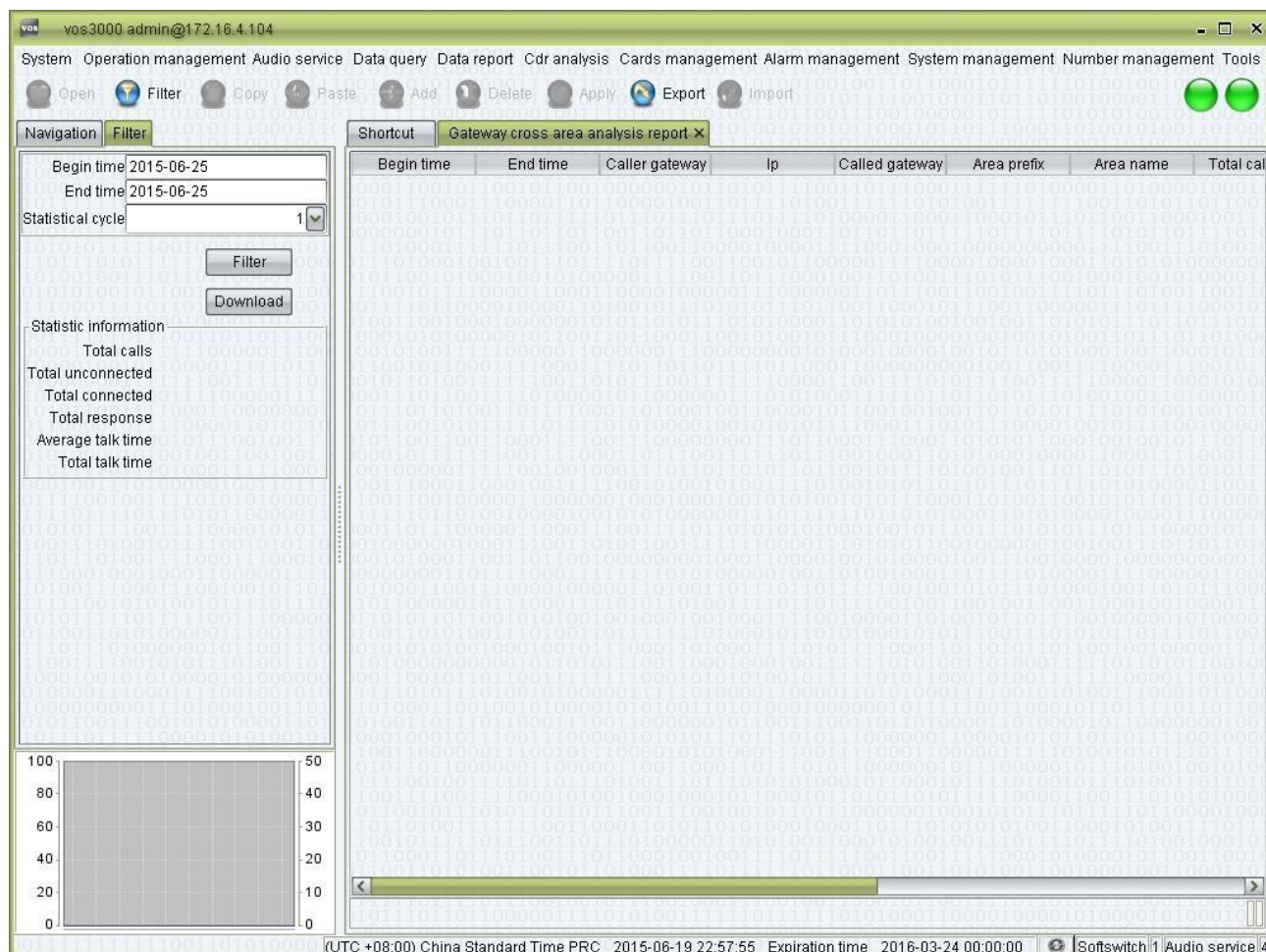


How to Start

Double-click “Navigation > Data report > Analysis report > Routing gateway location analysis report”

2.8.4.5 Gateway Cross Location Analysis Report

This function is used to analysis mapping gateway cross routing gateway.

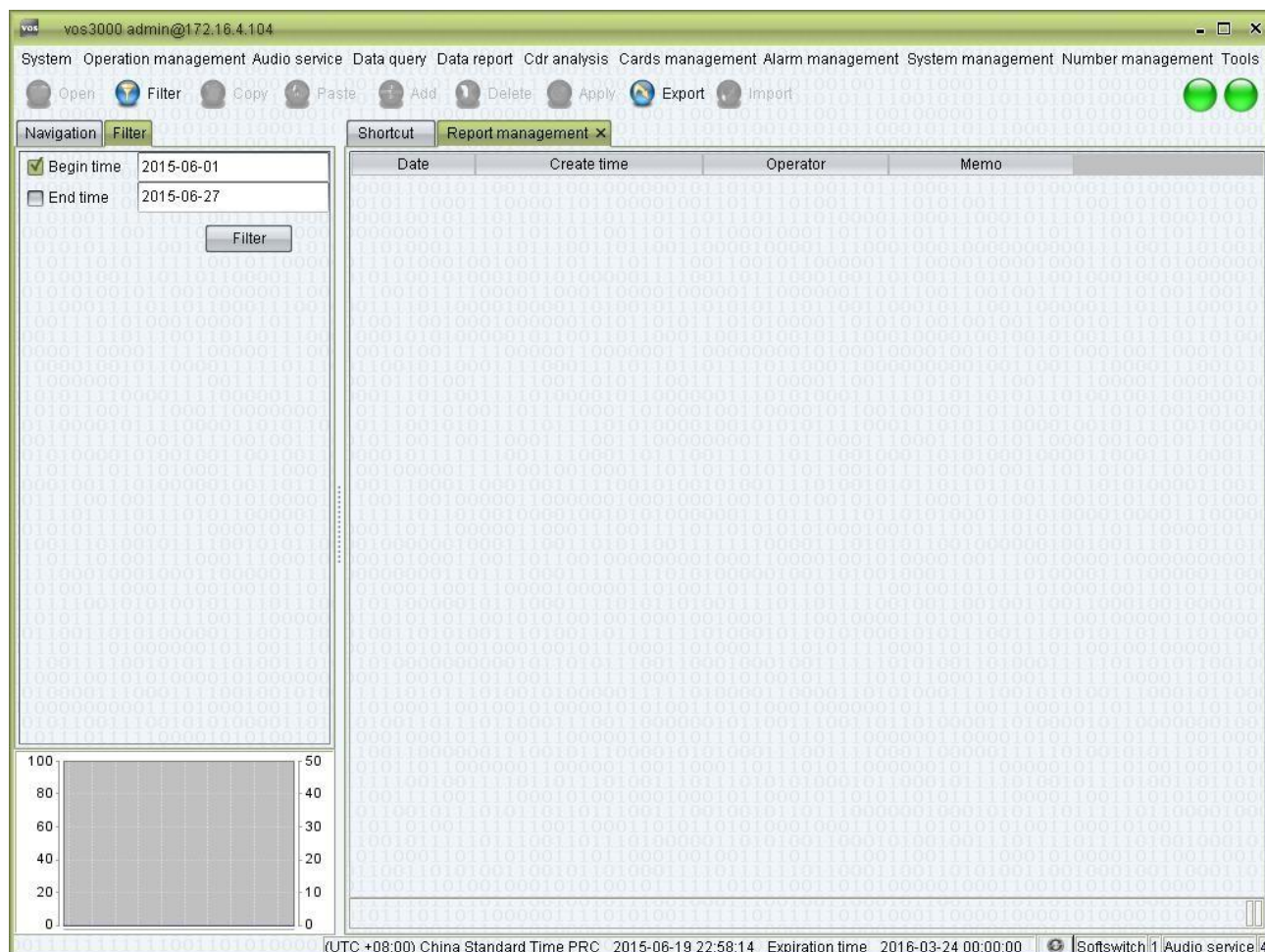


How to Start

Double-click "Navigation > Data report > Analysis report > Gateway cross location analysis report"

2.8.5 Report Management

This function is used to manage reports.



How to Start

Double-click "Navigation > Data report > Report management"

Table Items

Date: date of records in the report.

Create time: the date when the report is generated.

Operator: the operator that generated the report.

Memo: items included in the report.

Right-Click Menu

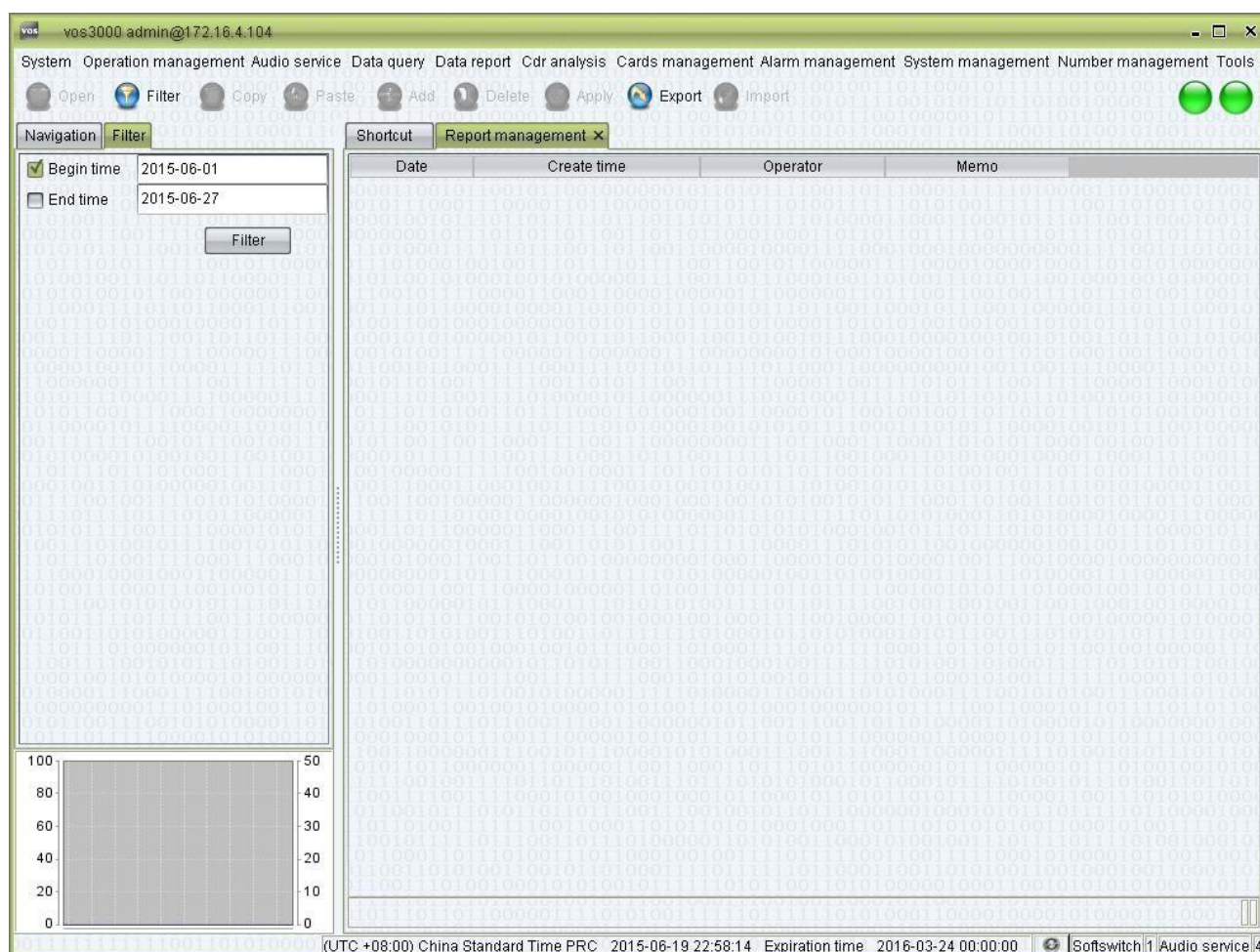
Generate all reports

Generate special report

Downloaded from VOS3000.Com

2.8.6 Other Income Report

This function is used to query income from suite rent, phone month rent, phone under consumption, suite under consumption.



How to Start

Double-click "Navigation > Data report > Other income report"

Table Items

Account id
Account name
Consumption
Account balance
Time: deduction time

Type: suite rent, phone month rent, phone under consumption, suite under consumption

Consumer Source

Memo

Agent id

Agent name

Serial number

Right-Click Menu

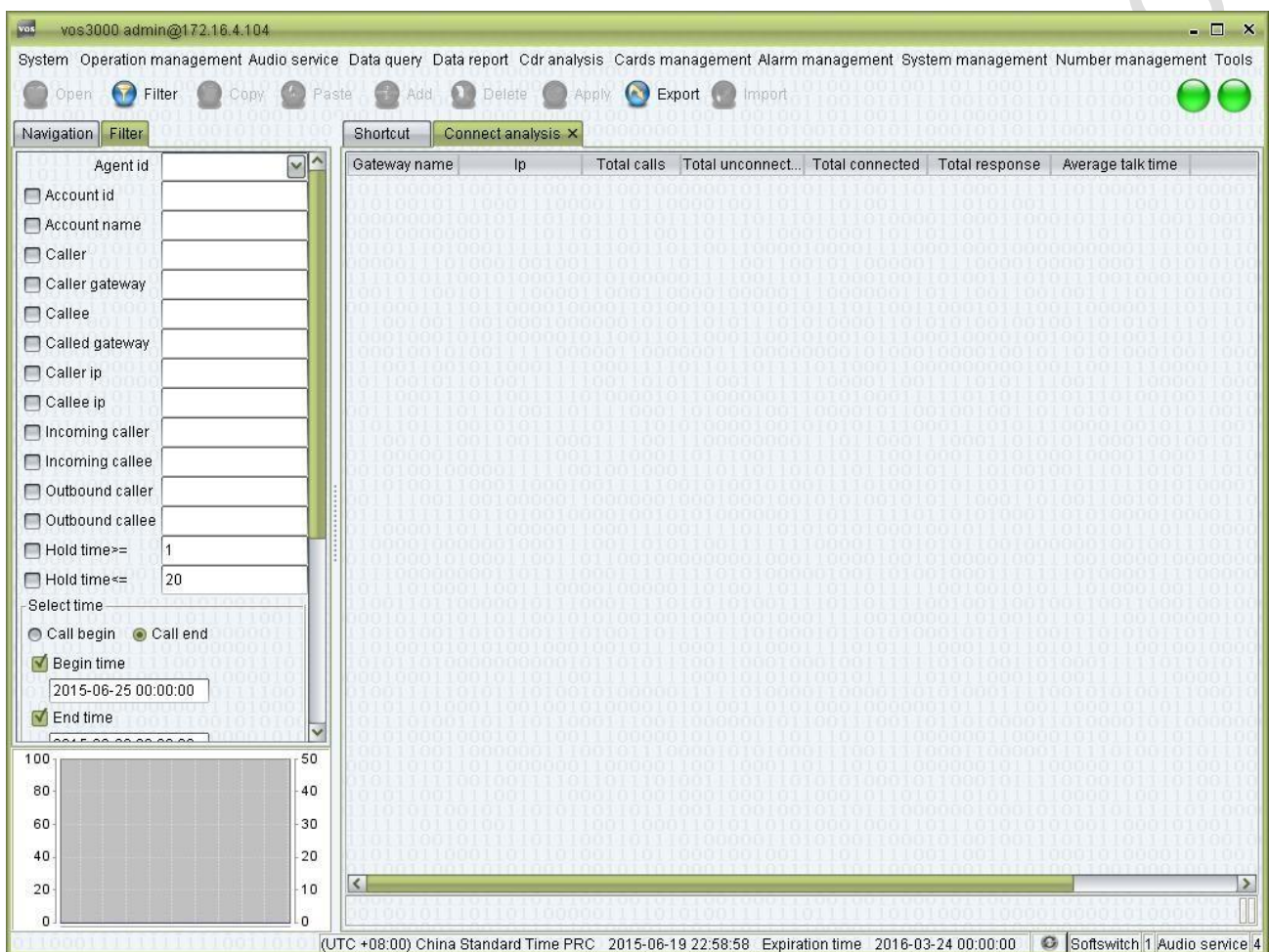
Total up

2.9 CDR Analysis

See the descriptions in “Analysis report”.

2.9.1 Connect Analysis

This function is used to analysis gateway connect.

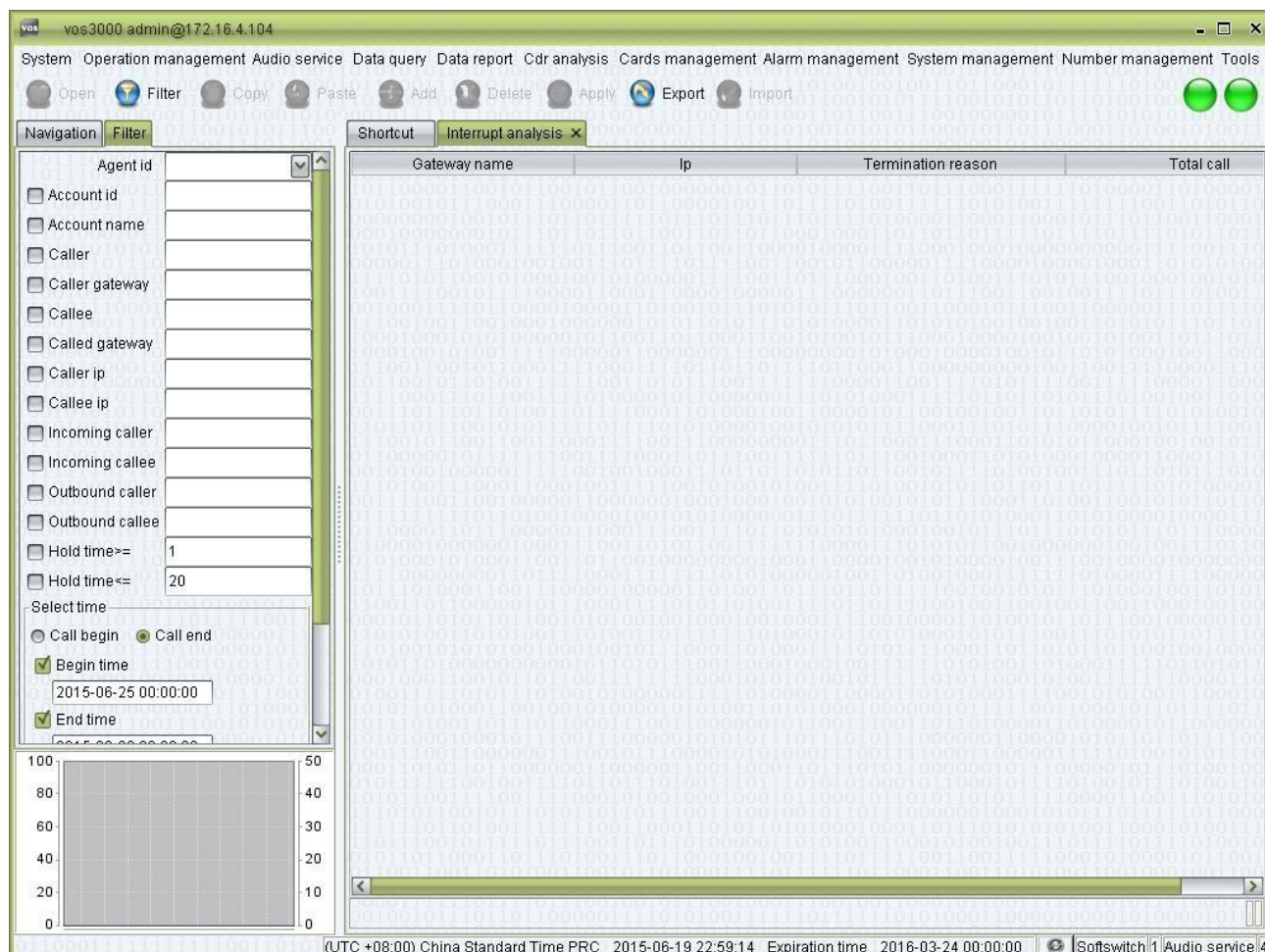


How to Start

Double-click “Navigation > CDR analysis > Connect analysis”

2.9.2 Interrupt Analysis

This function is used to analysis gateway interrupt.

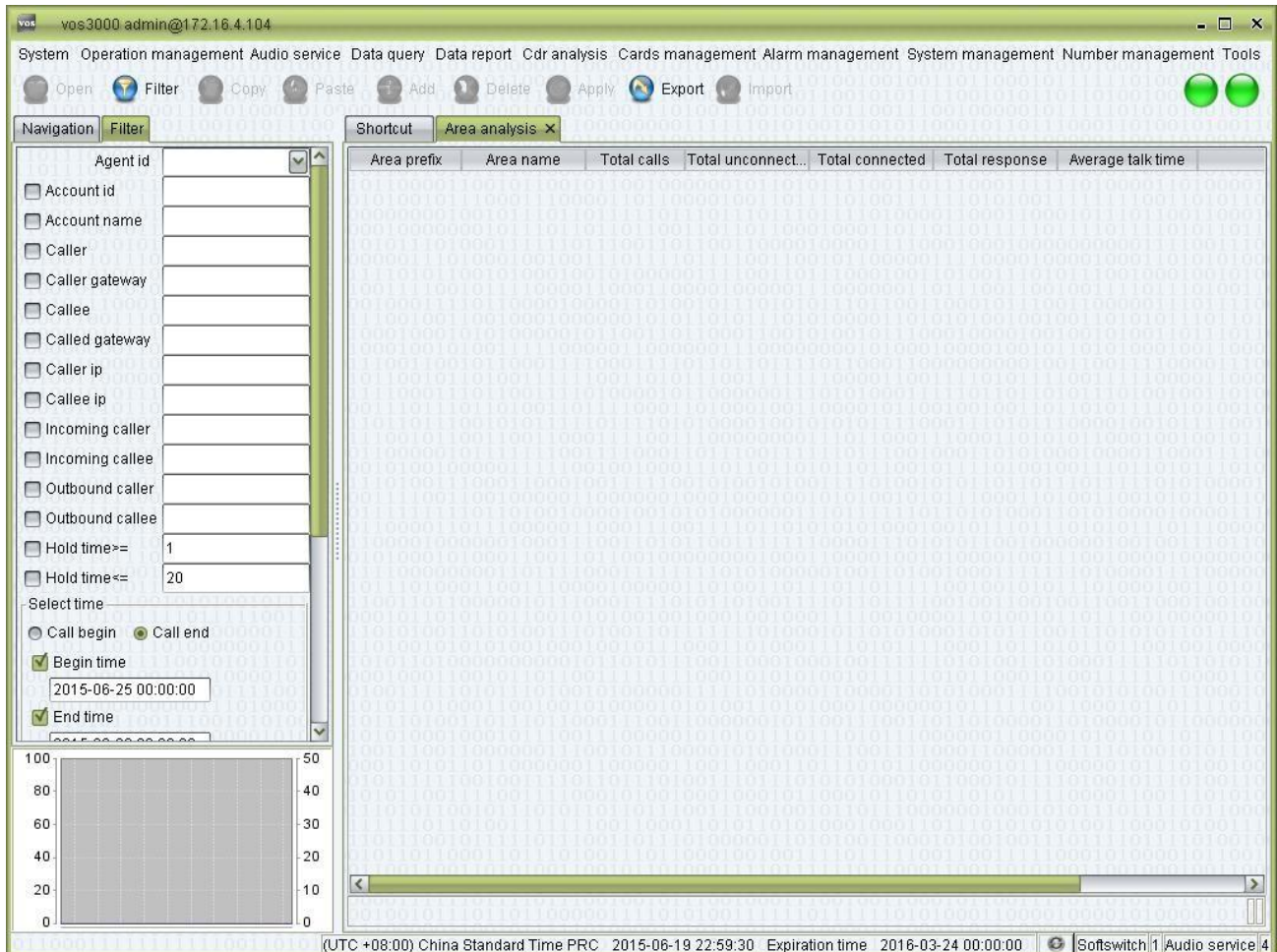


How to Start

Double-click "Navigation > CDR analysis > Interruption analysis"

2.9.3 Area Analysis

This function is used to analysis gateway area.

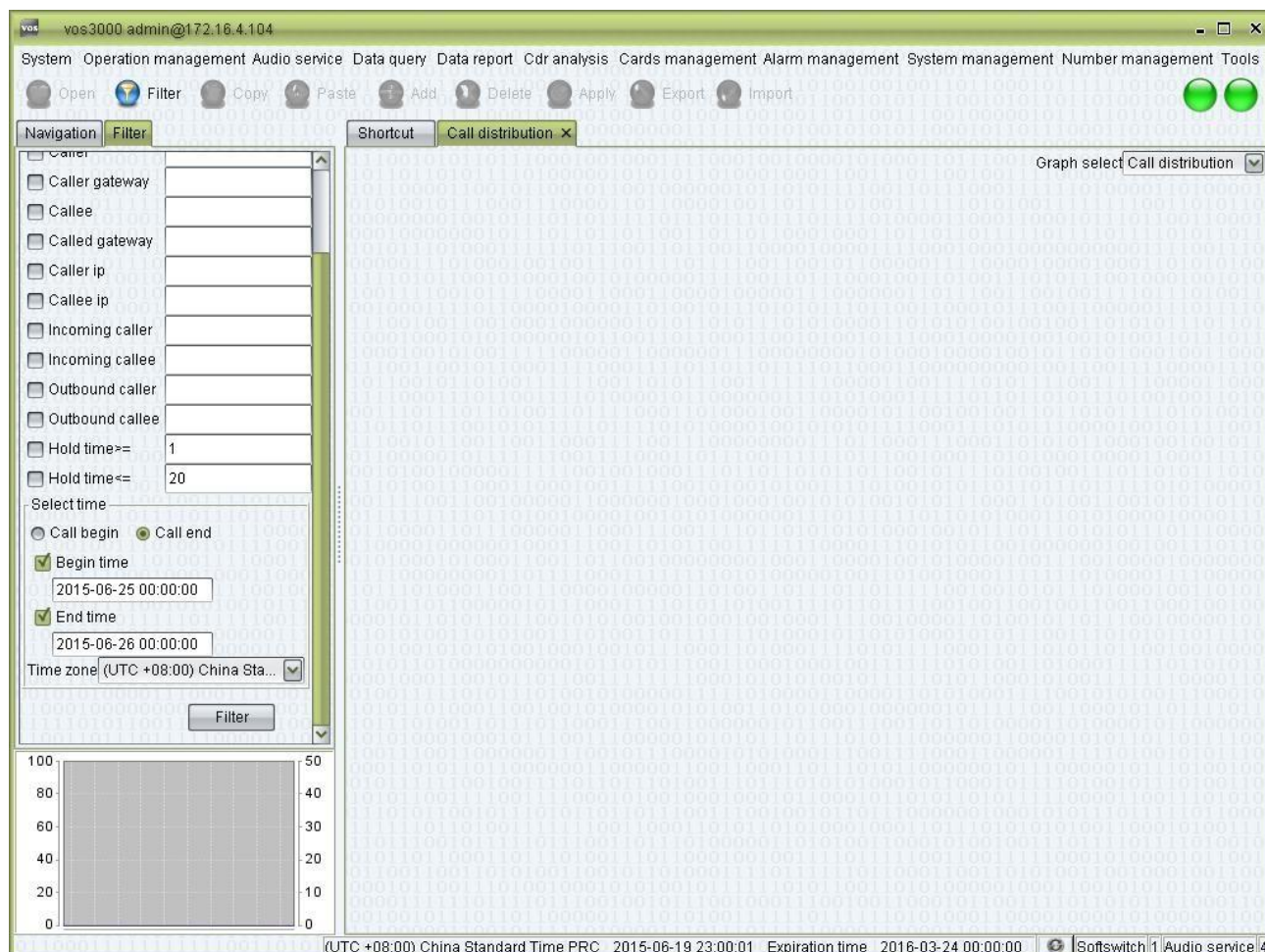


How to Start

Double-click "Navigation > CDR analysis > Area analysis"

2.9.4 Call Distribution

This function is used to show call distribution of 24 hours.



How to Start

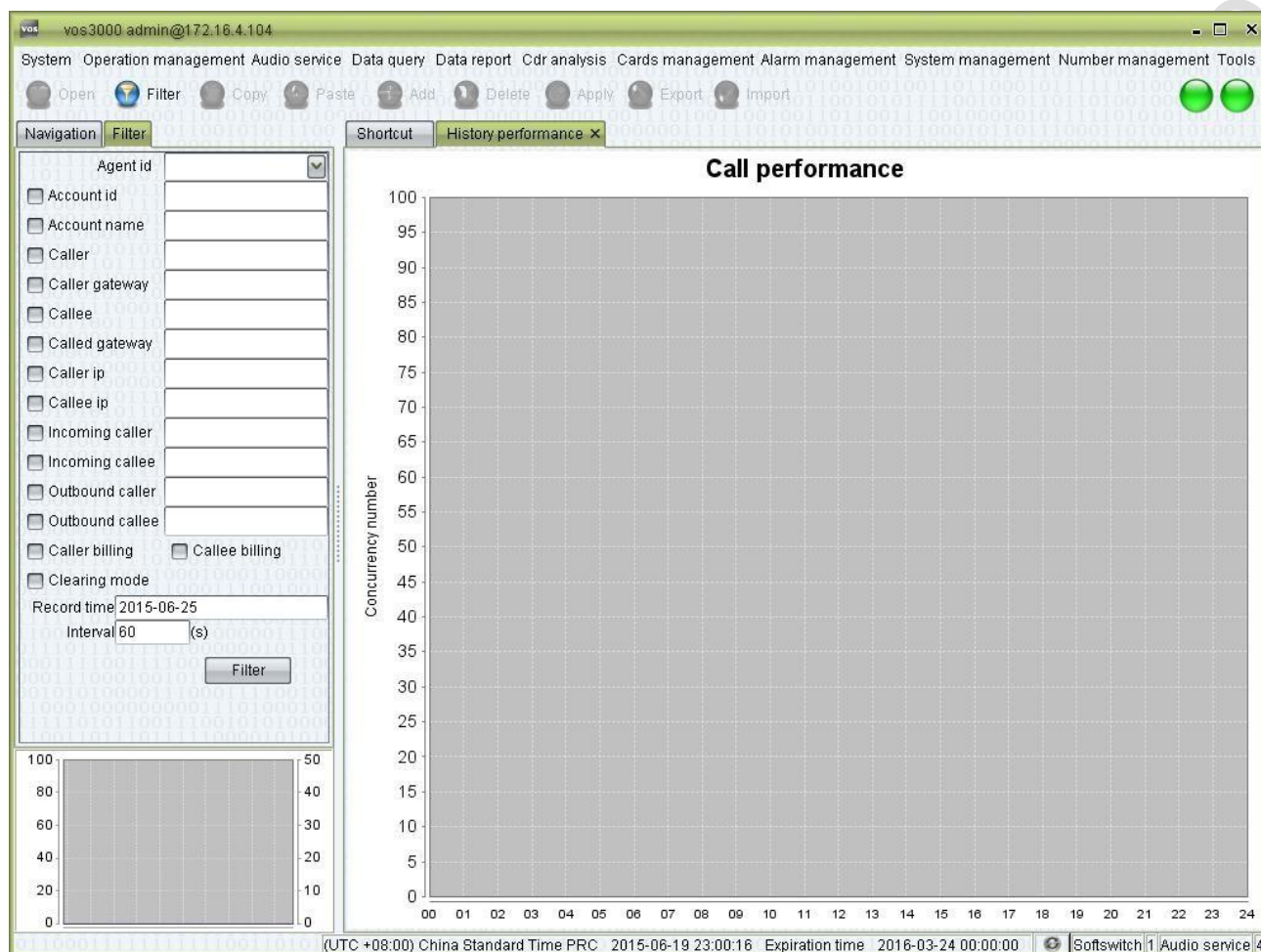
Double-click "Navigation > CDR analysis > Call distribution"

2.9.5 Historical Performance

This function is used to show the concurrent calls on specified date.

NOTE

Unsuccessful calls are not counted here, so the number shown in the chart will be slight lower than that in reality.



How to Start

Double-click "Navigation > CDR analysis > Historical performance"

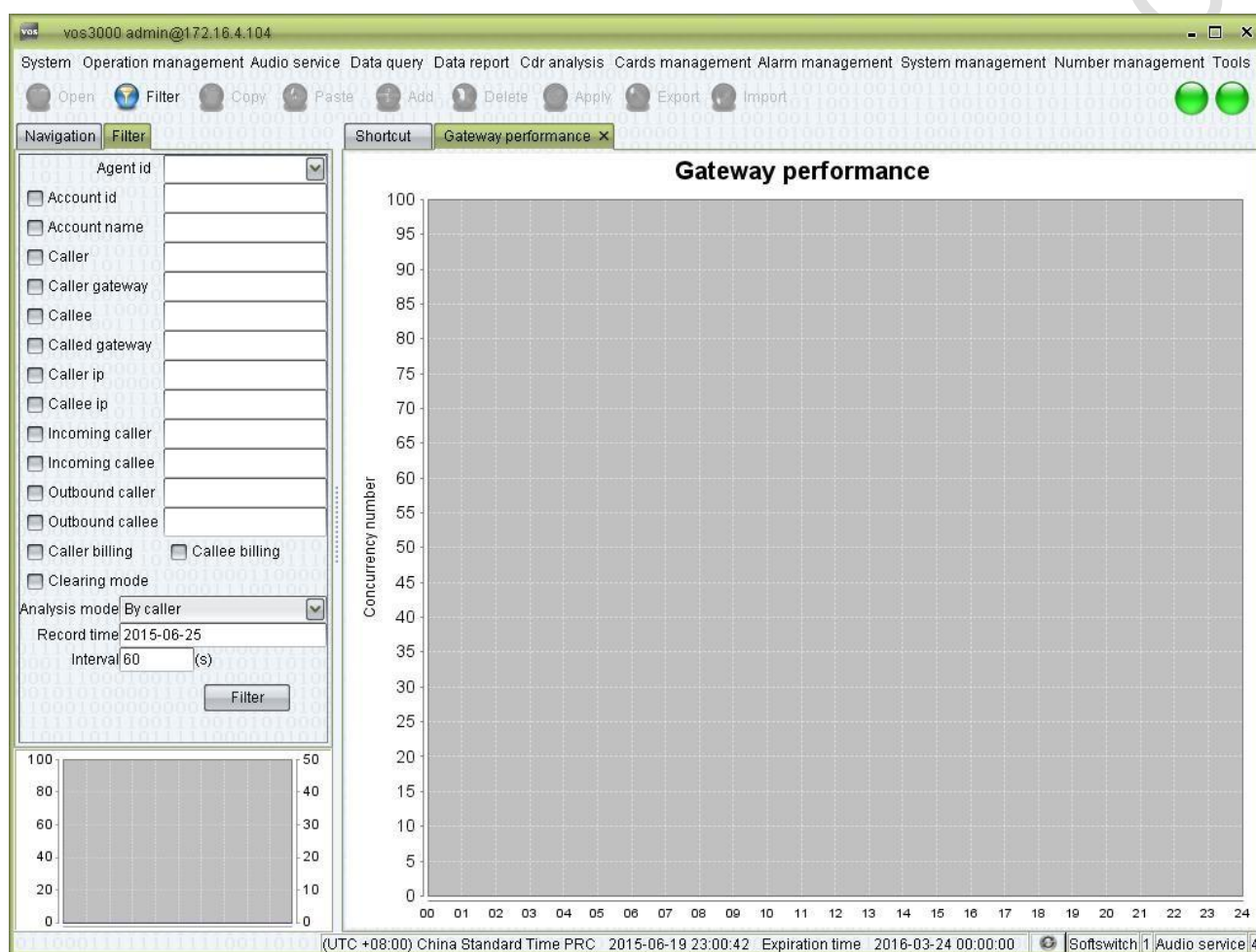
2.9.6 Gateway Performance

This function is used to show the concurrent calls of gateways on specified date.

NOTE

Unsuccessful calls are not counted here, so the number shown in the chart will be slight lower than that in reality.

The number of gateway can be set by “System management > System parameter > SERVER_DISPLAY_CHART_GATEWAY_SIZE”.

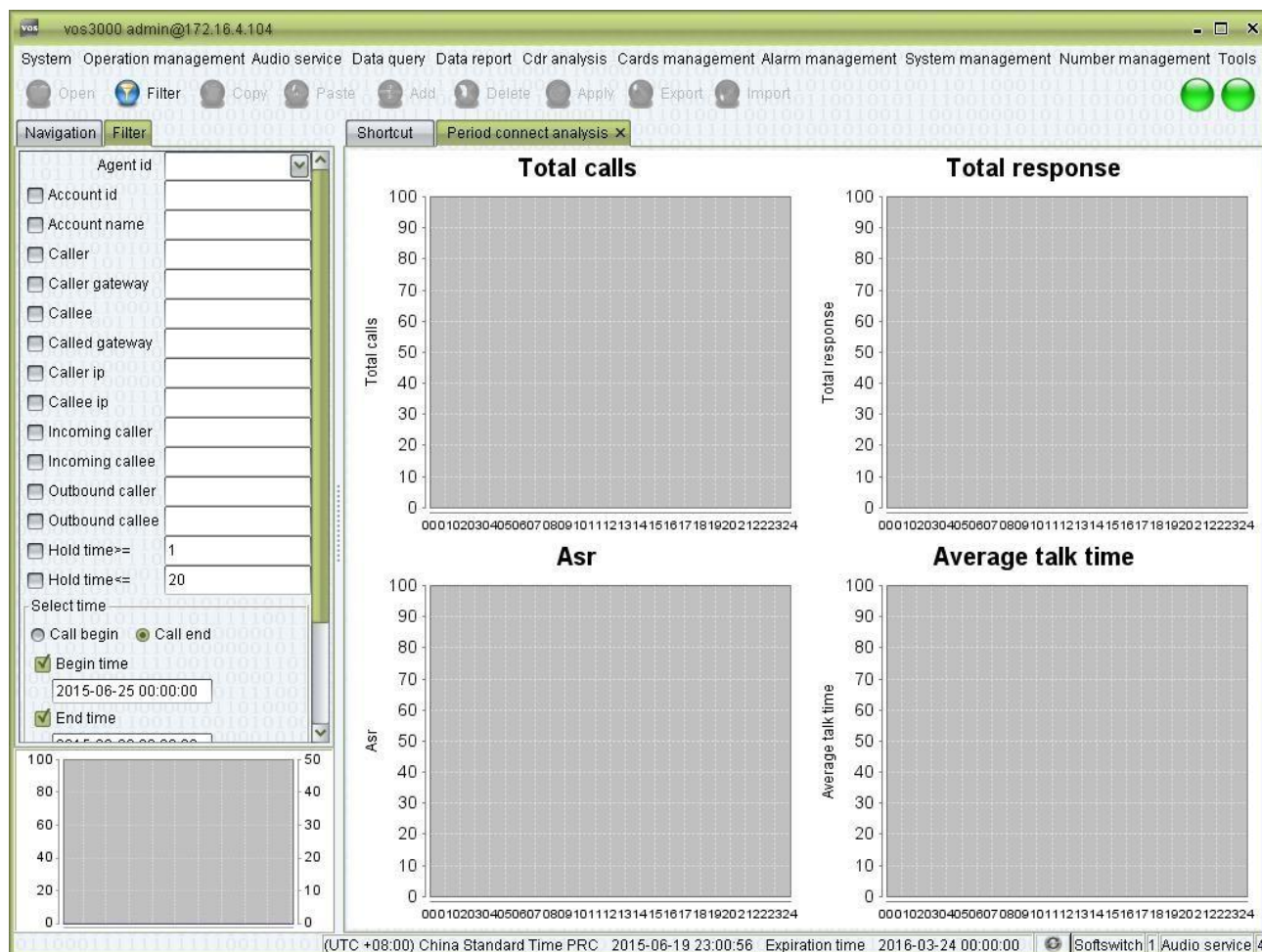


How to Start

Double-click “Navigation > CDR analysis > Gateway performance”

2.9.7 Period Connect Analysis

This function is used to analysis total calls/total response/ASR/ACD.



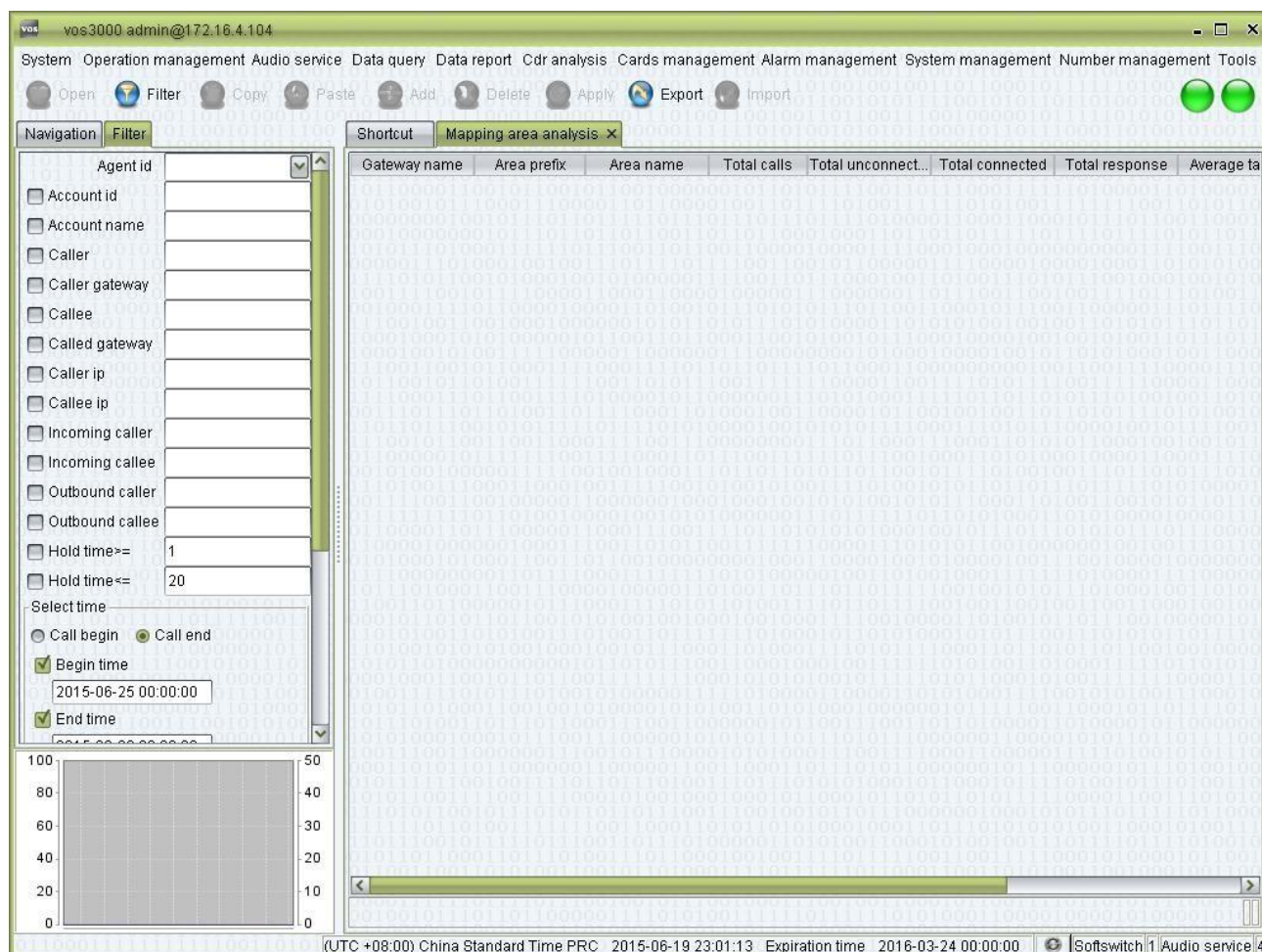
How to Start

Double-click "Navigation > CDR analysis > Period connect analysis"

2.9.8 Gateway Area Analysis

2.9.8.1 Mapping Area Analysis

This function is used to analysis mapping area.

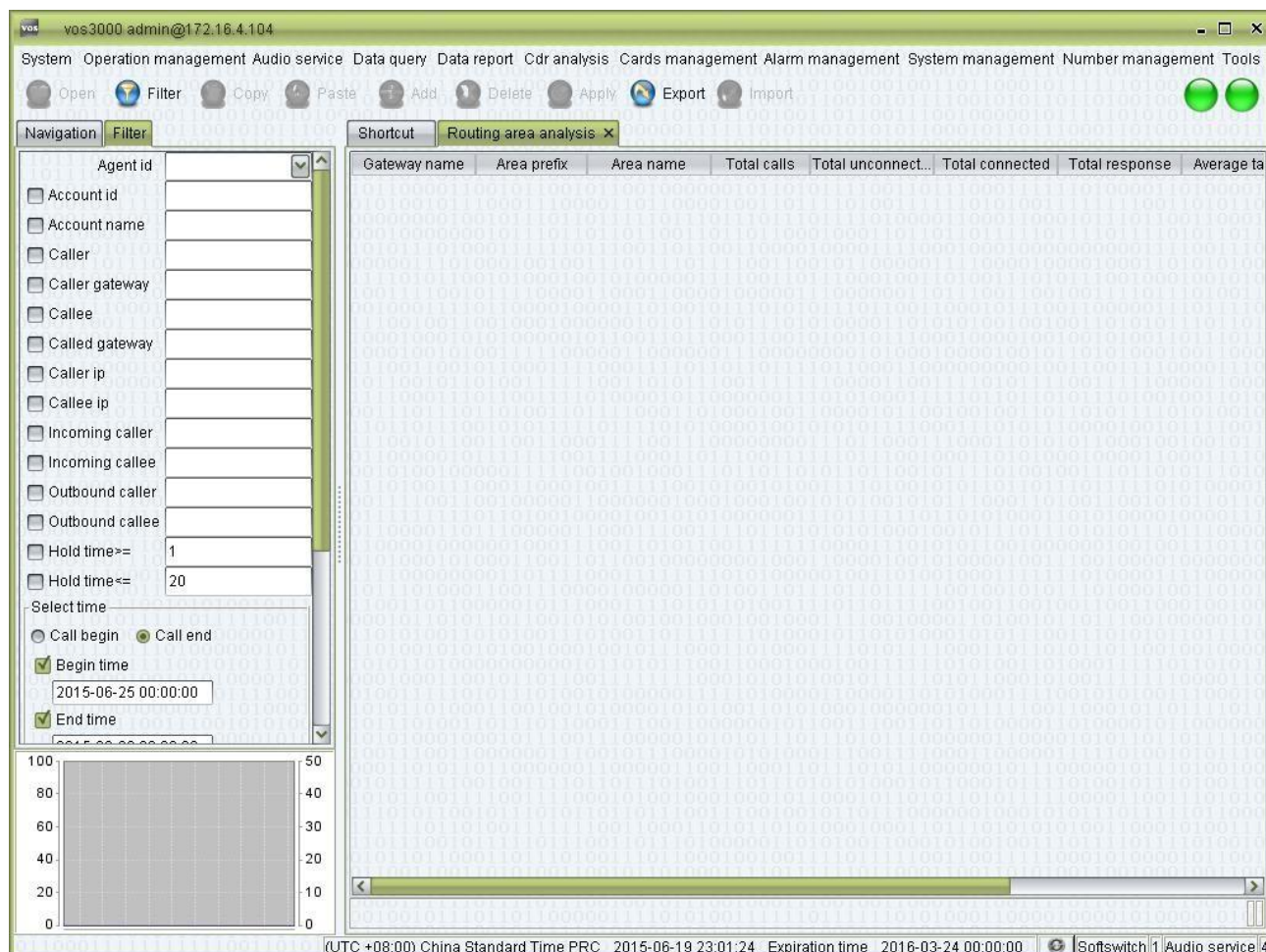


How to Start

Double-click "Navigation > CDR analysis > Gateway area analysis > Mapping area analysis"

2.9.8.2 Routing Area Analysis

This function is used to analysis routing area.

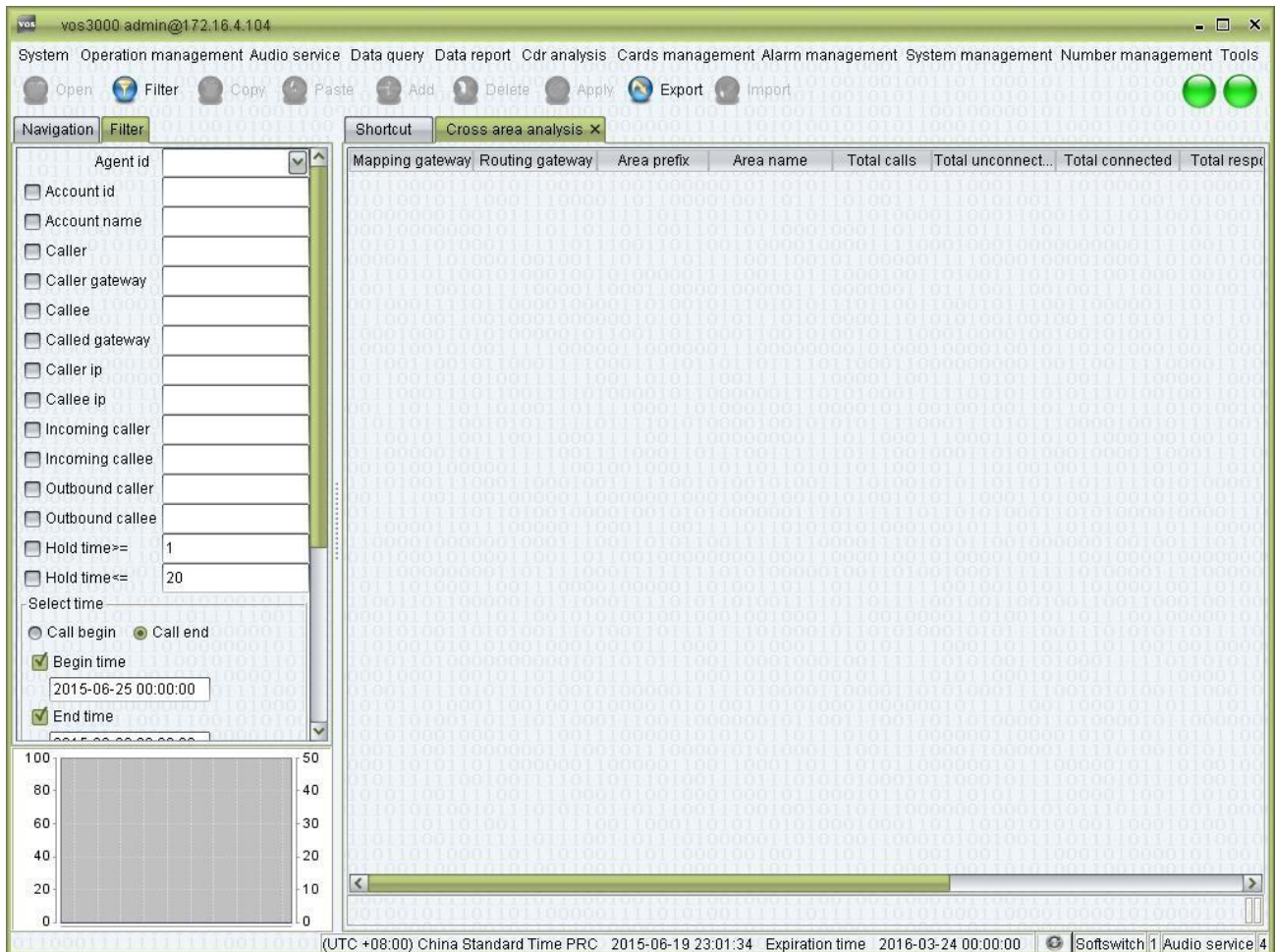


How to Start

Double-click “Navigation > CDR analysis > Gateway area analysis > Routing area analysis”

2.9.8.3 Cross Area Analysis

This function is used to analysis cross area.



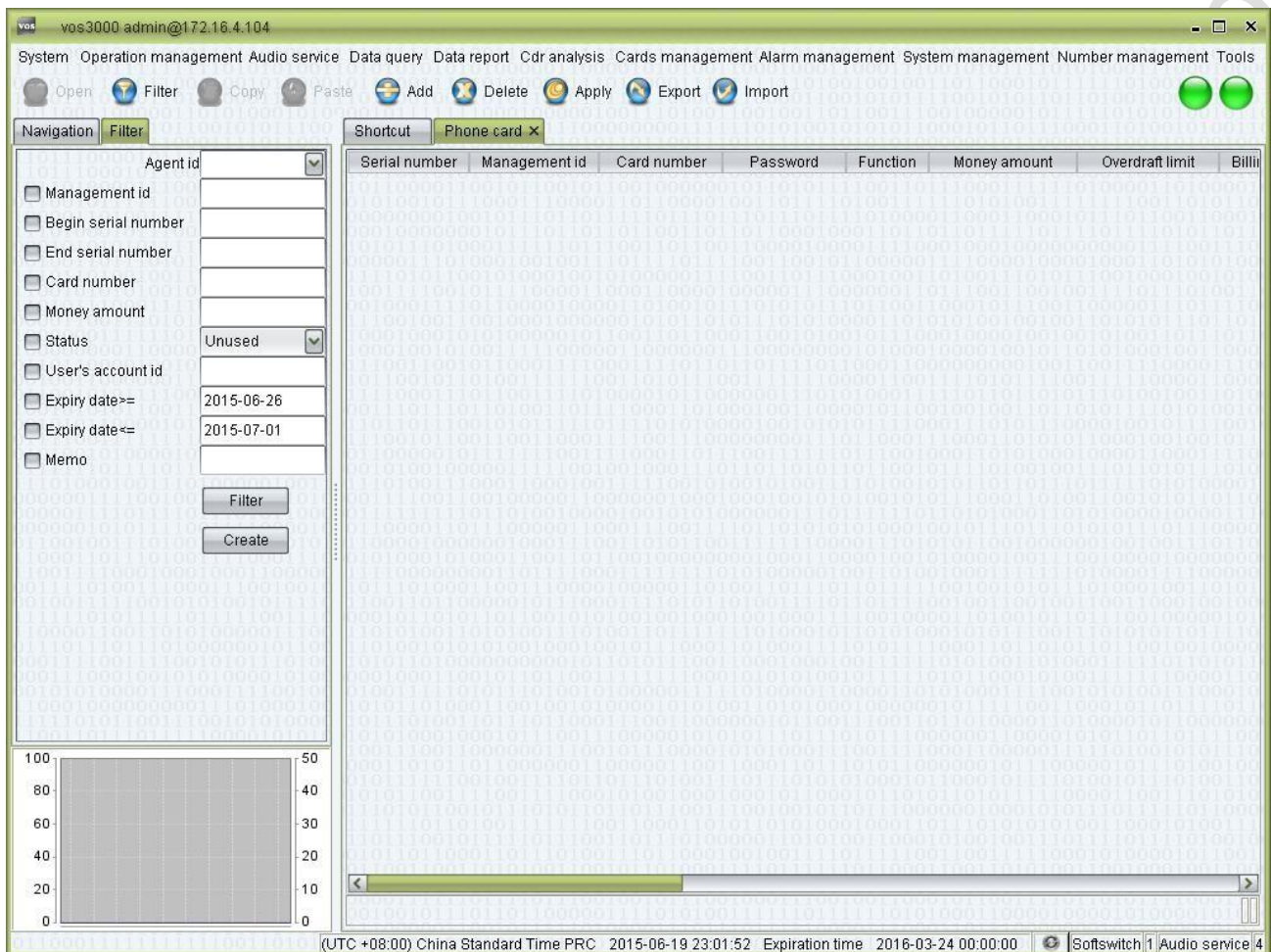
How to Start

Double-click “Navigation > CDR analysis > Gateway area analysis > Cross area analysis”

2.10 Cards Management

2.10.1 Phone Card

This function is used to manage phone card.



How to Start

Double-click "Navigation > Card management > Phone card"

Table Items

Serial number: the sequence number of activated cards. The initial number can be specified by users.

Management id

Card number: the unique id of a phone card. Card numbers of existing cards cannot be modified.

Password: the password used for authentication in recharge.

Function:



Create account: this card can be used to establish an account.

Active phone card: when using this card create account, to do an account and active phone card.

Platform phone: when using this card create account, to do platform phone, also can set default authorization type.

Recharge: this card can be used to recharge.

Replace account billing rate group: when using this card recharge, the current account billing rate is replaced by the set rate card.

Package joined account package order: when using this card recharge, the account package will add the set package card.

Money charged into the current account balance: when using this card create account or recharge, the current balance of this account can increase the value of card.

Money amount: the amount to be recharged.

Overdraft limit

NOTE

Negative is supported.

Billing rate: the rate used for phone card.

Package name: the name of the package used for phone card.

NOTE

Multi packages are supported.

Agent id: the agent account specified in account binding for phone card.

Lock type: "No Lock" or "Locked"

Sold

Bind number limit

Display caller id: the caller ID shown at the called end.

Produce time: the date when the card is created.

Expire time: the date of expiration.

Active(Day): the number of days when enabled before expiration time.

Enable date

Memo: descriptions of the card.

Type

- Phone card
- Internet phone card

User's account id: when used, the number of the account being recharged.

User's account name: when used, the name of the account being recharged.

Batch create phone cards

Settings

Function

- ☒ Create account ☒ Active phone card
- ☐ Platform phone Authorization type: International
- ☒ Recharge ☐ Replace account billing rate group
- ☐ Package joined account package order
- ☒ Money charged into the current account balance

Normal

Number of cards: Money amount:

Password mode: On Password length:

Begin card number:

Others

Begin serial number: Management id:

Card number prefix: Overdraft limit:

Billing rate: Package name:

Active(days): 0 Expire time: 2016-04-14

Bind number limit: 1 Lock type: No lock

Agent id:

Memo:

Ok Cancel

Number of cards: the number of cards to be created.

Money amount: the amount to be recharged.

Password mode: whether the cards being created have passwords.

Password length: the length of the passwords. Passwords are generated automatically by the system.

Begin card number: the initial card number of the cards.

Begin serial number: the initial sequence number of the cards. The last sequence number will be automatically determined according to the number of card to be created.

**NOTE**

If the initial sequence number is left blank, it will be set to the number that is one plus the largest sequence number activated up to now.

Management id:

Card number prefix: the prefix number of the cards being created.

Overdraft limit:

Billing rate:

Package name:

Active(days): how many days will account's expiry date be extended after charging.

Expire time:

Bind number limit

Lock type

Agent id: the agent account specified in account binding for phone cards.

**NOTE**

This card can only be used for the agent's sub-account when setting the agent account.

Memo

Related Instructions

Phone card as account cards case.

The active day of this phone card is 0 days, when using this card create account ,the account is valid for phone card expire time.

The active day of this phone card is N days,when using this card create account , the account is valid for the current time extended $N*24$ hours.

Phone card as recharge cards case.

If the extened mode was specified as "Superposition" in "System management > System parameter >

SERVER_PAY_PHONE_CARD_CUSTOMER_EXPIRE_DAY_MODE" , If phone card's active (days) is N, the account's valid time will be extended $N*24$ hours after recharge, from the current time.

If the extended mode was specified as "Standard" in "System management > System parameter >

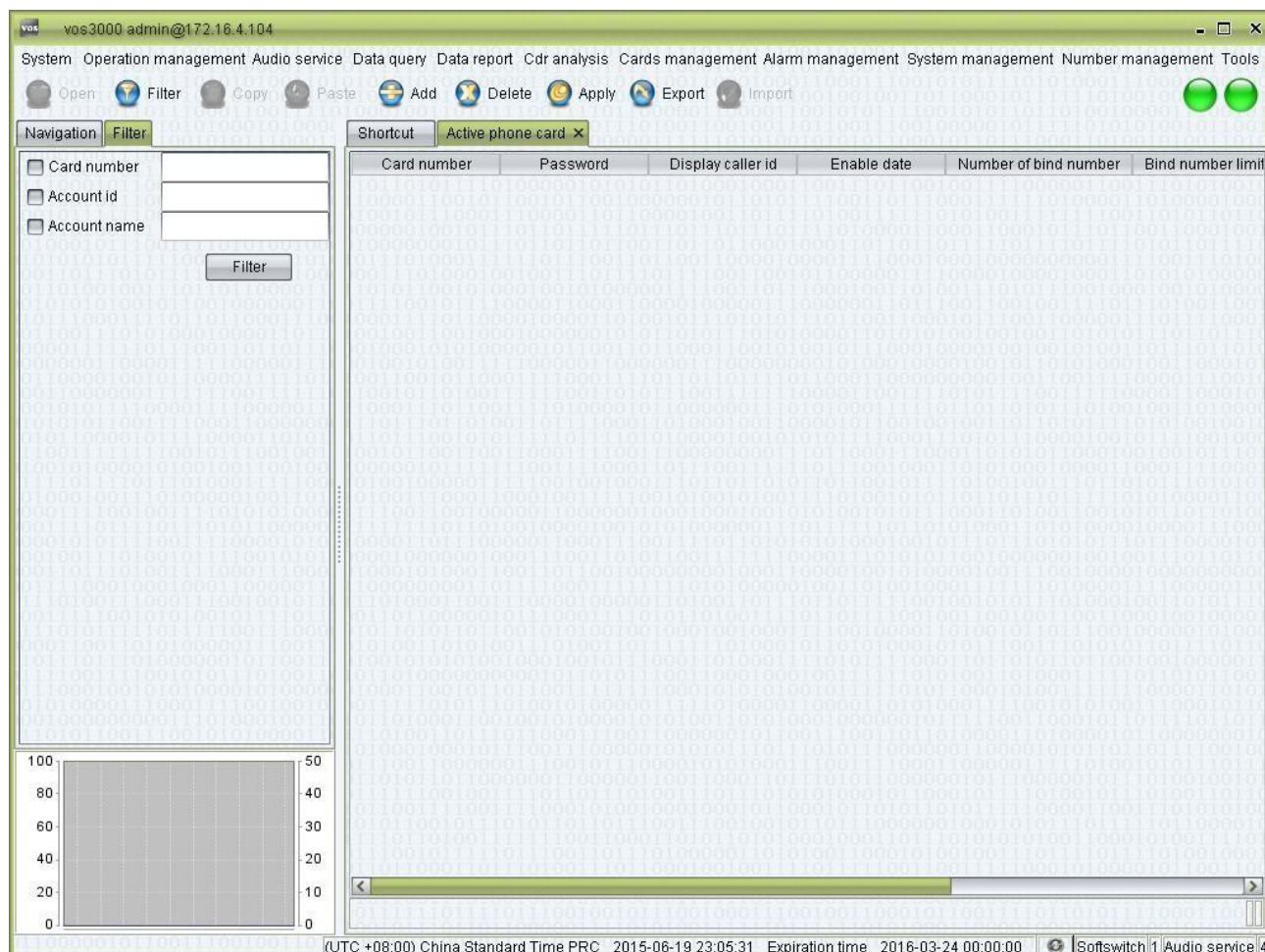
SERVER_PAY_PHONE_CARD_CUSTOMER_EXPIRE_DAY_MODE" .

If phone card's active (days) is 0, if the expiration date extension is specified in "System management > System parameter > SERVER_PAY_DELAY_CUSTOMER_EXPIRE_DAY", the expiration date of an account will be re-calculated accordingly after recharge.

If phone card's active (days) is N, the account's valid time will be extended $N \times 24$ hours, from the current time. then the expiration date of an account will be re-calculated accordingly compared to original expired time of account.

2.10.2 Active Phone Card

This function is used to manage active phone card.



How to Start

Double-click "Navigation > Card management > Active phone card"

Table Items

Card number
Password
Display caller id
Enable date
Number of bind number
Bind number limit

Account id

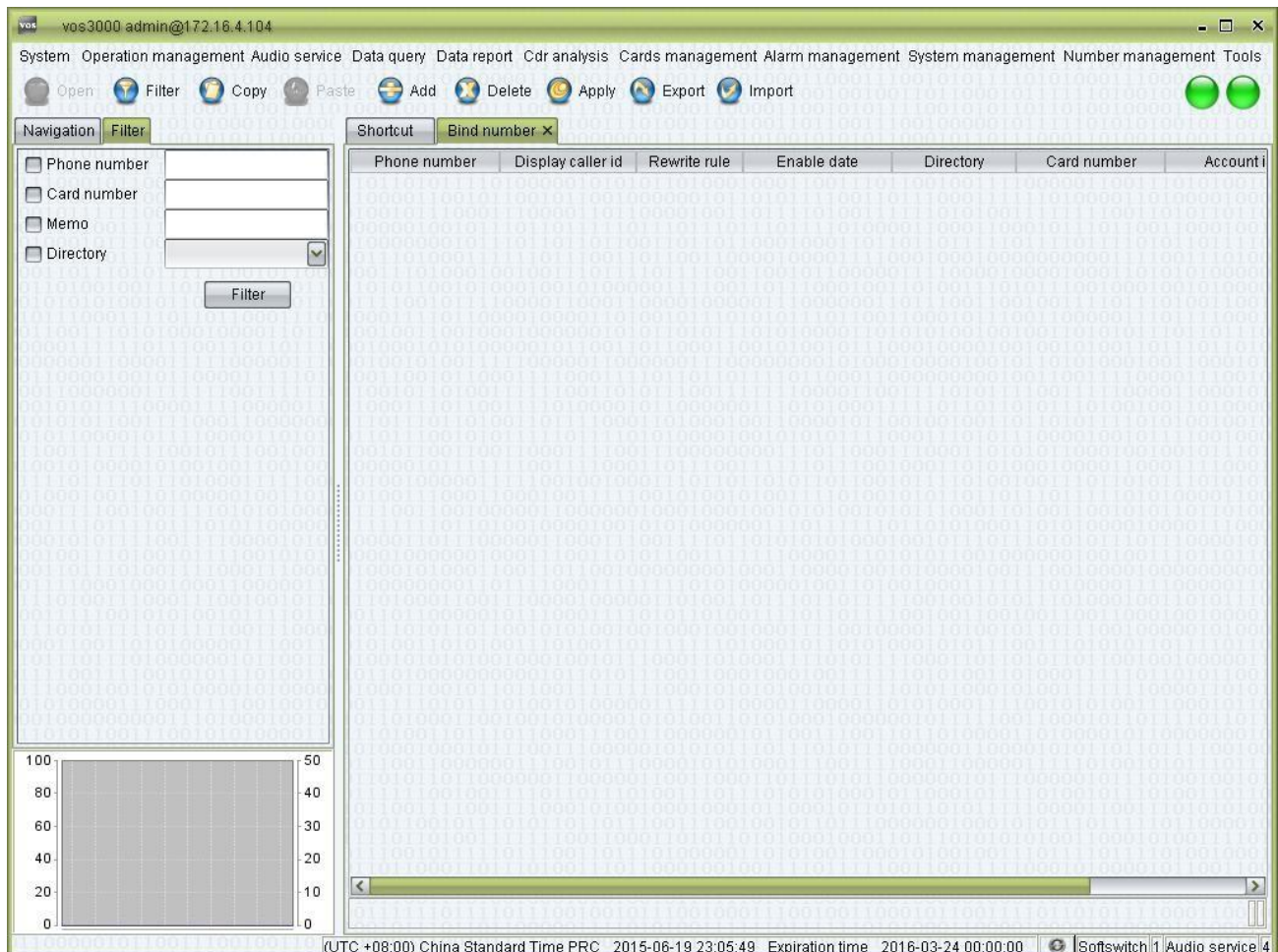
Account name

Memo

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2.10.3 Bind Number

This function is used to manage bind number.



How to Start

Double-click "Navigation > Card management > Bind number"

Table Items

Phone number

Display caller id: can be set according to callee.

Rewrite rule

Enable date

Language directory

Card number

Account id

Memo

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2.11 Alarm Management

Alarm level: General/Minor/Major/Critical.

Upper: upper bound which trigger the alarm.

Lower: lower bound which trigger the alarm.

Period: detection period.

Email alarm: can be set by "System management > System parameter > SERVER_ALARM_ENABLE_EMAIL".

Email: can be set by "System management > System parameter > SERVER_ALARM_EMAIL".

2.11.1 Alarm Settings

2.11.1.1 System Alarm

This function is used to monitor system.

vos3000 admin@172.16.5.166

System Operation manage Audio serv Data que Data rep Cdr analy Cards manage Alarm manage System manage Number manage Interface manage Smart check cloud man Tool

Open Filter Copy Paste Add Delete Apply Export Import

Navigation Filter

Monitor device All
Alarm type Cpu

Filter

Alarm settings
Optional settings>>

System alarm

Monitor device	Alarm type	Alarm severity	Upper	Period	Voice alarm	Voice alarm call number
All	Clock deviation	Major	300	Default	Default	
All	Conversation time	Minor	7,200	None	Default	
All	Cpu	Major	90%	Default	Default	
All	Database	Critical	None	60	Default	
All	Illegal call	Minor	600	60	Default	
All	Login mac Restricted	Major	None	Default	Default	
All	Master synchronize status	Major	None	60	Default	
All	Memory	Major	90%	Default	Default	
All	Pending cdr	Major	10,000	60	Default	
All	Slave synchronize status	Major	None	60	Default	
All	Standby enable	Major	None	Default	Default	

Selected 0 Row(s) Total: 11 Row(s)

(UTC +08:00) China Standard Time Asia/Shanghai 2017-05-24 14:50:35 Expiration time 2017-12-31 00:00:00 Softswitch 1 Audio service 3

How to Start

Double-click "Navigation > Alarm management > Alarm settings > System alarm"

Table Items

Monitor device:

Alarm Type:

Clock deviation: used to monitor the time deviation between master and slave server

Conversation time:

Cpu

Database

Illegal call

Login mac Restricted

Master synchronize status

Memory

Pending cdr

Slave synchronize status

Standby enable

2.11.1.2 Network Alarm

This function is used to monitor network.

Monitor device	Network device	Alarm type	Alarm severity	Upper	Lower	Period
SERVER_172.16.5.166	eth0	Packages transmit	Major	None	None	Default
SERVER_172.16.5.166	eth0	Bytes transmit	Major	None	None	Default
SERVER_172.16.5.166	eth0	Packages receive	Major	None	None	Default
SERVER_172.16.5.166	eth0	Bytes receive	Major	None	None	Default

How to Start

Double-click "Navigation > Alarm management > Alarm settings > Network alarm"

Table Items

Monitor device

Network device

Alarm type:

Package transmit

Bytes transmit

Package receive

Bytes receive

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2.11.1.3 Disk Alarm

This function is used to monitor disk.

The screenshot shows the VOS3000 Disk Alarm configuration interface. The navigation pane on the left includes 'Monitor device' and 'Alarm type' (Disk usage). The main area displays a table of disk usage for three devices: /dev/sda1, /dev/mapper/Vol..., and tmpfs, all showing 80% usage. A graph at the bottom left shows a disk usage trend over time.

Monitor device	Device id	Alarm type	Alarm severity	Upper	Period	Voice alarm	Voice als
SERVER_172.16.5.166	/dev/sda1	Disk usage	Major	80%	Default	Default	
SERVER_172.16.5.166	/dev/mapper/Vol...	Disk usage	Major	80%	Default	Default	
SERVER_172.16.5.166	tmpfs	Disk usage	Major	80%	Default	Default	

How to Start

Double-click "Navigation > Alarm management > Alarm settings > Disk alarm"

Table Items

Monitor device

Device id

Alarm type:

Disk usage

2.11.1.4 Process alarm

This function is used to monitor process.

Monitor device	Process name	Alarm type	Alarm severity	Period	Voice alarm	Voice alarm call number
SERVER_172.16.5.166	webserverd	Process terminated	Major	Default	Default	
SERVER_172.16.5.166	diald	Process terminated	Major	Default	Default	
SERVER_172.16.5.166	audioplayerd	Process terminated	Major	Default	Default	
SERVER_172.16.5.166	callserviced	Process terminated	Major	Default	Default	
SERVER_172.16.5.166	valueaddeddd	Process terminated	Major	Default	Default	
SERVER_172.16.5.166	empd	Process terminated	Major	Default	Default	
SERVER_172.16.5.166	vos3000d	Process terminated	Major	Default	Default	
SERVER_172.16.5.166	mediaagentd	Process terminated	Major	Default	Default	
SERVER_172.16.5.166	mgcd	Process terminated	Major	Default	Default	
SERVER_172.16.5.166	mbx3000d	Process terminated	Major	Default	Default	
SERVER_172.16.5.166	ipcentrextd	Process terminated	Major	Default	Default	
SERVER_172.16.5.166	webdatad	Process terminated	Major	Default	Default	

How to Start

Double-click "Navigation > Alarm management > Alarm settings > Process alarm."

2.11.1.5 Mapping Alarm

This function is used to monitor mapping gateway.

Begin time	End time	Gateway name	Alarm type	Alarm severity	Upper	Lower	Period
00:00:00	24:00:00	mapping-1	Mapping acd	Minor	None	10 (s)	Default
00:00:00	24:00:00	mapping-1	Mapping asr	Minor	None	None	Default
00:00:00	24:00:00	mapping-1	Mapping bilateral reconc...	Minor	10.000	-10.000	Default
00:00:00	24:00:00	mapping-1	Mapping concurrency dec...	Minor	None	None	60
00:00:00	24:00:00	mapping-1	Mapping gateway call rate	Minor	50(cps)	None	Default
00:00:00	24:00:00	mapping-1	Mapping gateway concurr...	Minor	None	None	60

How to Start

Double-click “Navigation > Alarm management > Alarm settings > Mapping alarm”

Table Items

Begin time

End time

Gateway name

Alarm type:

Mapping acd

Mapping asr

Mapping bilateral reconciliation deviation

Mapping concurrency decline

Mapping gateway call rate

Mapping gateway concurrent rise

Downloaded from VOS3000.Com

2.11.1.6 Routing Alarm

This function is used to monitor routing gateway.

Begin time	End time	Gateway name	Alarm type	Alarm severity	Upper	Lower	Period	Voice a
00:00:00	24:00:00	callee.123	Routing acd	Minor	None	10 (s)	Default	Default
00:00:00	24:00:00	callee.123	Routing asr	Minor	None	None	Default	Default
00:00:00	24:00:00	callee.123	Routing bilatera...	Minor	10.000	-10.000	Default	Default
00:00:00	24:00:00	callee.123	Routing concurr...	Minor	None	None	60	Default
00:00:00	24:00:00	callee.123	Routing gatewa...	Minor	10	None	60	Default
00:00:00	24:00:00	callee.123	Routing gatewa...	Minor	None	None	60	Default

How to Start

Double-click "Navigation > Alarm management > Alarm settings > Routing alarm"

Table Items

Begin time

End time

Gateway name

Alarm type:

Routing acd

Routing asr

Routing bilateral reconciliation deviation

Routing concurrency decline

Routing gateway call

Routing gateway concurrent rise

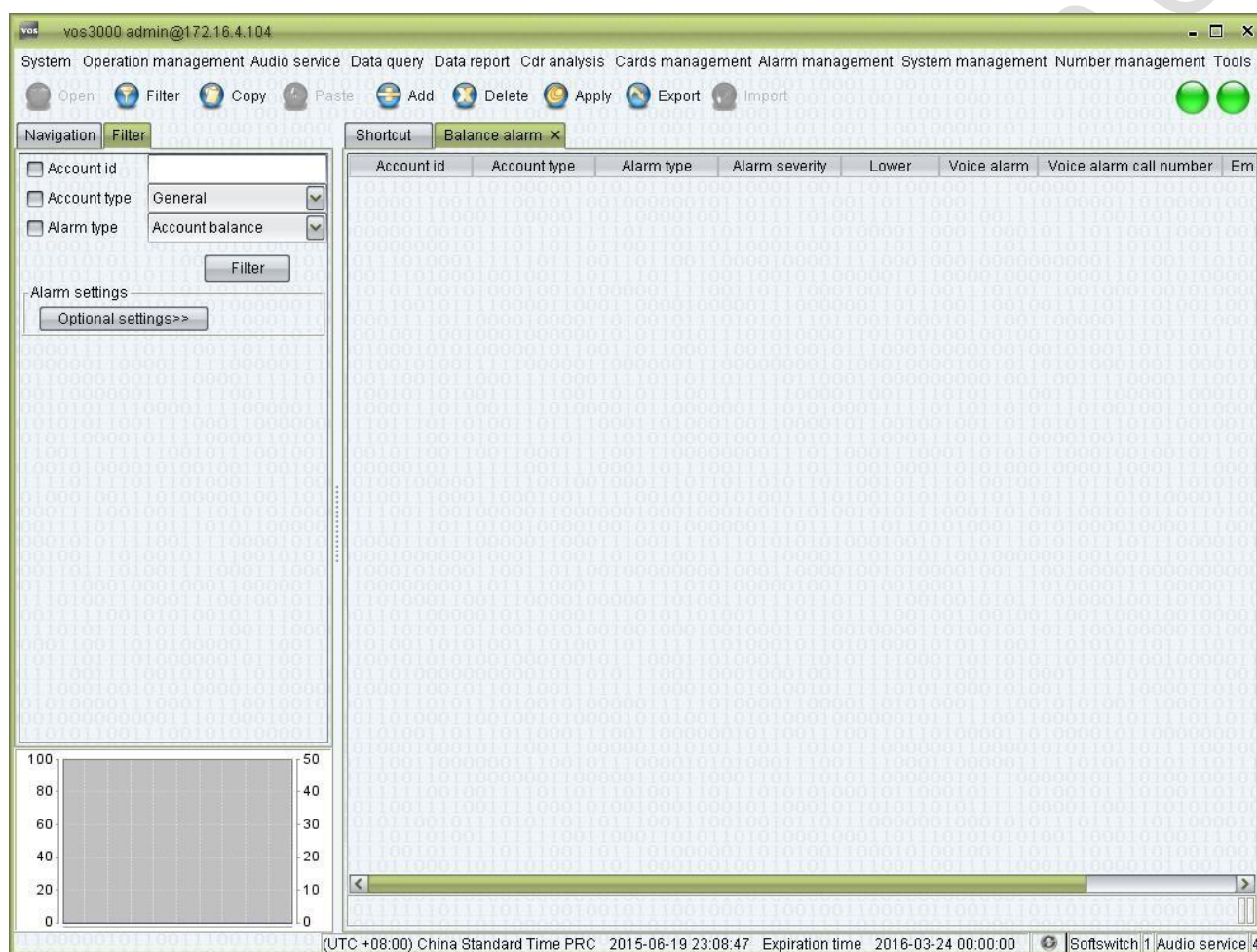
Downloaded from VOS3000.Com

2.11.1.7 Balance Alarm

This function is used to monitor account balance.

NOTE

The number of customer can be set by “System management > System parameter > SERVER_ALARM_CUSTOMER_BALANCE_MAX_SIZE”.

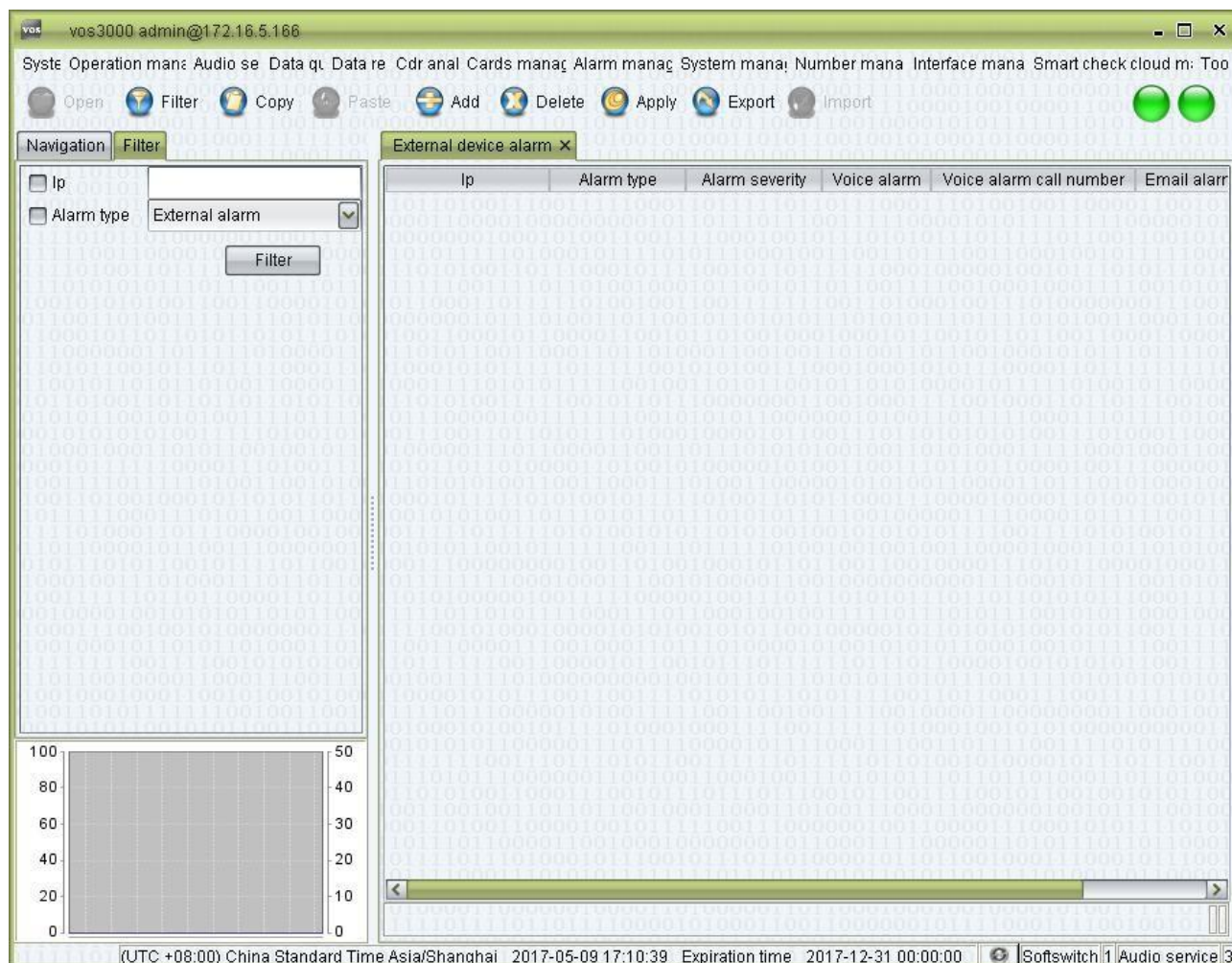


How to Start

Double-click “Navigation > Alarm management > Alarm settings > Balance alarm”

2.11.1.8 External Device Alarm

This function is used to monitor external device.

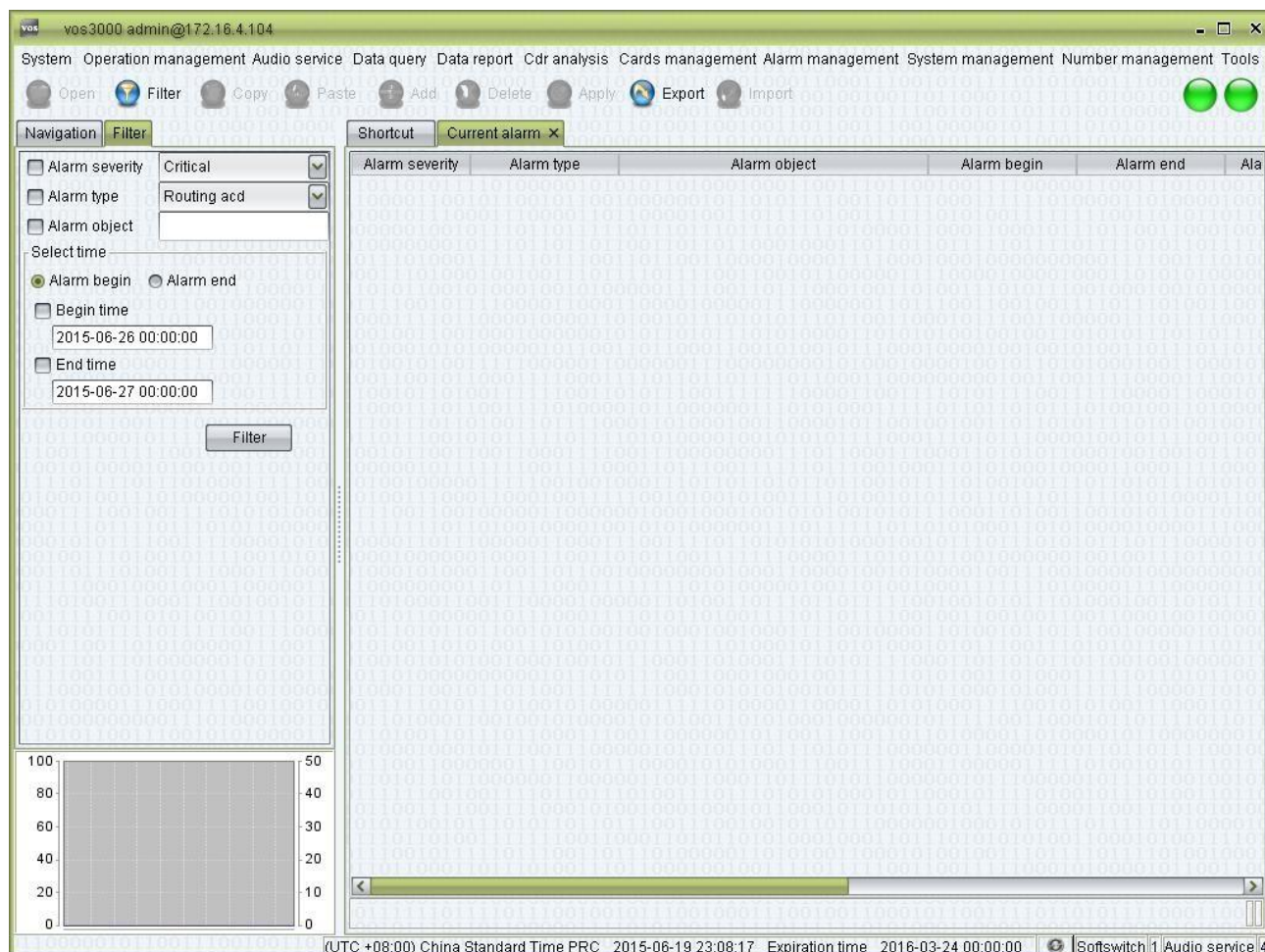


How to Start

Double-click "Navigation > Alarm management > Alarm settings > External device alarm"

2.11.2 Current Alarm

This function is used to manage current alarm.



How to Start

Double-click “Navigation > Alarm management > Current alarm”

Table Items

Alarm severity
Alarm type
Alarm object
Alarm begin
Alarm end
Alarm value

Upper

Lower

ACK user

ACK time

Memo

Right-Click Menu

Confirm: input memo to confirm alarm.

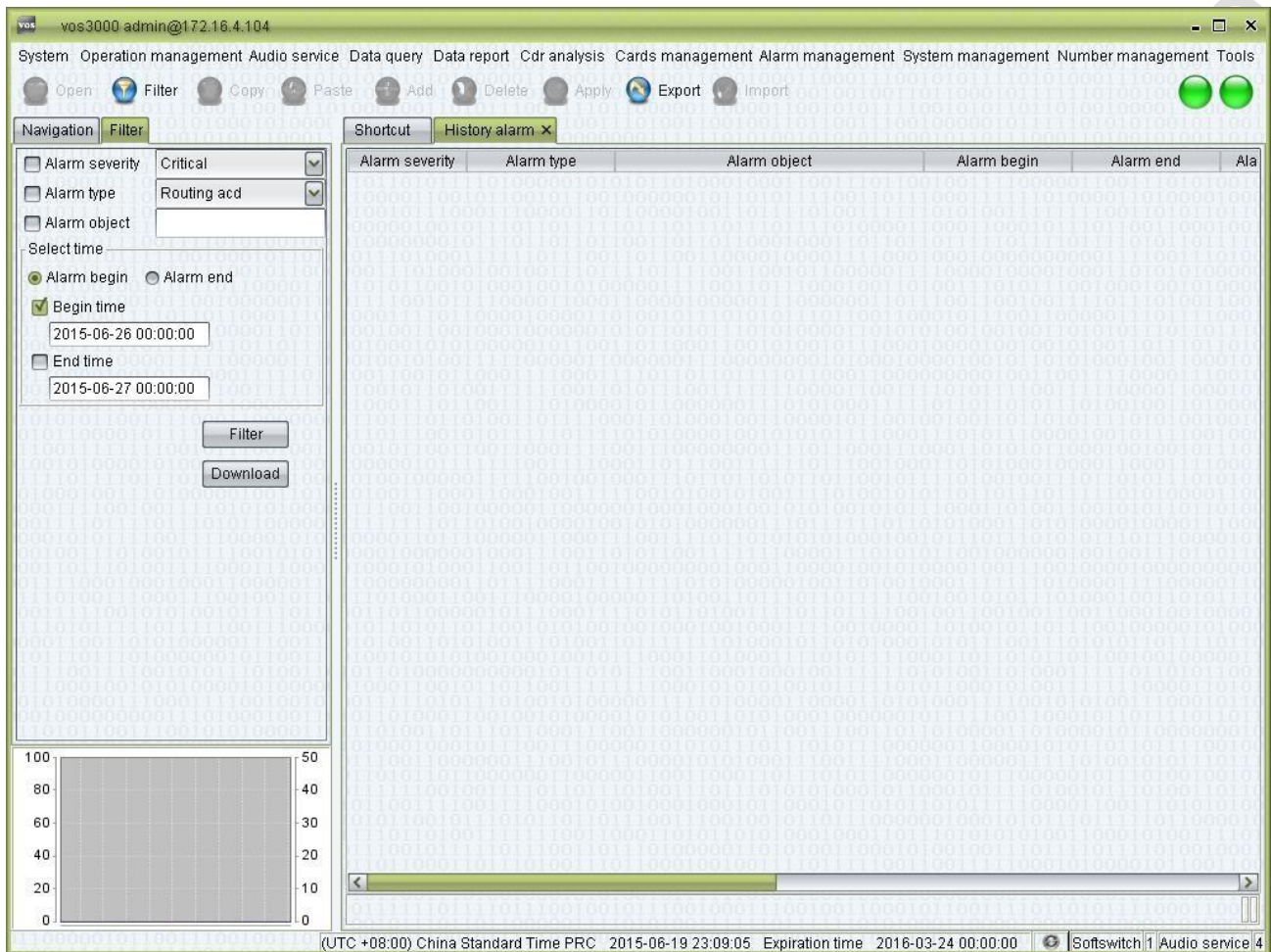
Empty: clear current alarm, and then change the alarm into history.

2.11.3 History Alarm

This function is used to query history alarm.

NOTE

If the current alarm, which is stopped, occurs again, the former alarm will be become history alarm.



How to Start

Double-click "Navigation > Alarm management > History alarm"

Table Items

Alarm severity

Alarm type

Alarm object

Alarm begin

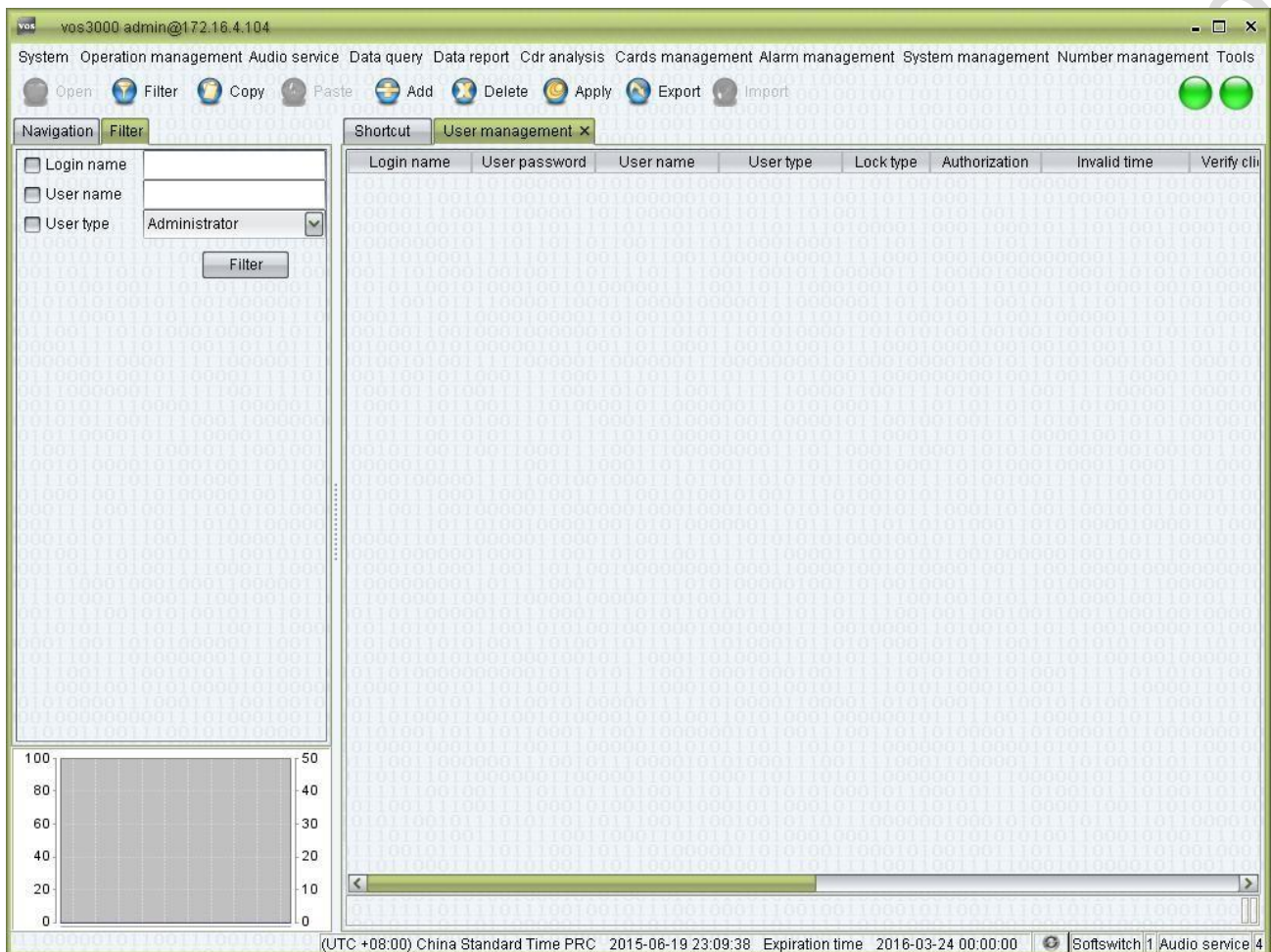
Alarm end
Alarm value
Upper
Lower
ACK user
ACK time
Memo
Cleared user
Cleared time

Downloaded from VOS3000.Com

2.12 System Management

2.12.1 User Management

This function is used to manage user.



How to Start

Double-click “Navigation > System management > User management”

Table Items

Login name: the id used for login.

User password: the login password.

User name: the name of the user.

User type

- Administrator: users with all authorizations.
- Operator: users with certain authorizations for operations.
- Agent: users that are only allow viewing the accounts, rate and service packages.

Lock type:

No lock: User permissions can be used normally.

Locked: User permission cannot be used.

Authorization: see below.

Invalid time: After this time the user is not allowed to log in, the login user is often used for temporary.

Verify client mac:

Off: Does not verify client network card Mac address.

Trigger alarm: If someone login vos with a network card Mac address which is not in <Client Mac List>, an alarming will be trigger when configured <Login Mac Restricted> warning in <System Alarm>.

Ban Login: Forbid user to login vos if the login client has a network card Mac address which is not in <Client Mac List>.

Client mac list: Fill in network card Mac address.

Memo: comments about the user.

Dynamic password: choose user and right click to enable or close this feature



NOTE

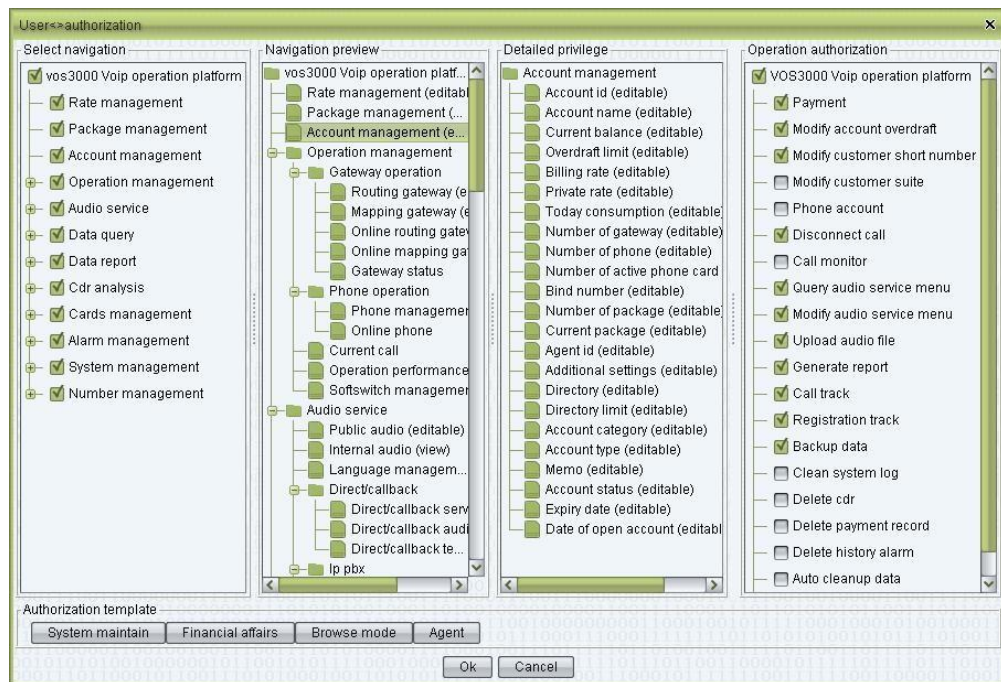
There will be an "Extraction code" window after you enable this feature, bind "app" with "Extraction code", then you can use dynamic password to login vos client.

App download link: http://www.linknat.com/chs/support/downloads/vos_otp.apk

Last login: last login time.

Last change password: last change password time.

Current client mac: display the current login client's network card Mac address.



Select navigation: define user's navigation tree.

Navigation preview: preview user's navigation tree, double click to change privilege (view or editable).

Detailed privilege: display the detailed content of navigation tree, double click to change privilege (view or editable).

Operation authorization: define the allowed operation privilege by user

Users can specify interfaces and operations available for a non-administrator user. All settings come into effect immediately.

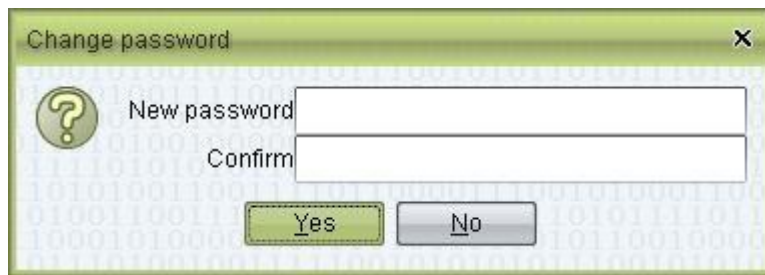
Authorization template: several templates for authorizations are provided. Users can select a template and then tune the configurations.

NOTE

Users of the "agent" type who have the authorization to create users will be able to see all the users they created in the table, while other users are invisible to them. It is the same when they specify the availability of rates, packages, and accounts to other users.

Right-Click Menu

Change password

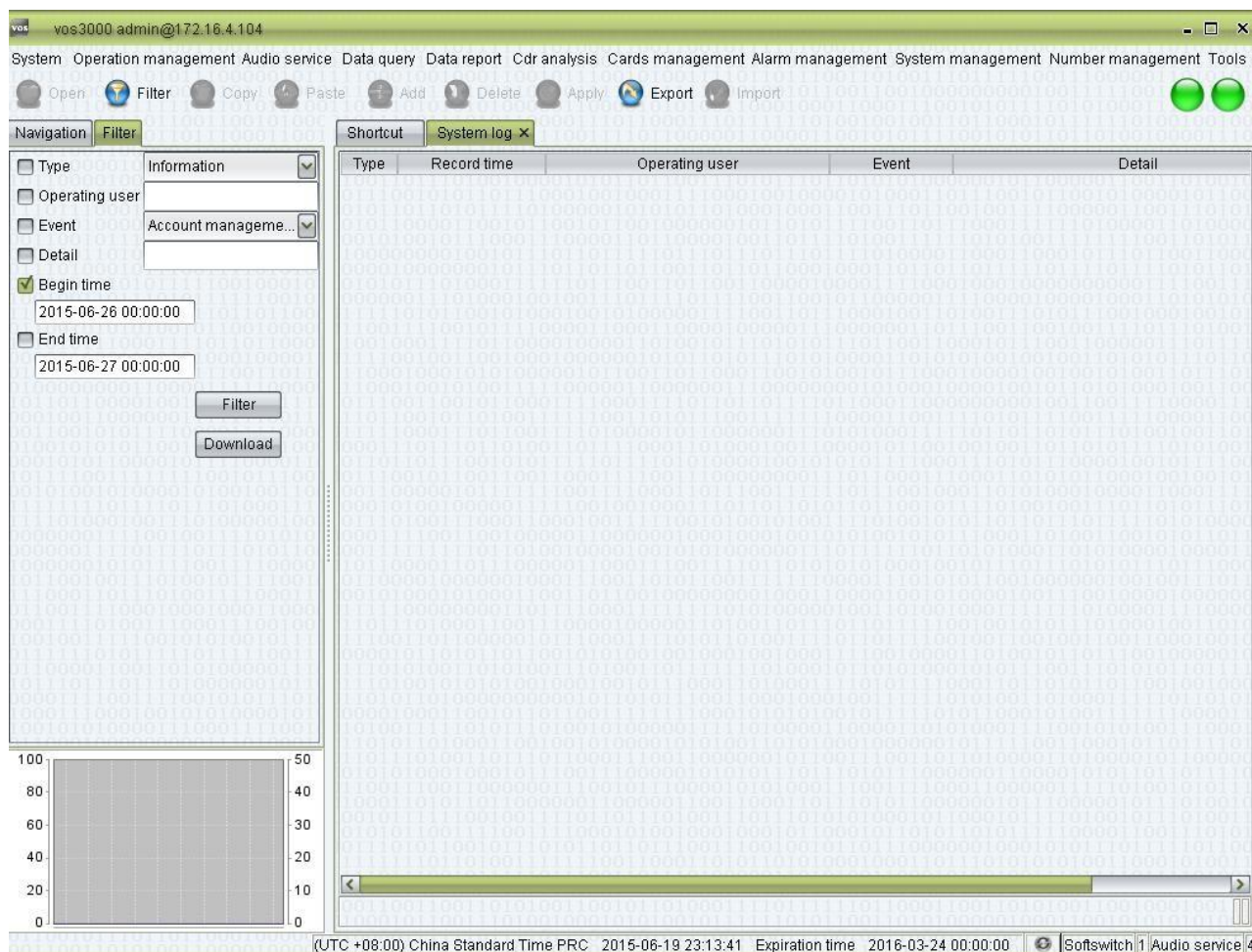


Password requirements:

- at least 6 characters (at least 8 characters for administrator)
- at least 2 of the following character combination:
 - at least one lowercase letter
 - at least one uppercase letter
 - at least one number
 - at least one special character: `~!@#\$%^&*()-_+=\|[]{};:'",<.>/? and space
(space cannot be used as starting or ending character.)
- cannot be the same as the name or the reverse of name

2.12.2 System Log

This function is used to query system log.



How to Start

Double-click “Navigation > System management > System log”

Table Items

Type: Information/General/Error

Record time: log time.

Operating User

IP:user logging IP.

Event

Detail

Right Click Menu

Detail

This function is used to show operating user and operation's detail.

System log

Type	Information	Record time	2013-09-03 14:53:50	Previous
Operating user	admin@172.16.1.131	Event	Rate management	Next
Detail	Rate group test modify rate 1			

Data detail

Original data	Update data
<pre>DB_E_FEERATE { feeprefix= areacode=1 locktype=0 fee=0.01 period=6 ivrfree=0.0 ivrperiod=6 type=4 feerategroup_id=324 feeratesections=[3] }</pre>	<pre>DB_E_FEERATE { feeprefix= areacode=1 locktype=0 fee=0.1 period=6 ivrfree=0.0 ivrperiod=6 type=4 feerategroup_id=324 feeratesections=[3] }</pre>

Off

2.12.3 System Parameter

This function is used to configure parameters.

Parameter name	Parameter value	Parameter description
EXTERNAL_WEB_SEND_PHONE_ONLINE	Off	Interface: phone online/offline transfer
SERVER_ALARM_CUSTOMER_BALANCE_MAX_SI...	1000	Max size of balance alarm's accounts 1000-5000
SERVER_ALARM_DATABASE_IGNORE_ERROR...		Database error code without warning
SERVER_ALARM_DISABLE	Off	Alarm suppression
SERVER_ALARM_E164S		Voice alarm call number
SERVER_ALARM_EMAIL		Alarm email address
SERVER_ALARM_EMAIL_DELAY	30	Delay of email alarm(seconds) 0-7200
SERVER_ALARM_ENABLE_EMAIL	Off	Email alarm
SERVER_ALARM_ENABLE_VOICE	Off	Voice alarm
SERVER_BILLING_FEE_PRECISION	0.0000000	Billing money accuracy 0-1000
SERVER_BILLING_FEE_UNIT	0.0000000	Billing money unit 0-1000
SERVER_BILLING_FORWARD_PREFIX		Call forward billing prefix
SERVER_BILLING_FREE_E164S		Special service number(charge free, with cdr, separated by
SERVER_BILLING_FREE_TIME	0	Free call duration(seconds) 0-86400
SERVER_BILLING_HOLD_TIME_PRECISION	1000	Billing call duration precision(milliseconds) 1-1000
SERVER_BILLING_NO_CDR_E164S		Free called number(no cdr, separated by commas
SERVER_BILLING_PROFIT_CALCULATE	<Call charges> ...	Call profit calculation
SERVER_BILLING_RECODR_SERVER_HANG_UP	On	Record server hang up call
SERVER_BILLING_RECODR_ZERO_HOLD_TIME	On	Record cdr with non call duration
SERVER_CDR_FILE_WRITE_INTERVAL	None	Additional write call record file, the new file created time inter
SERVER_CDR_FILE_WRITE_MAX	2048	Additional write call record file, the maximum number of rese
SERVER_DATABASE_VERSION	2.1.6.00	Version of database table structure
SERVER_DISPLAY_CHART_GATEWAY_SIZE	10	Default gateway number in gateway chart 1-50
SERVER_DISPLAY_CUSTOMER_ALARM_MONEY	20	Account warning money 0-100000
SERVER_DISPLAY_MONEY_PRECISION	3	Money display precision 2-7
SERVER_DNS_UPDATE_INTERVAL	600	Domain update cycle 300-86400
SERVER_GATEWAY_ROUTE_PREFIX		Routing additional routing prefix, separated by commas.
SERVER_IPPBX_DEFAULT_LANGUAGE	chinese	Default language of ip pbx service
SERVER_LOGIN_FAILED_DISABLE_TIME	120	Time of disable user login when failed several times(second
SERVER_MAIL_CUSTOMER_BILL	Off	Smtp send account consumption email automatically
SERVER_MAIL_REPORT_DAY_OF_WEEK	Monday	Time of weekly email
SERVER_MAIL_SMTP_FROM	127.0.0.1	Smtp domain name
SERVER_MAIL_SMTP_FROM_USER		Smtp sender
SERVER_MAIL_SMTP_SERVER	127.0.0.1	Smtp server ip
SERVER_MAIL_SYSTEM_NOTICE		System notification email address
SERVER_MAX_CDR_PENDING_LIST_LENGTH	100000	Length of cdr queue limit 10000-100000

How to Start

Double-click "Navigation > System management > System parameter"

Table Items

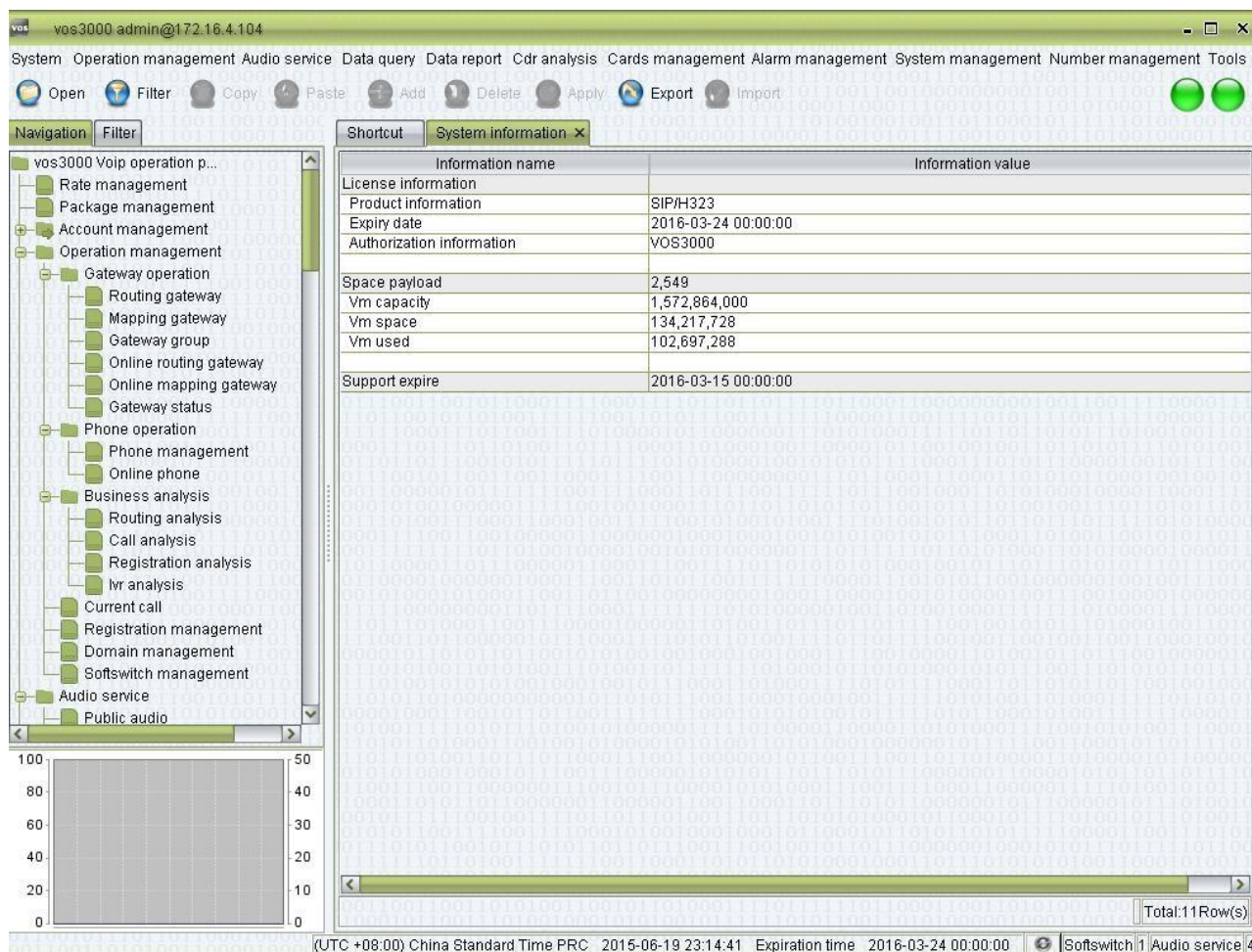
Parameter name

Parameter value

Parameter description

2.12.4 System Information

This function is used to show server's information.



Information name	Information value
License information	SIP/H323
Product information	SIP/H323
Expiry date	2016-03-24 00:00:00
Authorization information	VOS3000
Space payload	2,549
Vm capacity	1,572,864,000
Vm space	134,217,728
Vm used	102,697,288
Support expire	2016-03-15 00:00:00

Total: 11 Row(s)

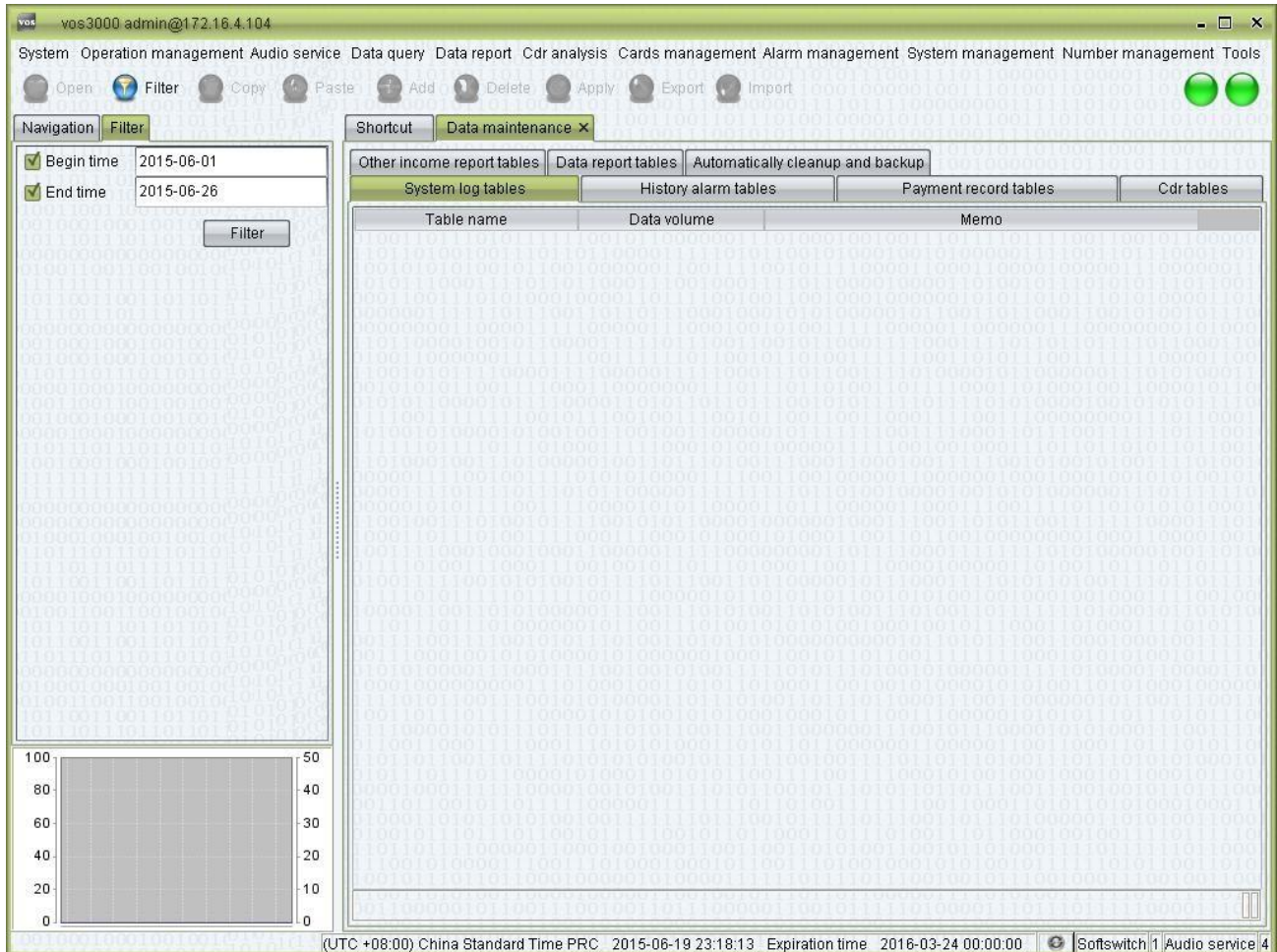
(UTC +08:00) China Standard Time PRC 2015-06-19 23:14:41 Expiration time 2016-03-24 00:00:00 Softswitch 1 Audio service 4

How to Start

Double-click "Navigation > System management > System information"

2.12.5 Data Maintenance

2.12.5.1 System log Tables



How to Start

Double-click "Navigation > System management > Data maintenance"

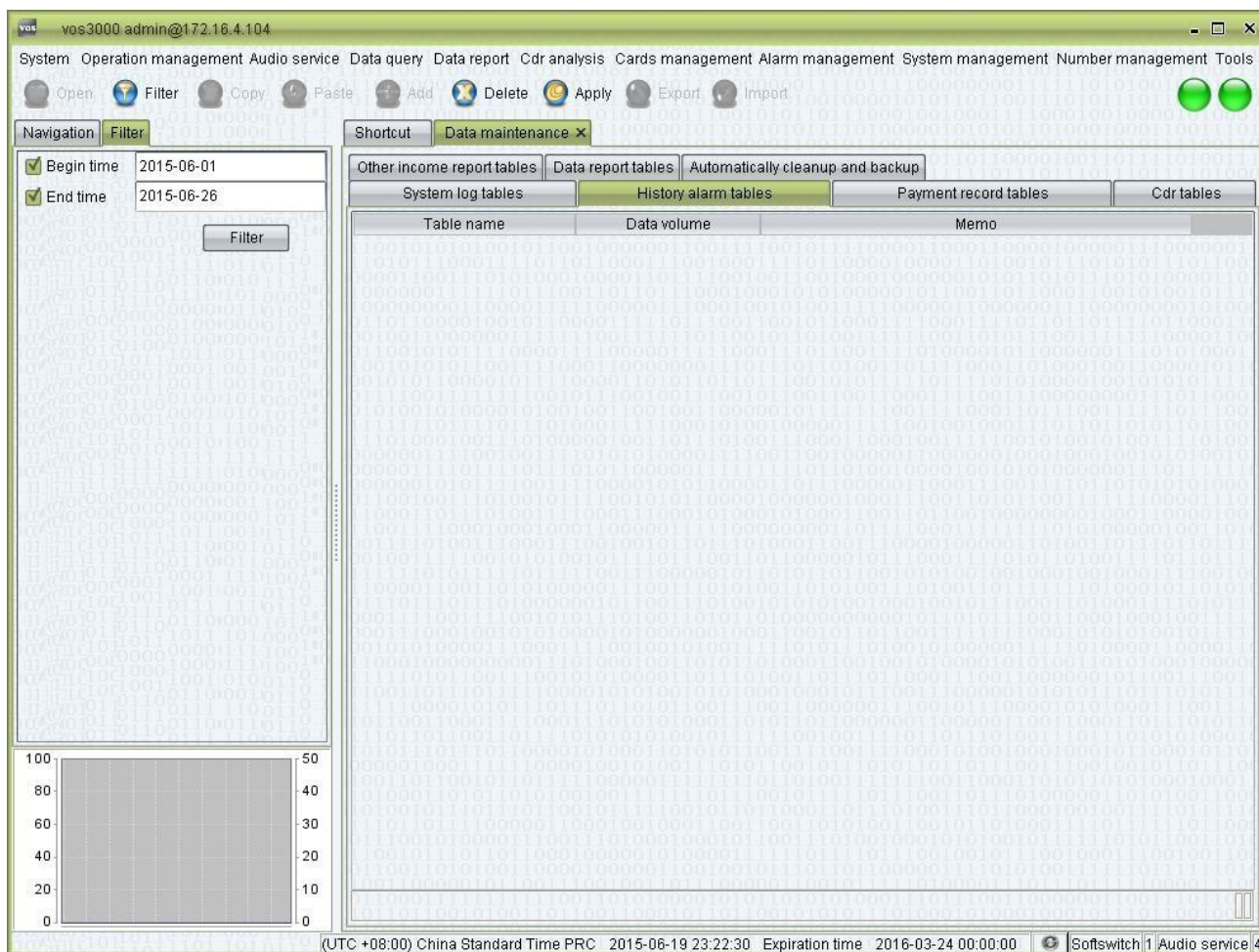
Table Items

Table name: suffix is the date of system log.

Data volume: number of records.

Memo

2.12.5.2 History Alarm Tables



How to Start

Double-click "Navigation > System management > Data maintenance"

Table Items

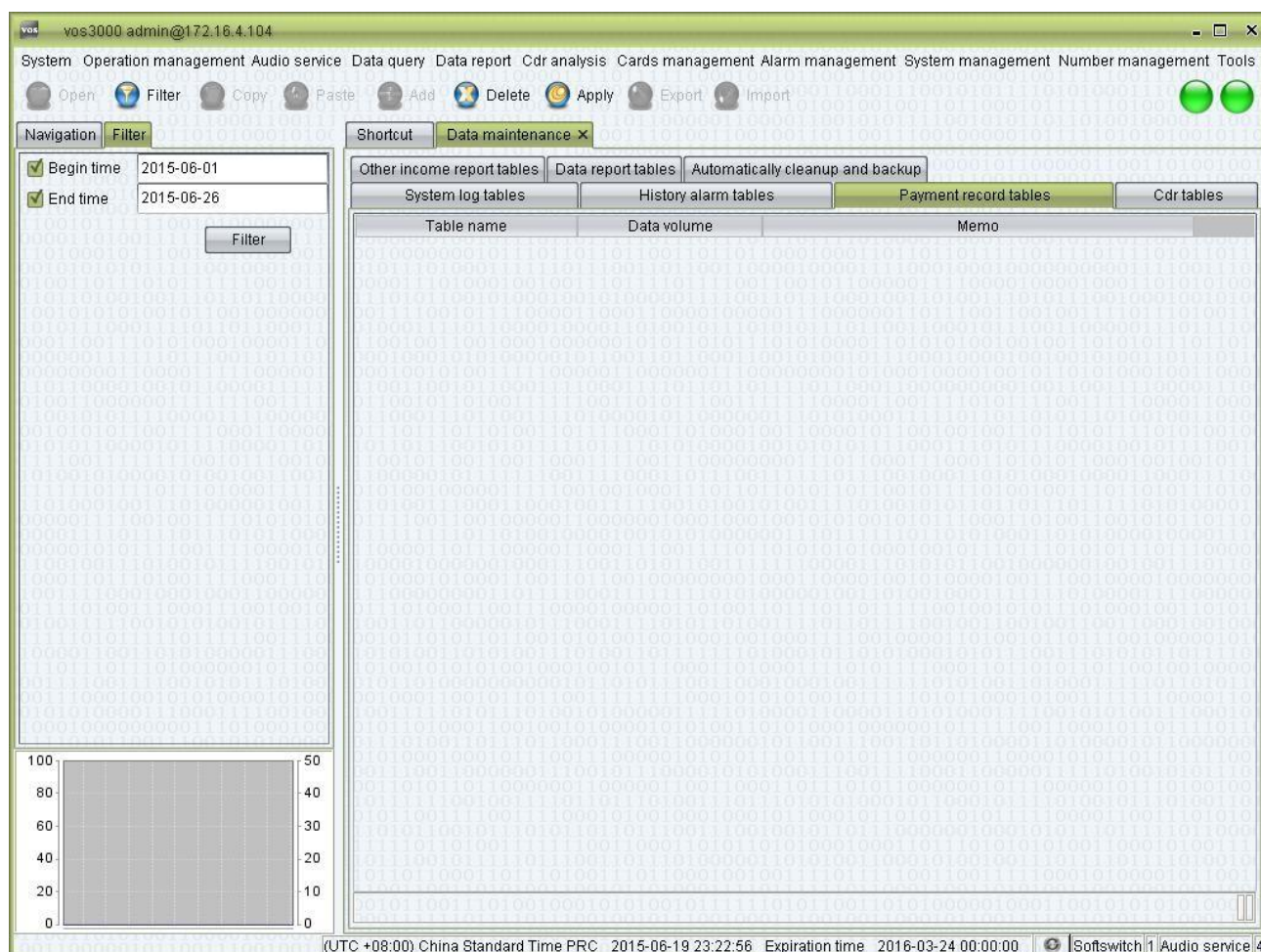
(VOS3000.Com)

Table name: suffix is the date of history alarm.

Data volume: number of records.

Memo

2.12.5.3 Payment Record Tables



How to Start

Double-click "Navigation > System management > Data maintenance"

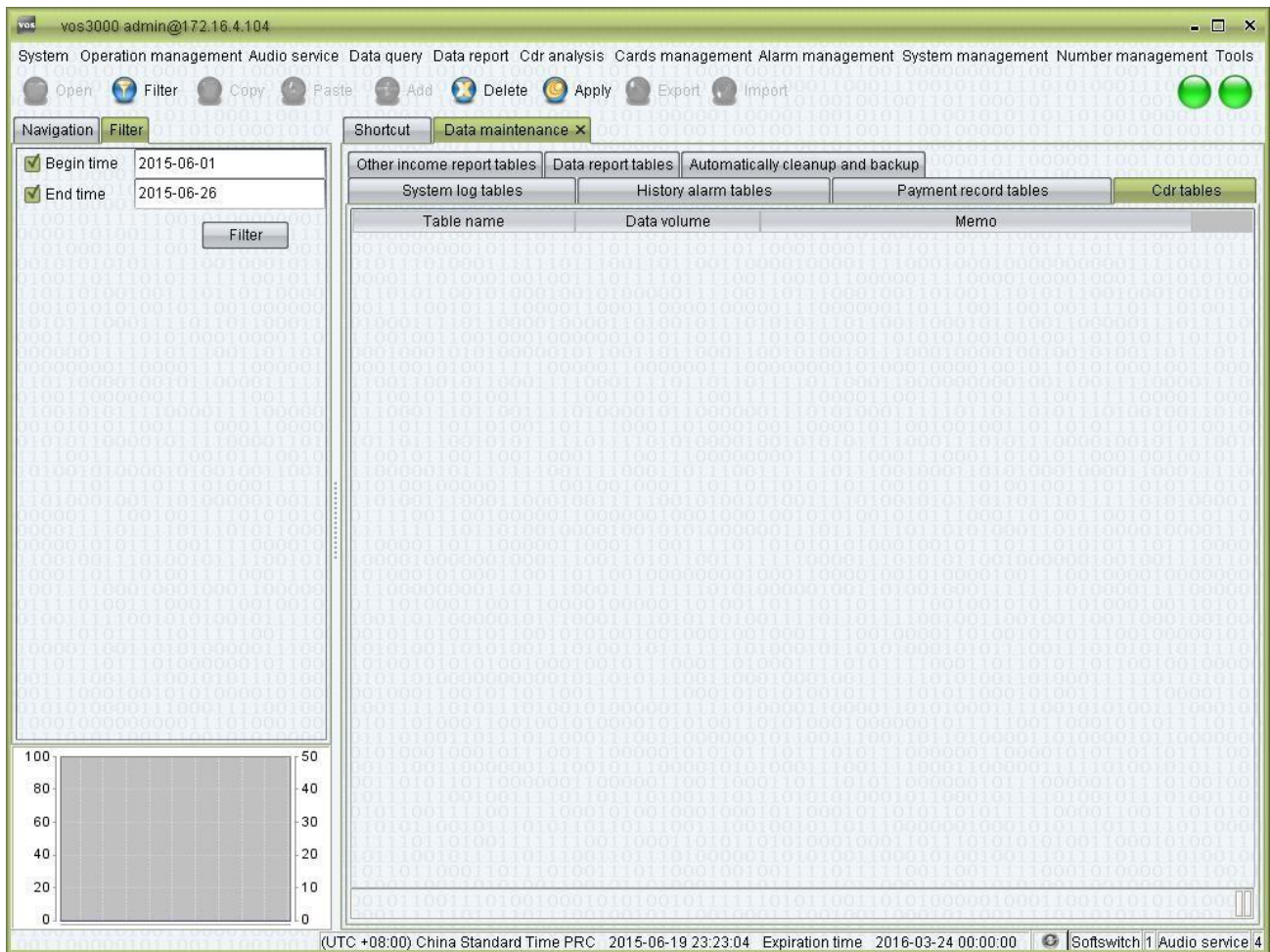
Table Items

Table Name: suffix is the date of payment record.

Data Volume: number of records.

Memo

2.12.5.4 CDR Tables



How to Start

Double-click "Navigation > System management > Data maintenance"

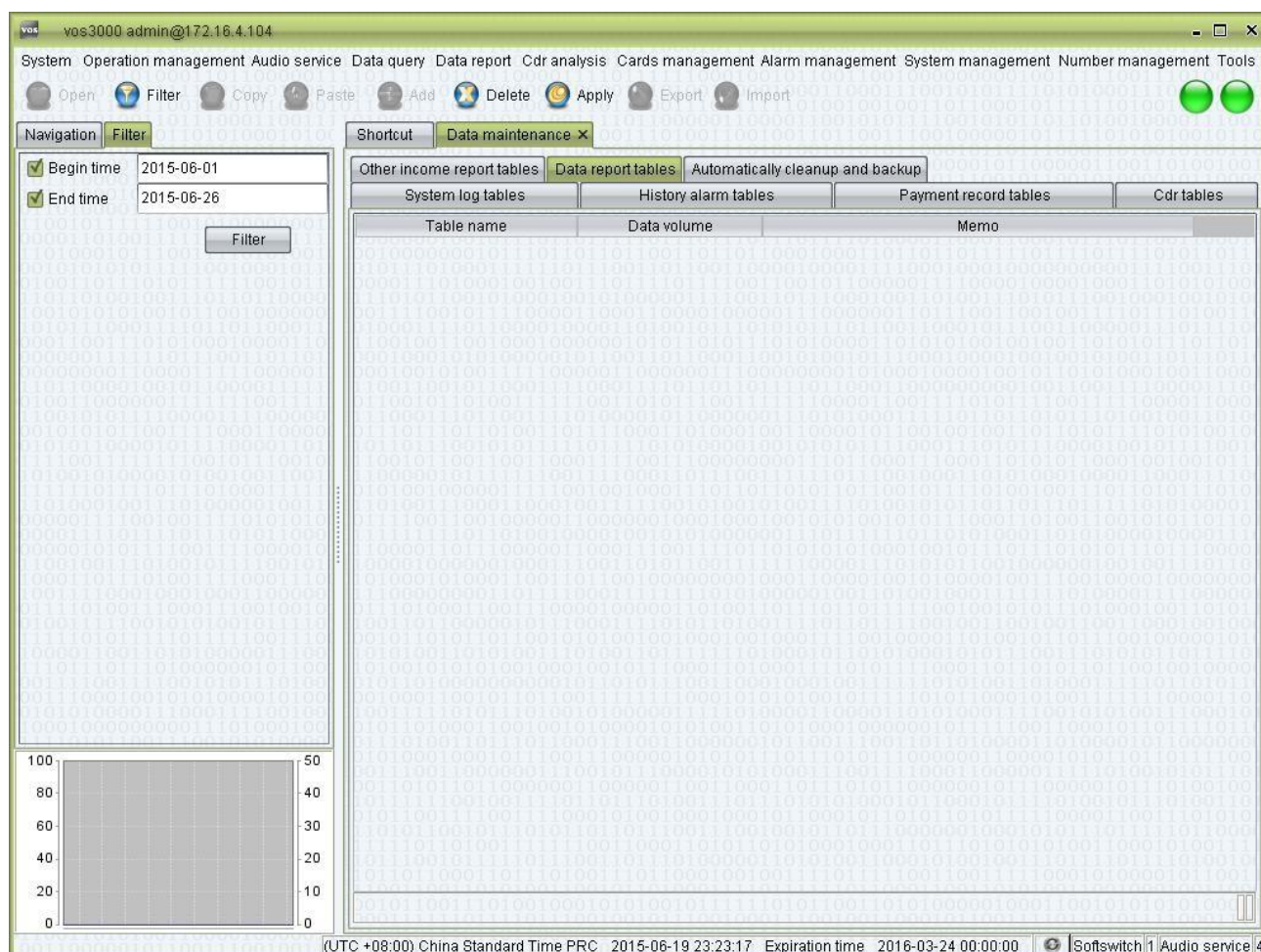
Table Items

Table name: suffix is the date of CDR.

Data volume: number of records.

Memo

2.12.5.5 Data Report Tables



How to Start

Double-click "Navigation > System management > Data maintenance"

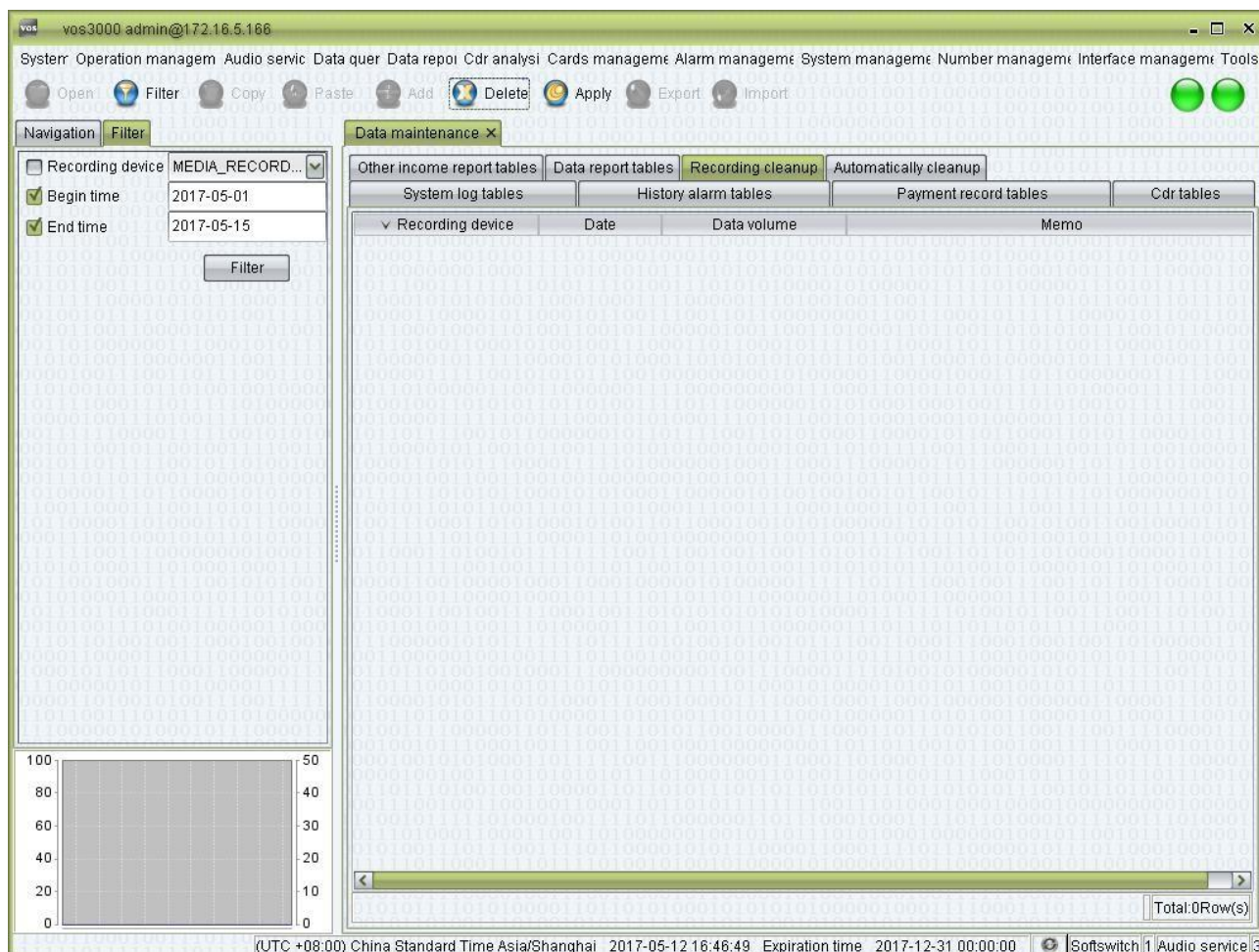
Table Items

Table name: suffix is the date of report.

Data volume: number of records.

Memo

2.12.5.6 Recording cleanup



How to Start

Double-click "Navigation > System management > Recording cleanup"

Table Items

Recording device: name of recording device

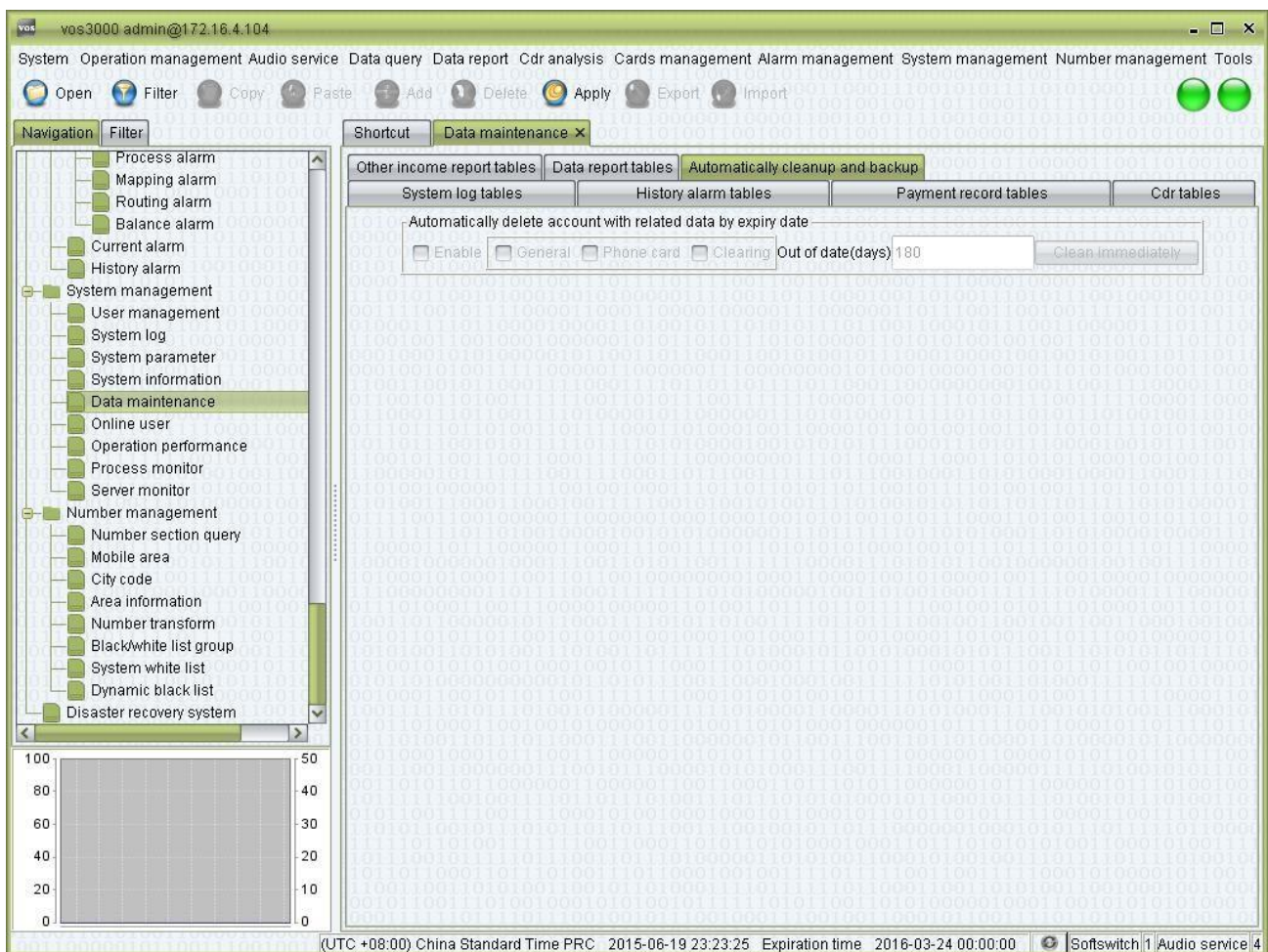
Date: date of recording

Data volume: number of records.

Memo

Downloaded from VOS3000.Com

2.12.5.7 Automatically CleanUp and Backup



How to Start

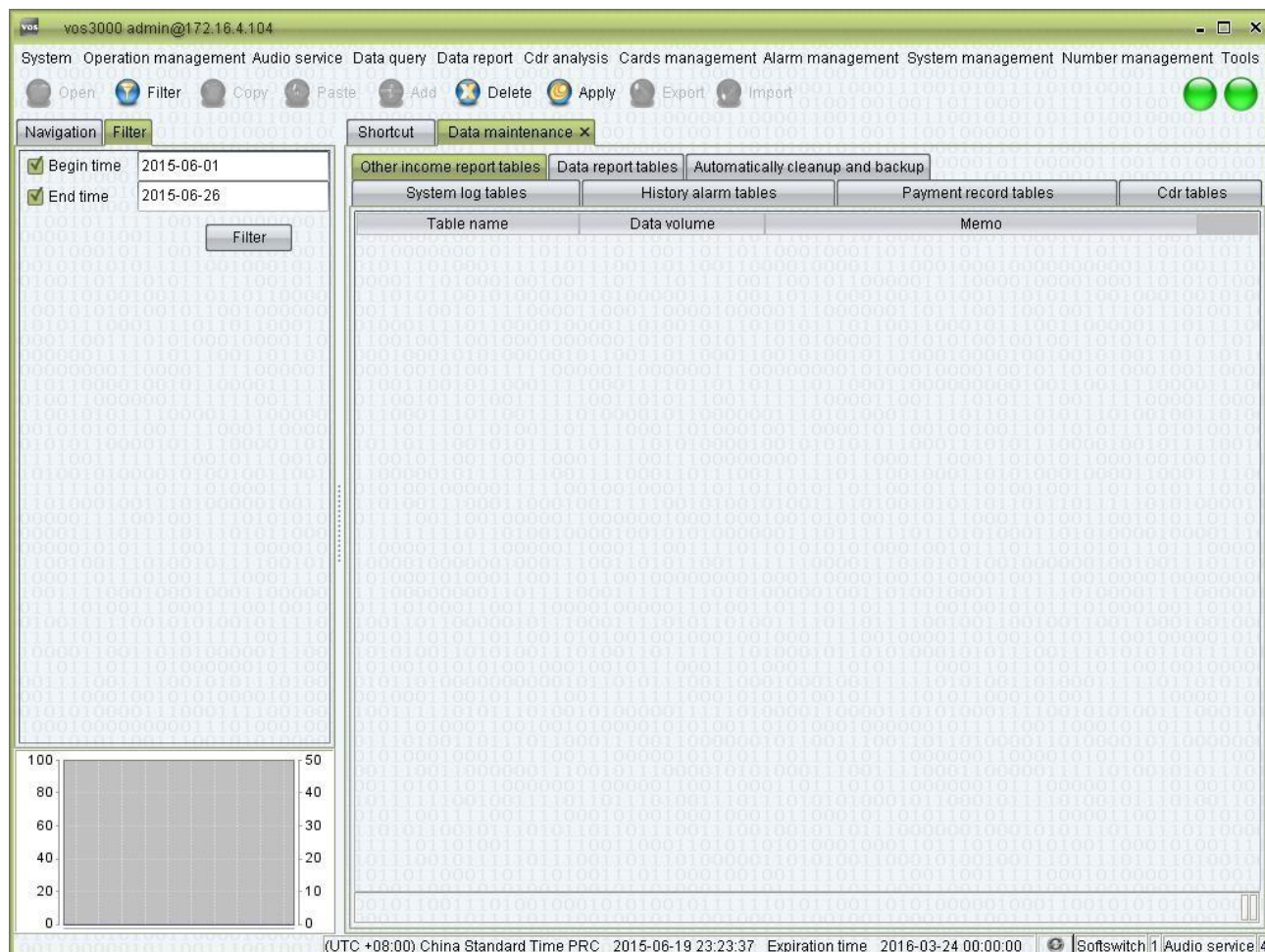
Double-click "Navigation > System management > Data maintenance"

Usage

Use "Filter" to get current settings.

If auto cleanup is enabled, vos will cleanup out of date data every day, including account/account's gateway/account's phone.

2.12.5.8 Other income report tables



How to Start

Double-click "Navigation > System management > Data maintenance"

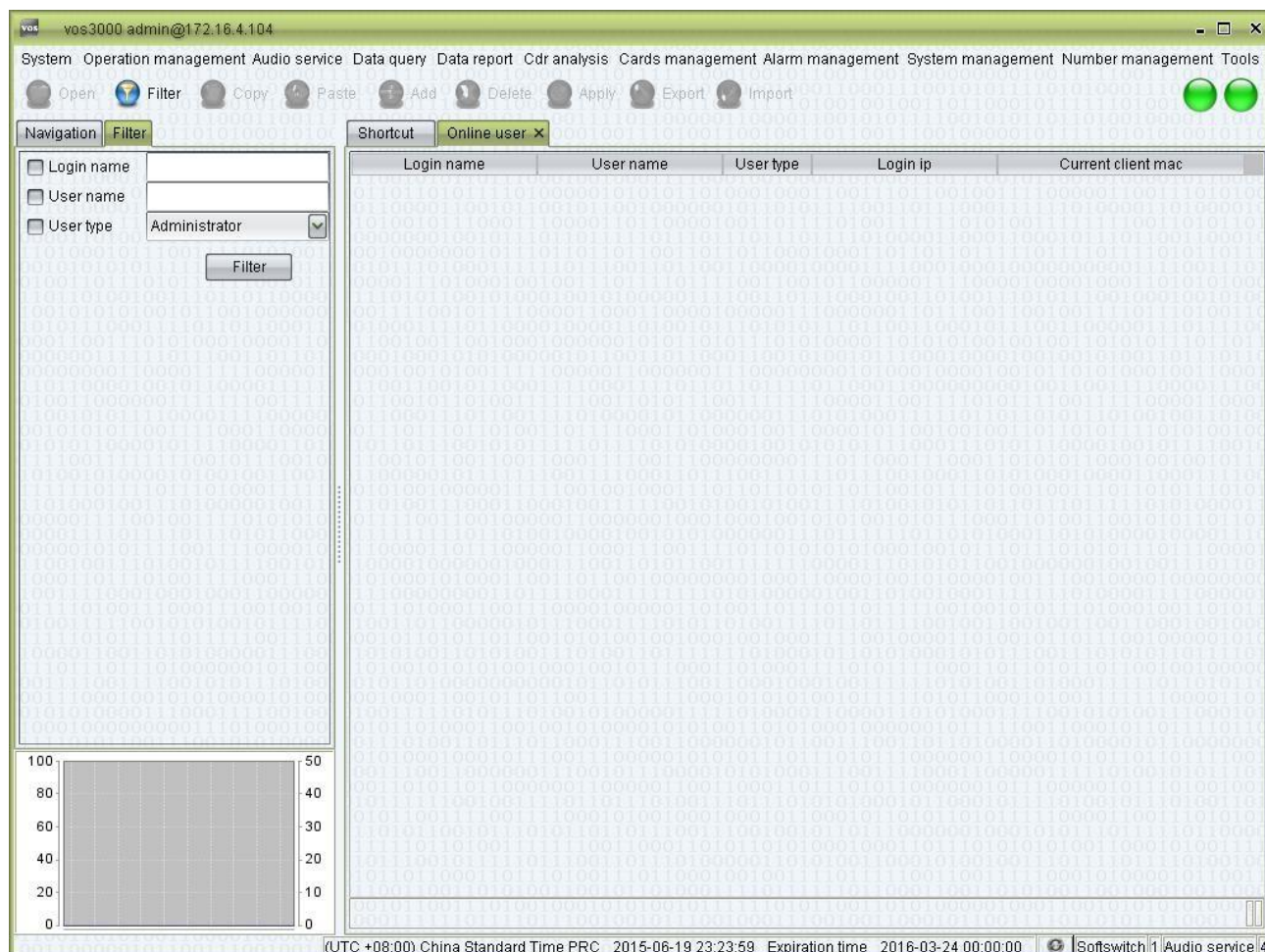
Usage

Use "Filter" to get current settings.

If auto cleanup is enabled, vos will cleanup out of date data every day, including account/account's gateway/account's phone.

2.12.6 Online User

This function is used to query online user.

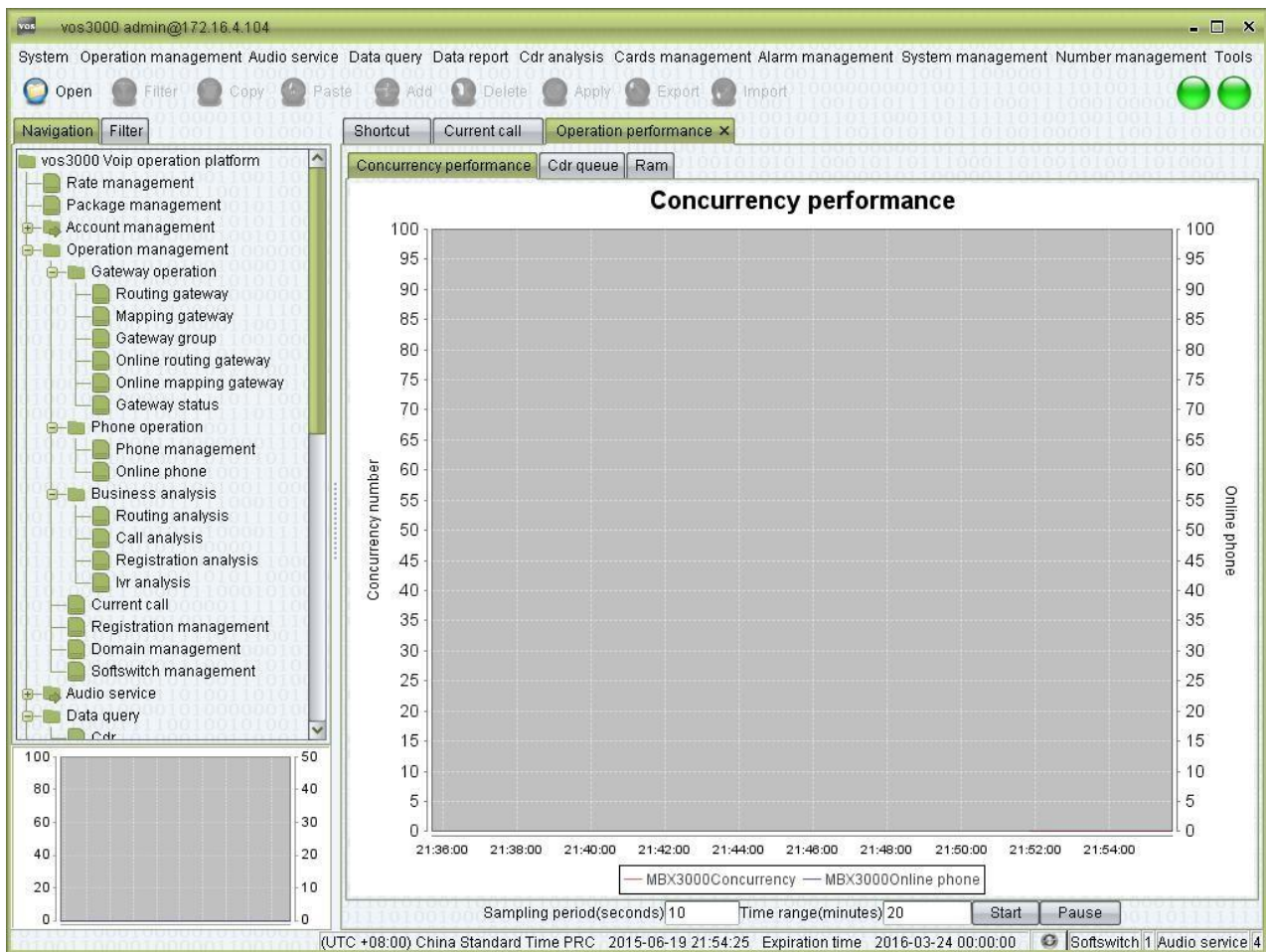


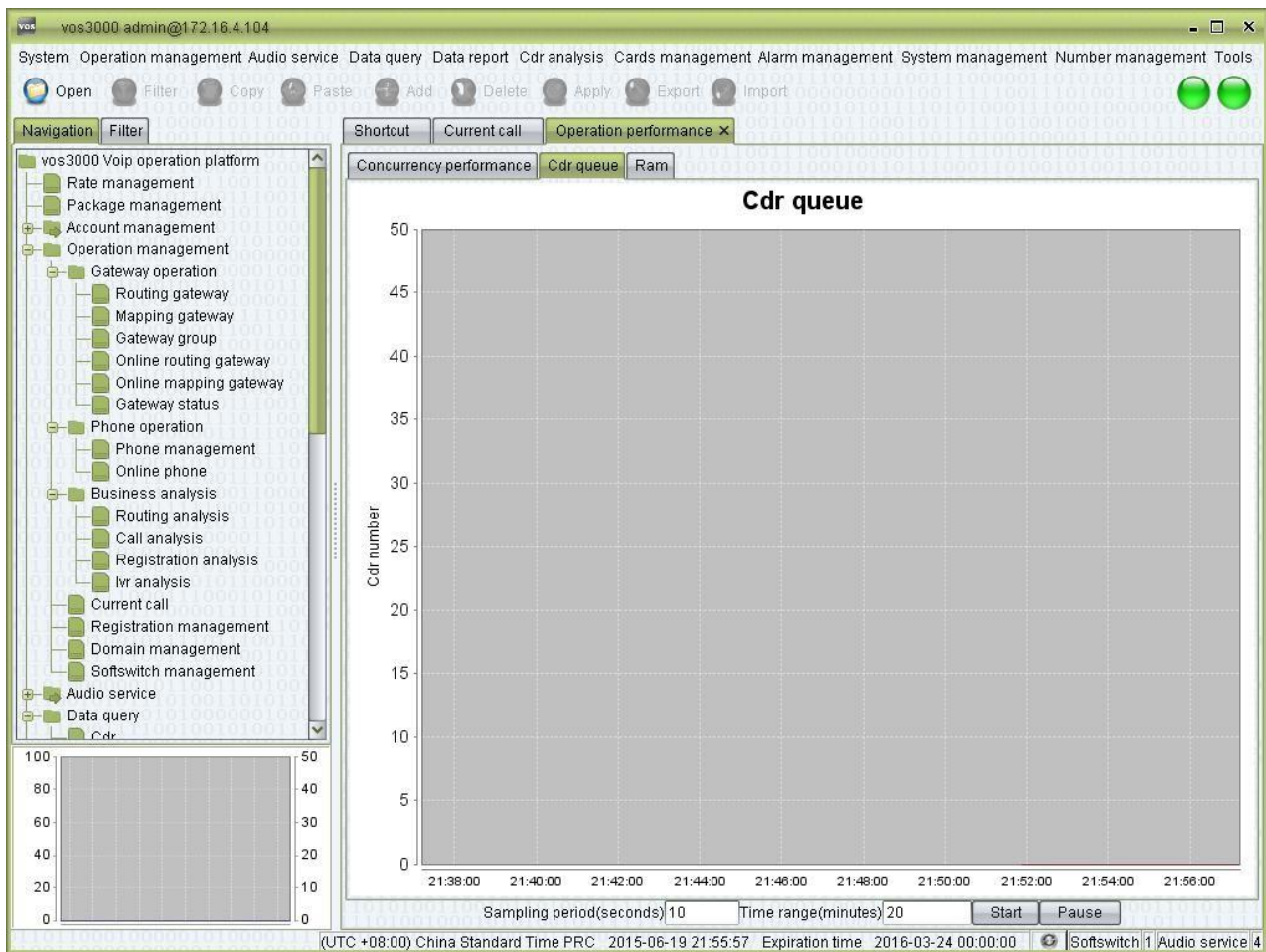
How to Start

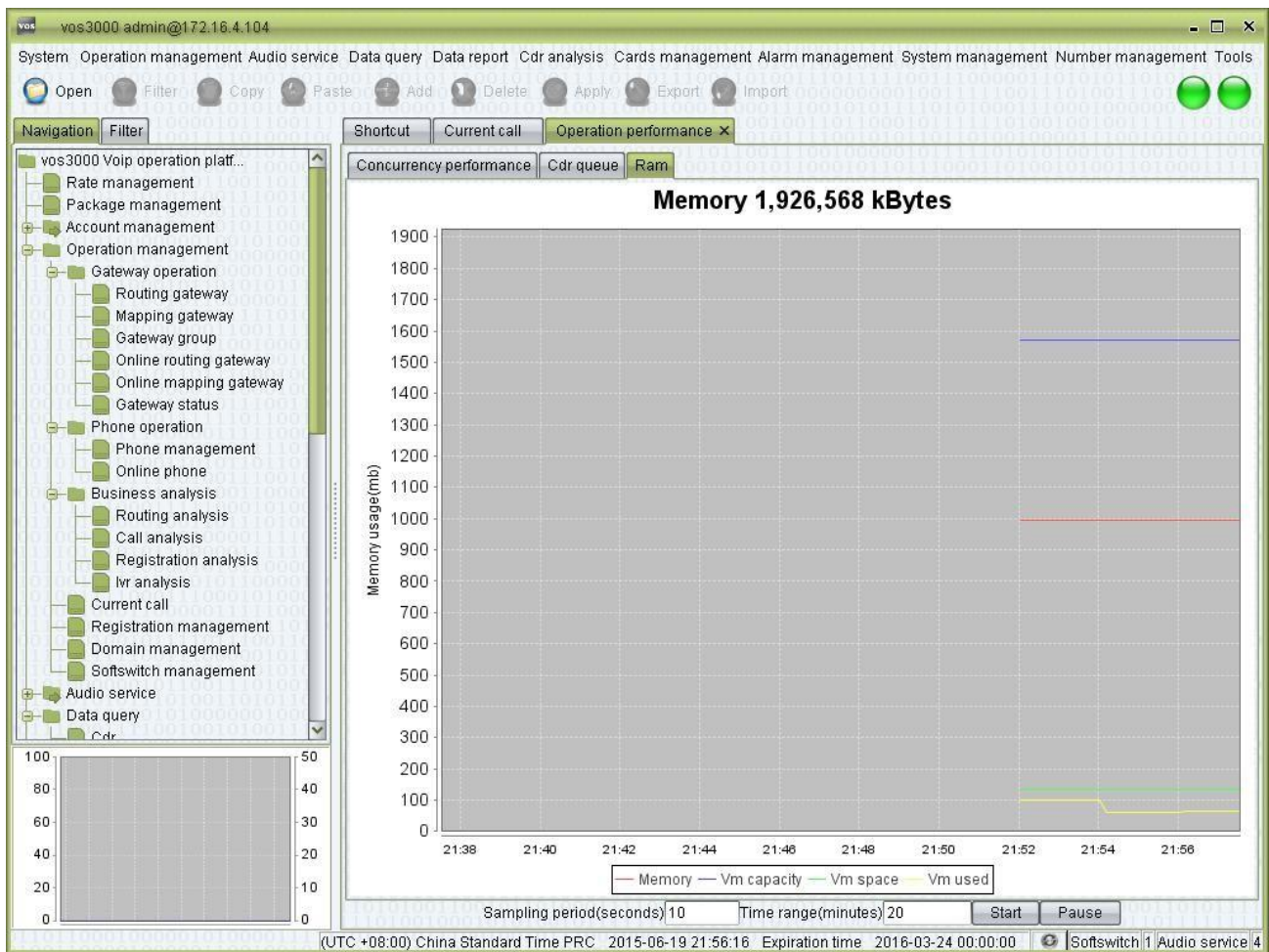
Double-click "Navigation > System management > Online user"

2.12.7 Operation performance

This function is used to monitor system performance.





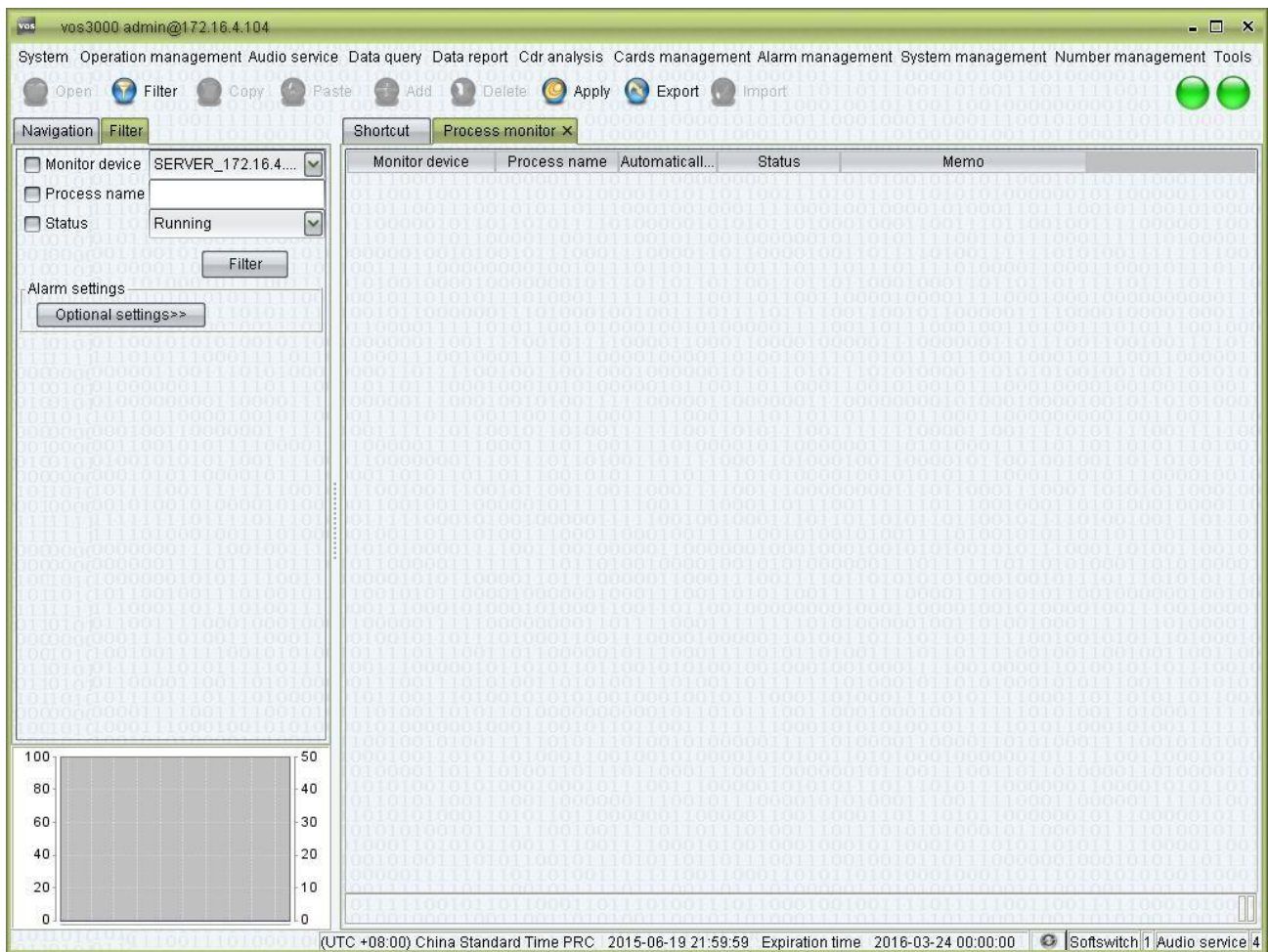


How to Start

Double-click "Navigation > System management > Operation performance"

2.12.8 Process monitor

This function is used to monitor process.



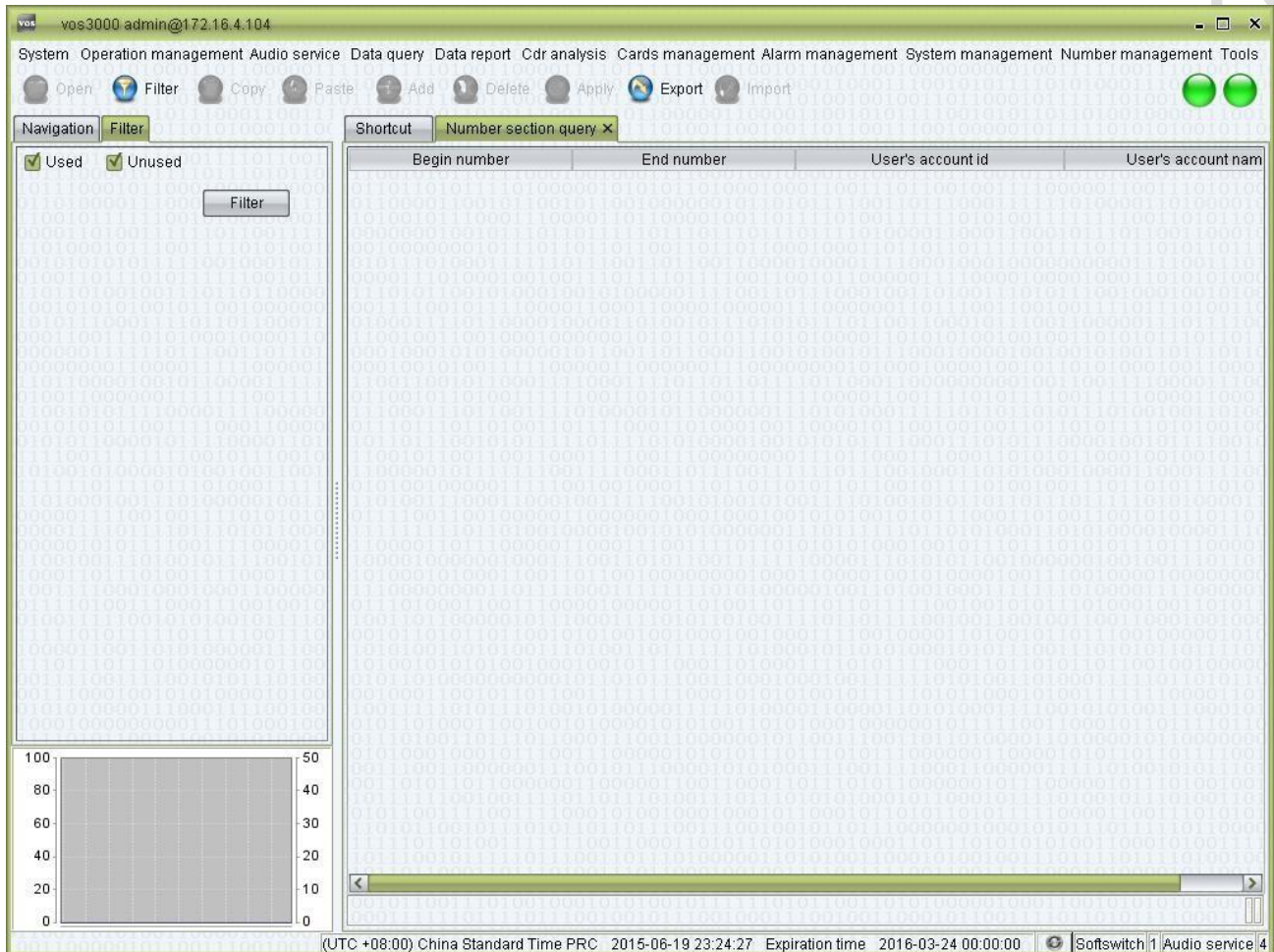
2.12.9 Server monitor

This function is used to monitor server.

2.13 Number Management

2.13.1 Number Section Query

This function is used to query number section.



How to Start

Double-click "Navigation > Number management > Number section query"

Table Items

Begin number

End number

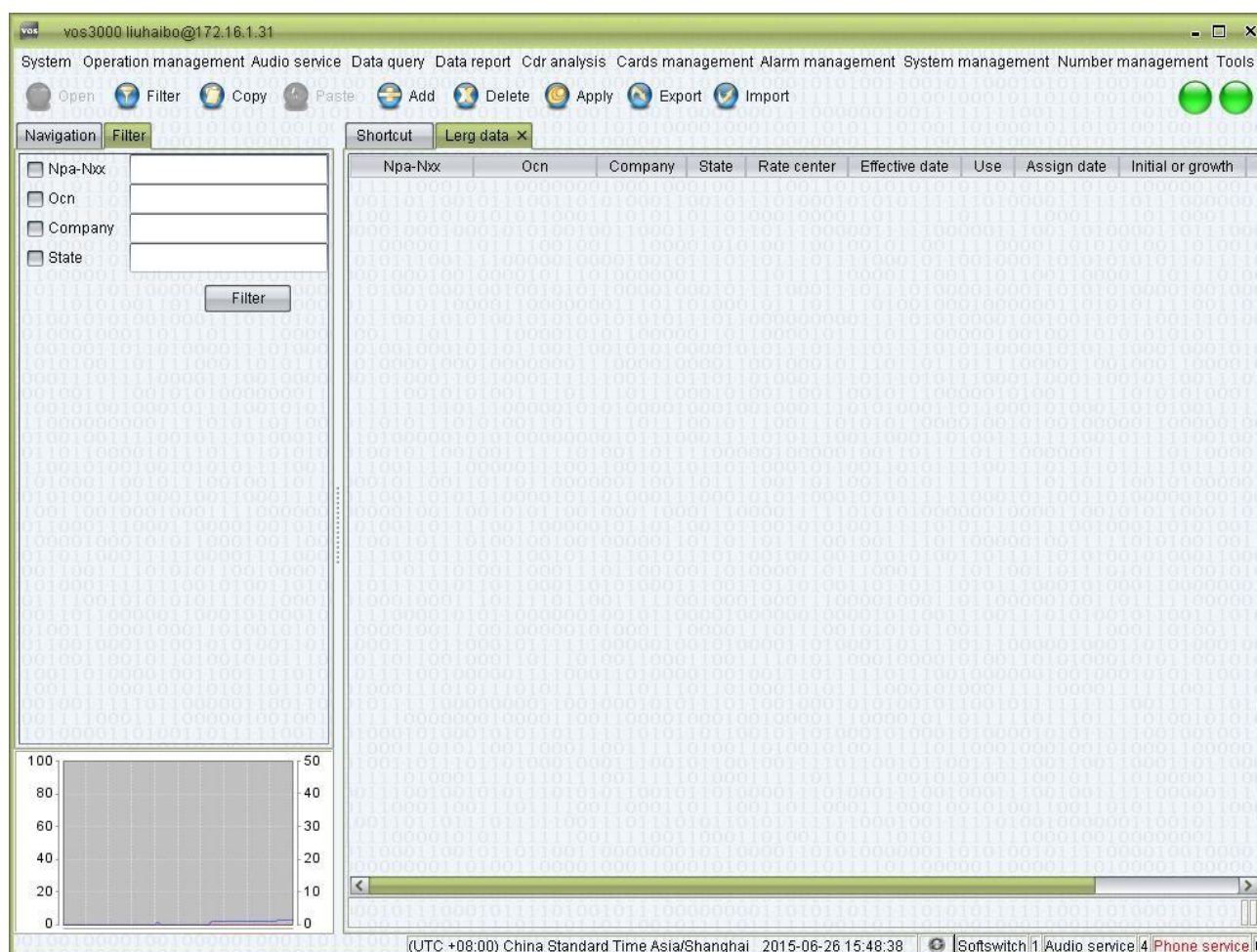
User's account id

User's account name

Downloaded from VOS3000.Com

2.13.2 Lerg Data

This function is used to distinguish interstate call, if caller and callee from different state, the call will be treat as interstate.



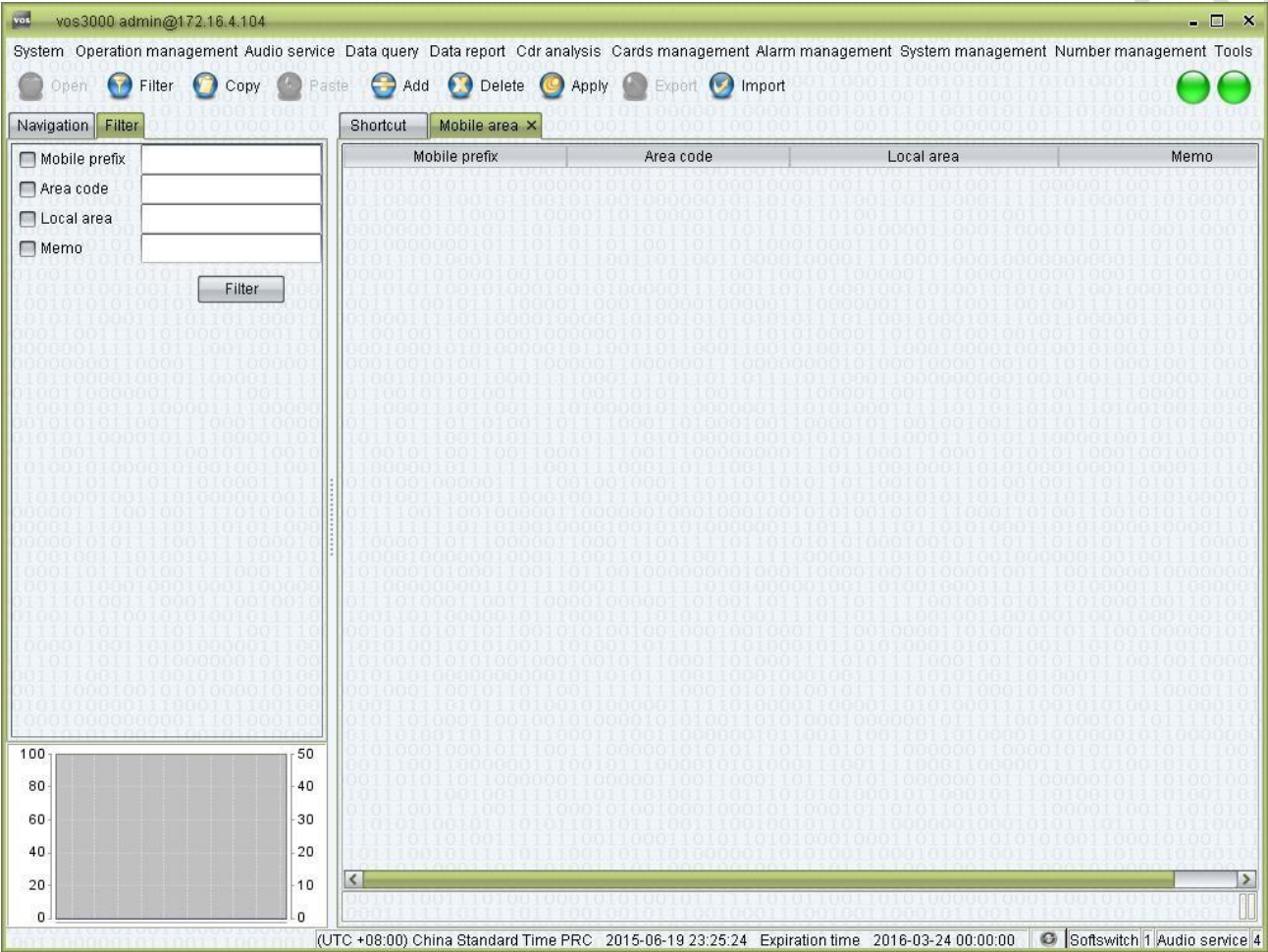
How to Start (VOS3000.Com)

Double-click "Navigation > Number management > Lerg data"

2.13.3 Mobile Area

This function is used to automatically add area code for mobile and make area limitation.

This setting will be used in routing settings of routing gateway, mapping gateway and phone.



How to Start

Double-click “Navigation > Number management > Mobile area”

Table Items

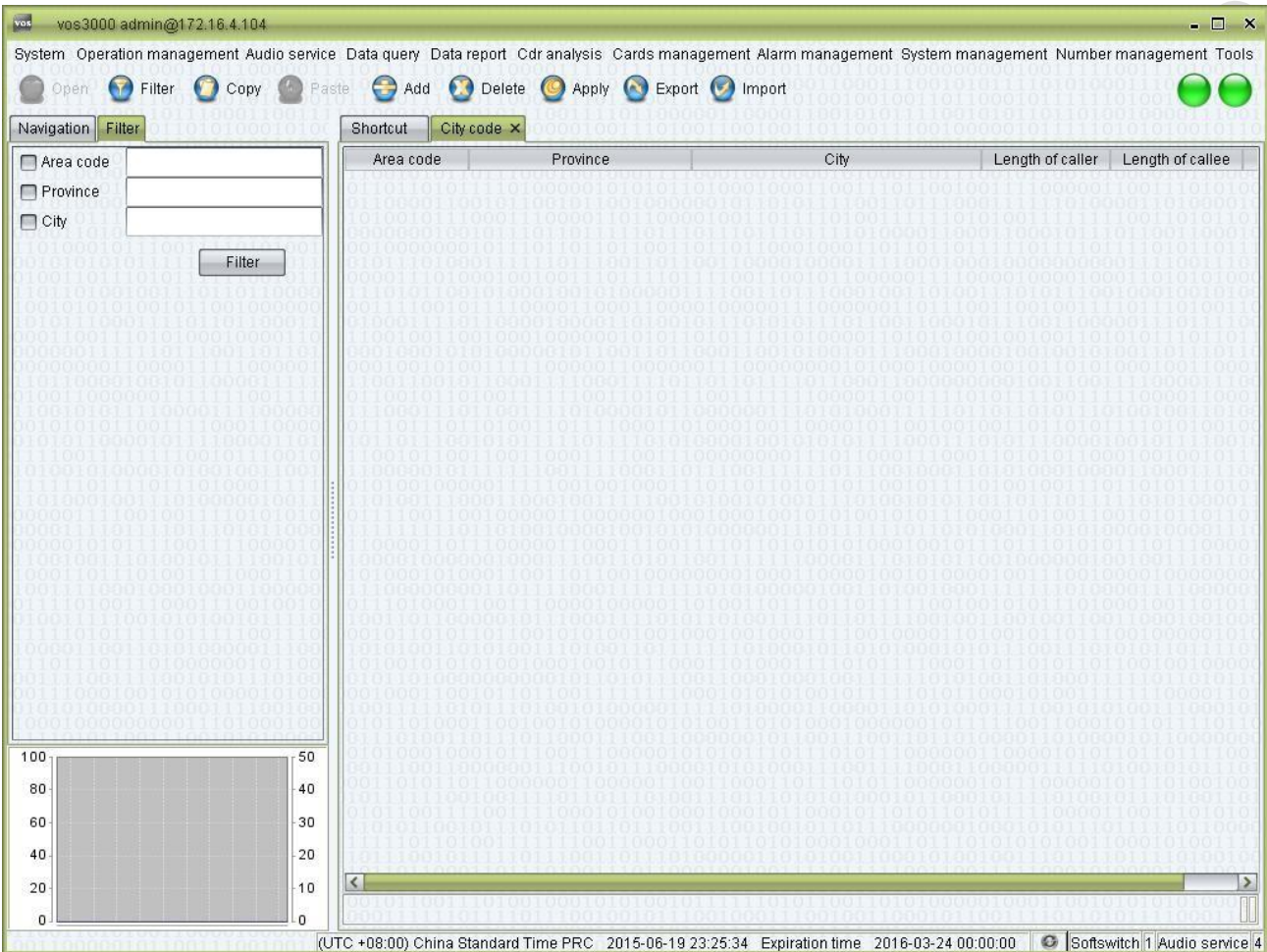
Mobile prefix
Area code
Local area
Memo

Downloaded from VOS3000.Com

2.13.4 City Code

This function is used to manage city code.

This setting will be used in Routing Settings of routing gateway, mapping gateway and phone.



How to Start

Double-click "Navigation > Number management > City code"

Table Items

Area code

Province

City

Length of caller: without area code, use comma to separate.

Length of callee: without area code, use comma to separate.

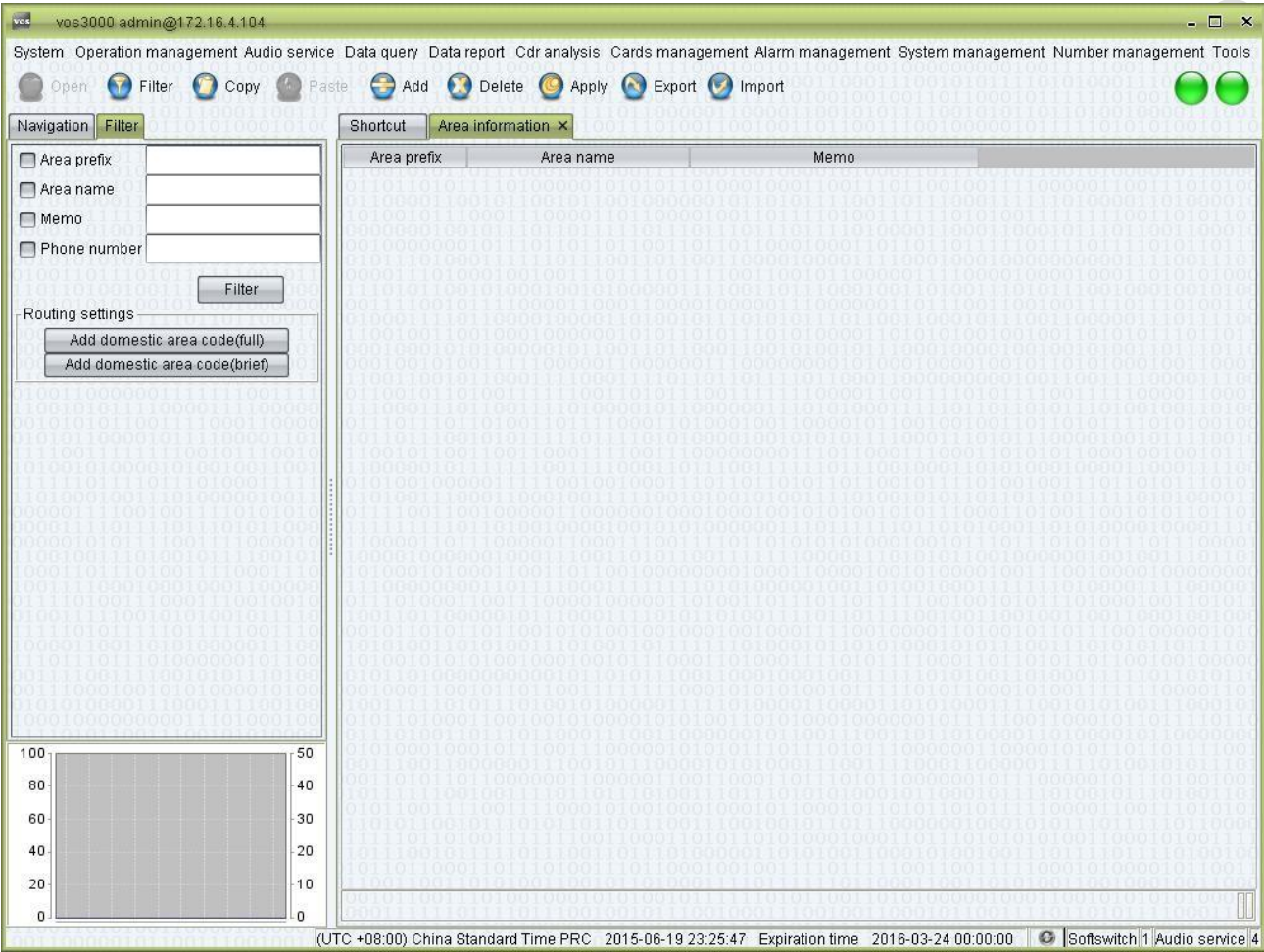
Area

Downloaded from VOS3000.Com

2.13.5 Area Information

This function is used to define area prefix's area name, country code, dialing prefix.

This setting will be used in rate management to show prefix's area name, using longest match.



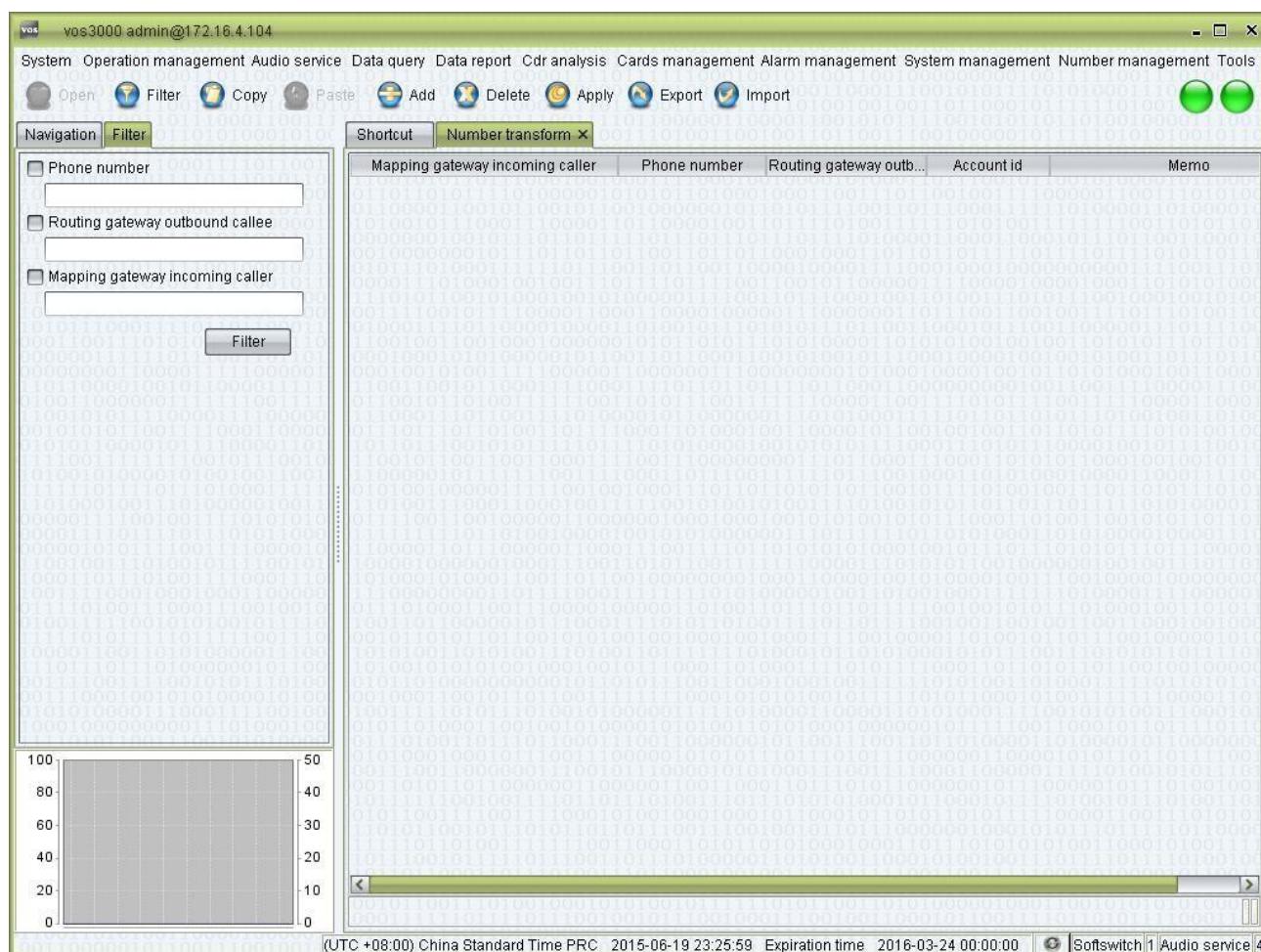
How to Start

Double-click “Navigation > Number management > Area information”

Table Items

- Area prefix
- Area name
- Memo

2.13.6 Number Transform



How to Start

Double-click "Navigation > Number management > Number transform"

2.13.7 Black/White List Group

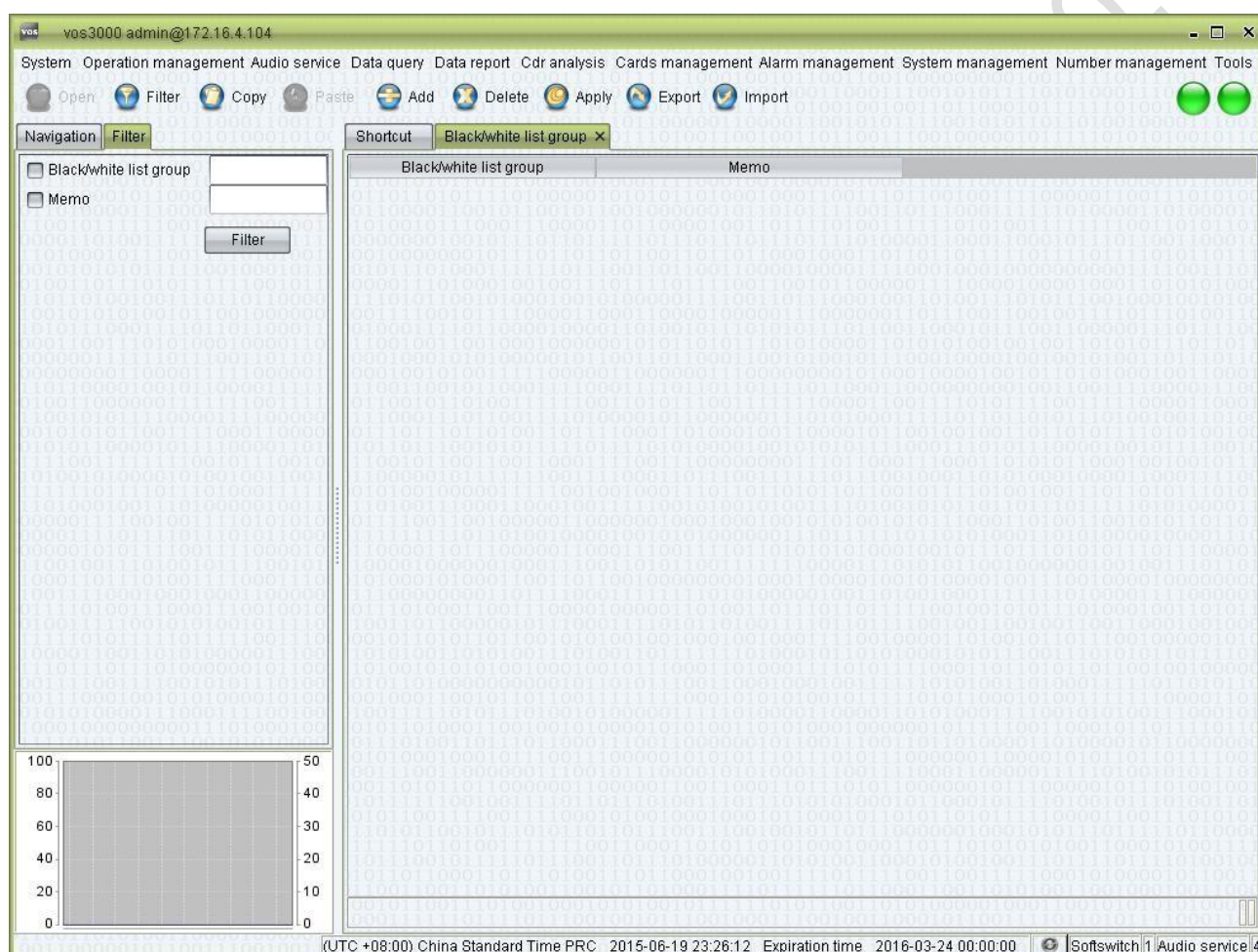
This function is used to define list group.

This setting will be used in caller/callee black/white list group of routing gateway, mapping gateway and phone.

TIP

Black/white list group is full match, more efficient than prefix match.

For a large number match, use this function instead of prefix match.



How to Start

Double-click "Navigation > Number management > Black/White List Group"

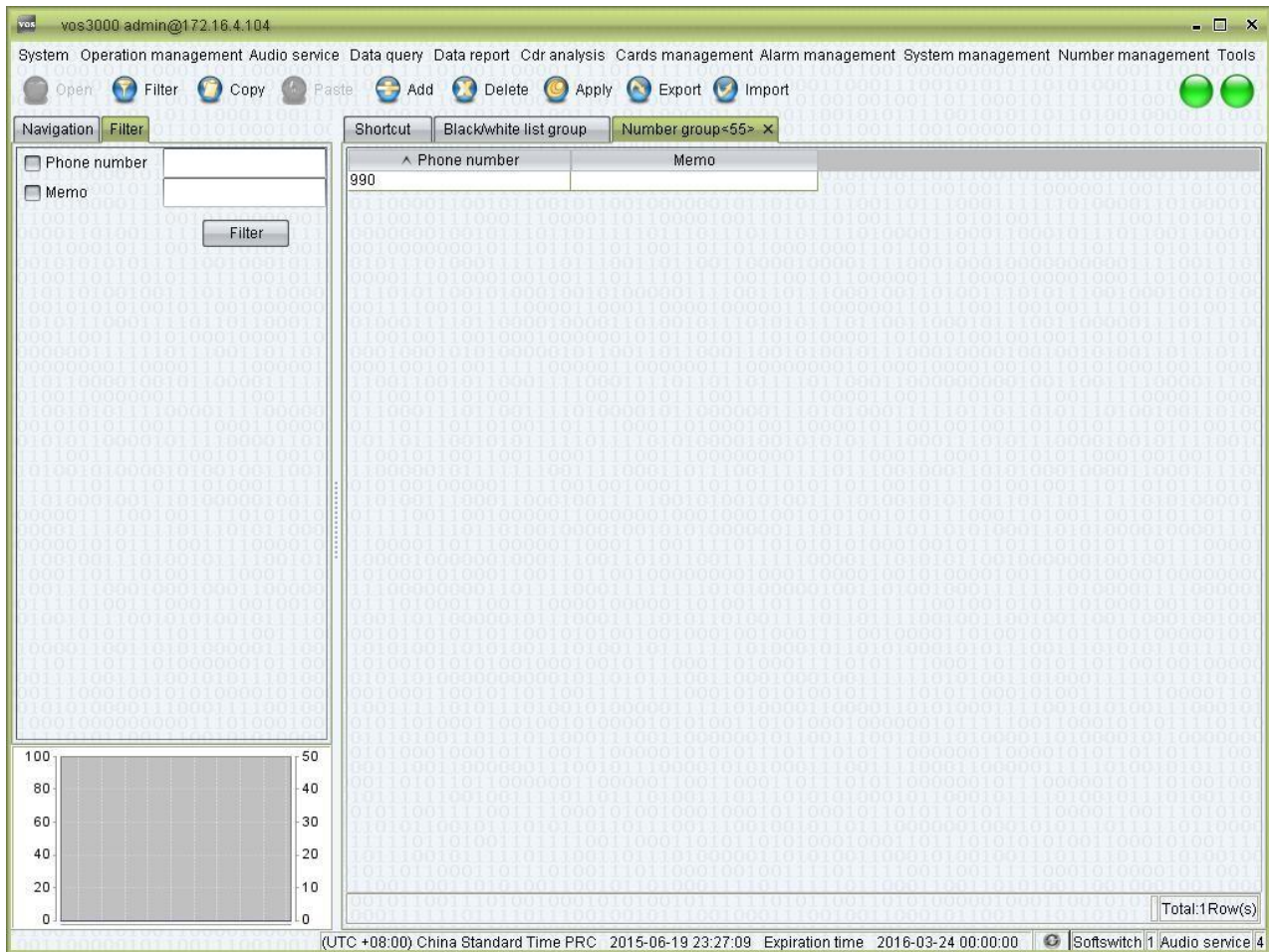
Table Items

Black/white list group: name of the group.

Memo

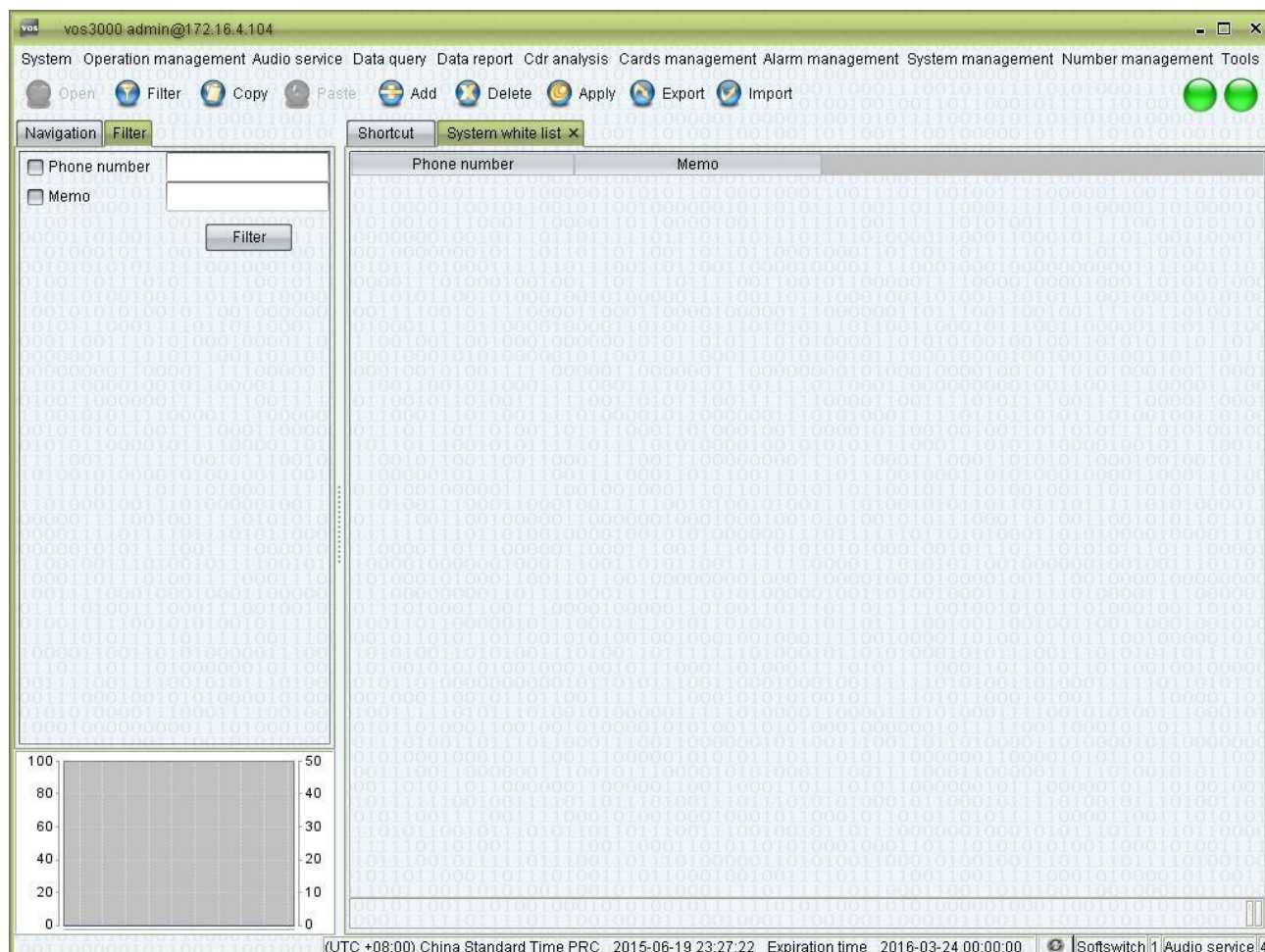
Other Operation

Double click black/white group to edit number list.



2.13.8 System White List

This function is used to manage system white list.



How to Start

Double-click "Navigation > Number management > System white list"

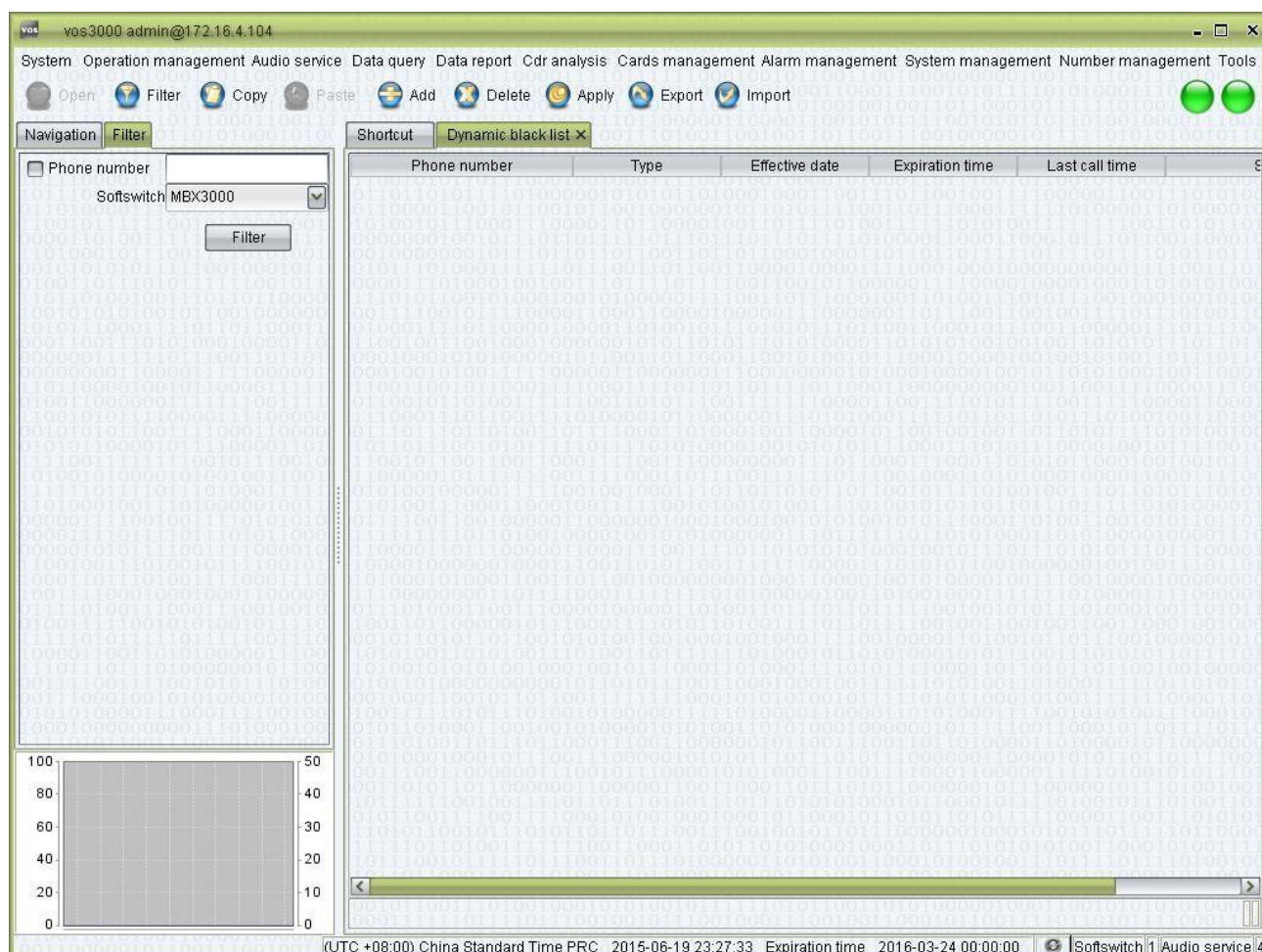
Table Items

Phone number

Memo

2.13.9 Dynamic Black List

This function is used to manage system black list.



How to Start

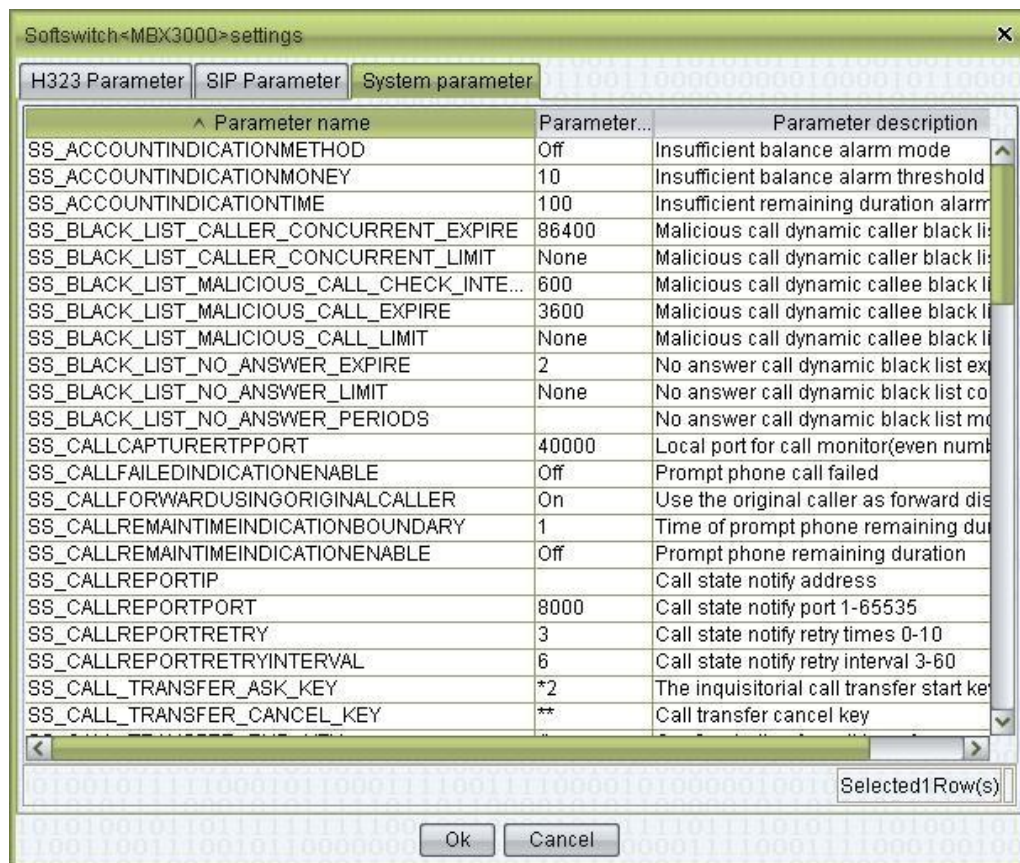
Double-click “Navigation > Number management > Dynamic black list”

Table Items

- Phone number
- Type: malicious call or no answer
- Effective date
- Expiration time
- Last call time
- Softswitch

Configure

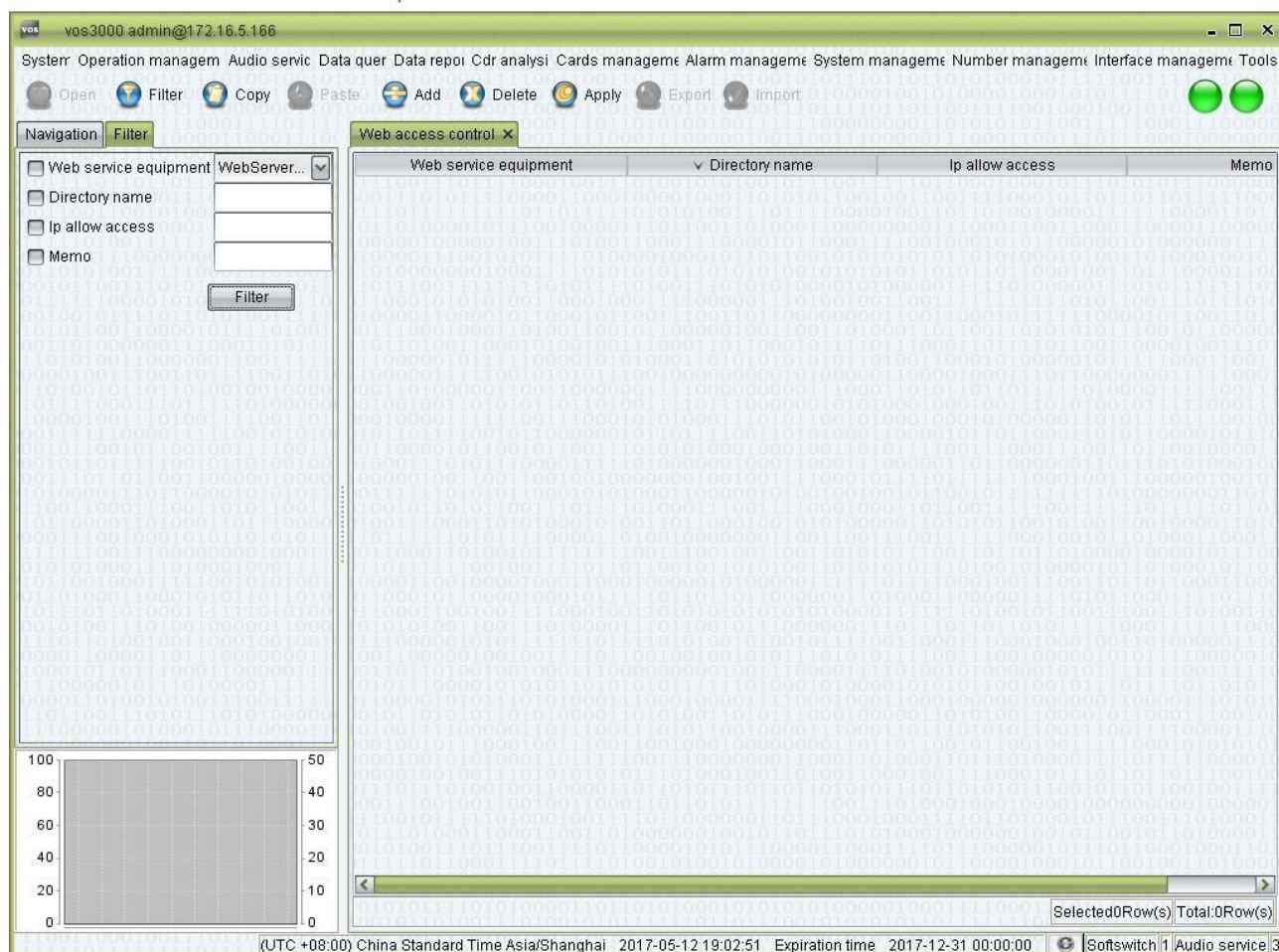
Double-click “Navigation > Operation management > Softswitch management > Additional settings > System parameter”



2.14 Interface Management

2.14.1 Web access control

This function is used to manage external access ip



How to Start

Double-click "Navigation > Interface management > web access control"

Table Items

Web service equipment

Directory name

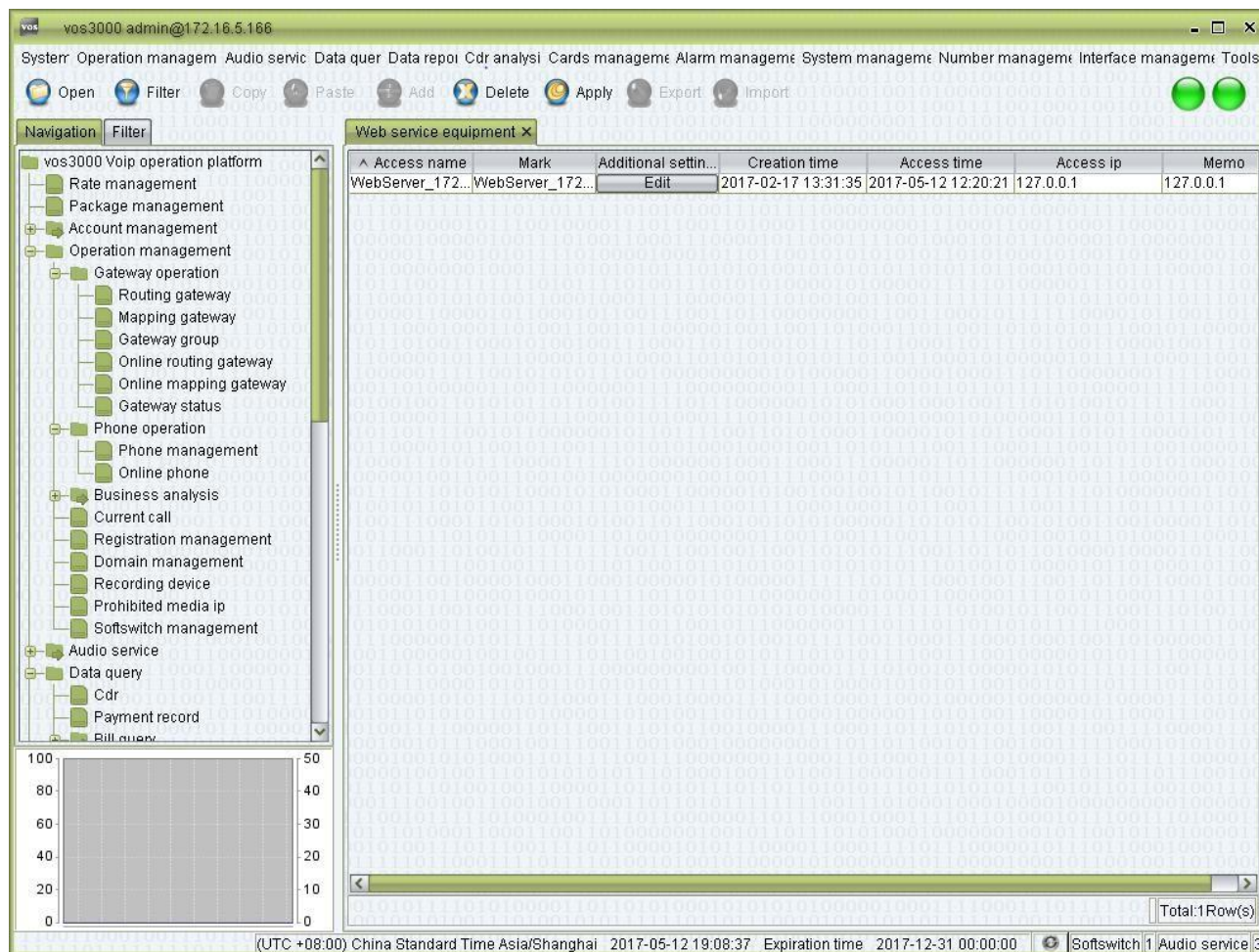
Ip allow access

Memo

Downloaded from VOS3000.Com

2.14.2 Web Service Equipment

This function is used to query web service equipment



How to Start

Double-click "Navigation > Interface management > web service equipment"

Table Items (VOS3000.Com)

Access name

Mark

Additional setting

Creation time

Access time

Access ip

Memo

2.15 Tools

2.15.1 Customer fee rate automatically create

This function is used to create fee rate automatically

Customer fee rate automatically create

Input fee rate

Filter

☐ Rate prefix ☐ Area prefix

☐ Rate type Net ☐ Area name

Base fee rate

Supplier fee rate

Desired profit 0.000 %

Generate fee rate

Customer fee rate

Compare files

Generate Off

How to Start

Double-click "Navigation >Tools> Customer fee rate automatically create"

Table Items

Rate prefix

Area Prefix

Rate type

Area name

Base fee rate:

Supplier fee rate:

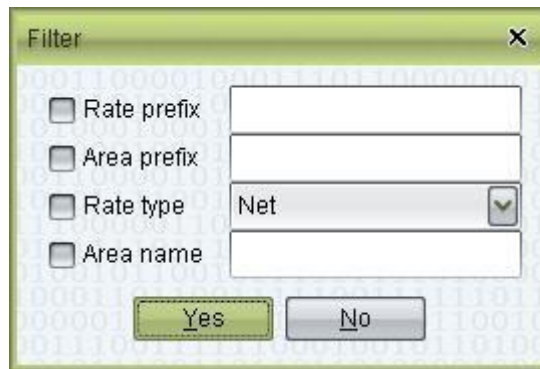
Desired profit

Customer fee rate

Compare files

2.15.2 Compare Fee Rate Group Minute Cost

This function is used to compare fee rate group



A dialog box titled "Filter" with a close button (X) in the top right corner. It contains four checkboxes on the left: "Rate prefix", "Area prefix", "Rate type", and "Area name". To the right of these checkboxes are four input fields. The "Rate type" field is a dropdown menu currently showing "Net". At the bottom of the dialog are two buttons: "Yes" and "No".

How to Start

Double-click "Navigation >Tools>compare fee rate group minute cost"

Table Items

Rate prefix

Area prefix

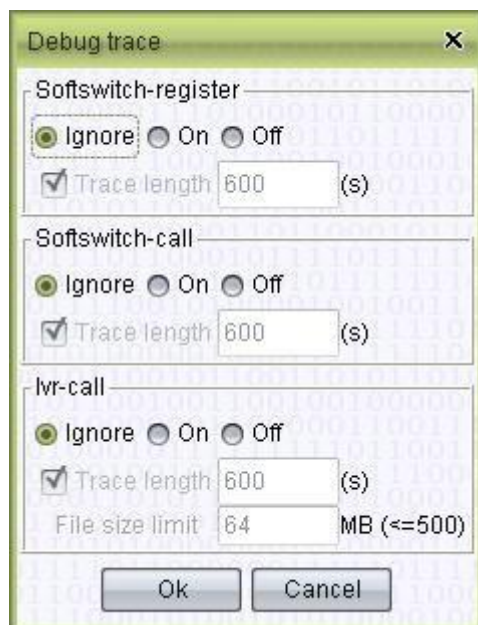
Rate type

Area name

2.16 Other Operation

2.16.1 Debug trace

This function is used to track call signaling.



How to Start

Click "System > Debug trace"

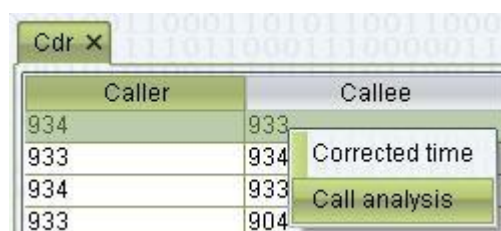
Usage

Check "On" to enable track, uncheck to disable.

Trace length: calculate from current time, stop track when timeout. Uncheck to track all the time.

File size limit: size of the signaling file. If over the limitation, file will be recovered.

After enable track, right click in "Current call" or "Cdr" to see call analysis.



 NOTE

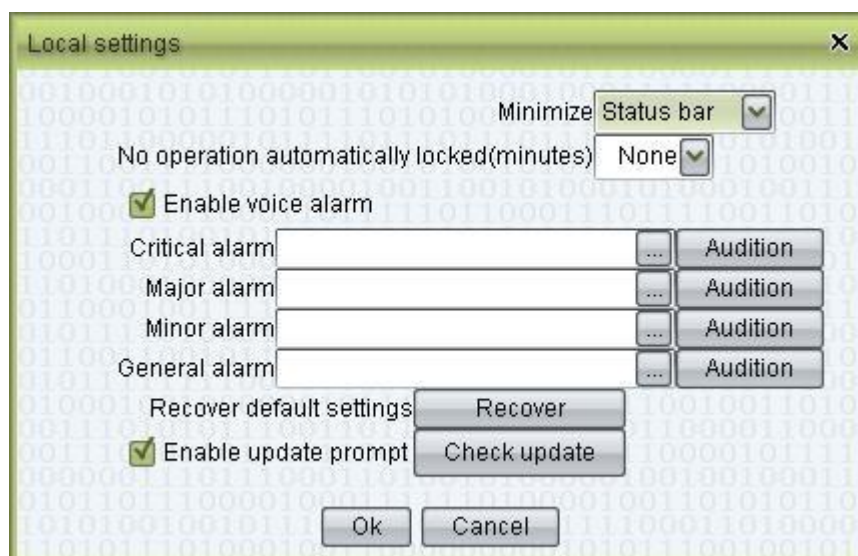
System uses 2 files to record track signaling, so the actual space will be double. If one file over the file size limit, system will use the other file.

All track signaling will be saved, unless file has been covered.

Downloaded from VOS3000.Com

2.16.2 Local Settings

This function is used to configure client settings.



How to Start

Click "System > Local settings"

Usage

Minimize: Status bar/System tray

No operation automatically locked

Enable voice alarm: customer alarm audios.

Recover default settings

Enable update prompt: click to enable.

2.16.3 Lock System

This function is used to hide client interface.



vos3000 Voip operation platform--Lock system

Server ip 172.16.4.104

User name admin

Password

Login Exit

www.linknat.com Product version vos3000.2.1.6.00

How to Start

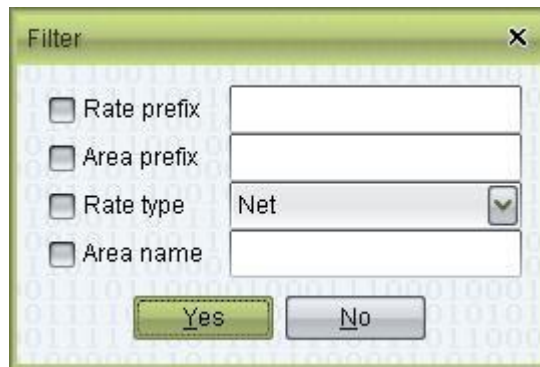
Click "System > Lock system"

2.16.4 Compare Rate

This function is used to compare minute rate.

How to Start

Click "Tools > Compare rate group minute cost"

A screenshot of a 'Filter' dialog box. It has a title bar with 'Filter' and a close button. Inside, there are four checkboxes: 'Rate prefix', 'Area prefix', 'Rate type', and 'Area name'. Each checkbox is followed by a text input field. The 'Rate type' field is a dropdown menu showing 'Net'. At the bottom, there are two buttons: 'Yes' and 'No'.

Filter

☐ Rate prefix

☐ Area prefix

☐ Rate type Net

☐ Area name

Yes No

Filters can be set before comparing.

2.16.5 Product Instruction

This function is used to show product instruction.

How to Start

Click "System > Product instruction"

Downloaded from VOS3000.Com

3

Supplement

About This Chapter

This chapter introduces the characteristics of VOS3000 interface.

3.1 Interface Description

3.1.1 Filters

The wildcard characters “*” and “?” can be used to specify filter criteria. For example, the filter criterion “800*” indicates all strings starting with “800”, and the filter criterion “888??00” represents all 7-digit strings that start with “888” and end with “00”.



NOTE

Use “*” and “?” to filter themselves, other characters have no transferred meaning.

The time of filter criteria in CDR and related spreadsheets can be specified according to either “Beginning of the call” or “End of the call”. When “Beginning of the call” is specified, then calls started in the specified time span will be matched. Otherwise, calls ended in the specified time span will be matched. Usually, operators use the time of the “End of the call” to classify calls.

3.1.2 Shortcuts

F5: Enable filtering

CTRL + C: Copy the selected table cells

ALT + F: Open “Rate management”

ALT + K: Open “Shortcuts”

ALT + S: Open "Package management"

ALT + D: Open "Mapping gateway"

ALT + G: Open "Routing gateway"

ALT + C: Open "Account management"

ALT + P: Open "Phone management"

ALT + A: Open "Current call"

ALT + H: Open "Cdr"

ALT + U: Open "User management"

ALT + L: Open "System log"

3.1.3 Cell Colors

Colors of Table Cell

White: Normal



: To be added after clicking the "apply" button



: To be modified after clicking the "apply" button



: To be deleted after clicking the "apply" button



: Selected



: Operation failed

Colors of Gateway Table Cell



: Bar outgoing/incoming calls



: Bar all calls



: Conflict IP address configurations for mapping gateways

Colors of Account Table Cell



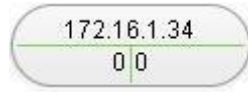
: Insufficient balance



: Account terminated

3.1.4 Floating Icon

users with the query authority except agent, will see the floating icon as following:



Top: server ip

Left bottom figure: shows current concurrency

Right bottom figure: shows queue length of cdr to be processed

3.2 System Description

3.2.1 Rewrite Rule

Original Prefix: prefix before rewrite.

Target Prefix: prefix after rewrite.

Table 3-1 Rewrite Rule Description

Original Prefix	Target Prefix	Number	Number after Rewrite	Memo
0	0	02584316146	02584316146	do not rewrite prefix 0
010	025	01012345678	02512345678	
025		02584316146	84316146	remove prefix 025
*	025*	117	025117	add prefix 025
025*	010	02584316146	010	change number with prefix 025 into 010
	010*	02584316146	01002584316146	add prefix 010
*		02584316146		delete all number

Original Prefix	Target Prefix	Number	Number after Rewrite	Memo
*	12345678	02584316146	12345678	change all number into 12345678
025*	8008100-8008121	02584316146	8008100 or 8008101 or...	change number into one of 8008100 to 8008121
*	12345?78	02584316146	12345178 or 12345278 or...	? will become a random number
*	12345678;8008100-8008121;12345?78	02584316146	12345678 or number between 8008100 and 8008121 or 12345?78	randomly choose one of the rules, then rewrite
0??8431	8431	02584316146	84316146	for original prefix, there can be any two digits between 0 and 8
0134	\$\$\$	0134131	013131	\$ means keep and not rewrite the position

Multiple substitution targets can be specified, separated by the symbol “;”.

When the symbol “?” exists in the substitution targets, each “?” will be replaced by randomly generated digit. The “?” can occur more than once in the target pattern.

When multiple Rewrite Rules exist, the longest matching pattern will be selected. For example, if there are two Rewrite Rules: one replaces “0” with “0”, while the other replaces “010” with “025”, then the number “01012345678” will be replaced with “02512345678” (since “010” is longer than “0”).

The pattern “*” has the lowest priority and will only be matched when there are no other matching patterns.

Escape character: L, E ,G

L: if character behind L is a number, means the number length will shorter than the specified value. For example: 13L9 represents all 11-digit numbers that start with 13 prefix

E: if character behind L is a number, the number length will equal the specified value.

For example: 010G7 represents all 10- digit numbers that start with 010 prefix

F: if character behind F is a number, the number length will longer than the specified value.

For example: 010G7 represents all 10- digit numbers that start with 010 prefix

Note : escape character with backslash before(/L,/E,/G) has no transferred meaning.

3.2.2 Media Proxy

Parameters below can be set in “Operation management > Softswitch management > Additional settings > System parameter”.

See SS_MEDIAPROXYMODE:

On: Enable Media Proxy.

Off: Disable Media Proxy.

Auto: decide by softswitch, see below.

Step 1 If caller or callee disabled media proxy, then disable.

Step 2 If caller or callee enabled media proxy, then enable.

Step 3 If callee enabled local ring, then enable.

Step 4 If phone and gateway use dynamic register and encrypt, then enable.

Step 5 If SS_MEDIAPROXYBETWEENNET is on, caller and callee from different network, then enable.

Step 6 If SS_MEDIAPROXYBEHINDNAT is on.

If phone and gateway in the same NAT, SS_MEDIAPROXYSAMENAT is on, then enable.

If phone and gateway in different NAT, one of them in private network, then enable.

Step 7 Finally, disable.

---End

3.2.3 Routing Gateway Sorting

Use callee to find routing gateway.

Parameters below, start with "SS_", can be set in "Operation management > Softswitch management > Additional settings > System parameter".

Parameters below, start with "SERVER_", can be set in "System management > System parameter".

Step 1 Match longest prefix

Step 2 Priority (the smaller the higher)

Step 3 If SS_GATEWAYASRROUTESORTCONFIG is Before Line Usage, then sort by route's ASR. Routings which disabled real-time computing ASR priority than enabled one.

If SS_GATEWAYFEERATEROUTESORTCONFIG is Before Line Usage, then sort by routing gateway's Lowest Rate per Second. Gateway which disabled Sort by Lowest Rate per Second is prior than enabled.

If the above 2 parameter values are the same, then sort by SS_GATEWAYFEERATEROUTEBEFOREASR.

Step 4 Sort by Line Usage

If SS_GATEWAYASRROUTESORTCONFIG is Before Current Day Total Call, then sort by route's ASR. Routings which disabled real-time computing ASR priority than enabled one.

If SS_GATEWAYFEERATEROUTESORTCONFIG is Before Current Day Total Call, then sort by routing gateway's Lowest Rate per Second. Gateway which disabled Sort by Lowest Rate per Second is prior than enabled.

If the above 2 parameter values are the same, then sort by
SS_GATEWAYFEERATEROUTEBEFOREASR.

Step 5 Sort by Current Day Total Call

If SS_GATEWAYASRROUTESORTCONFIG is Before Gateway ID, then sort by
route's ASR. Routings which disabled real-time computing ASR priority than enabled
one.

If SS_GATEWAYFEERATEROUTESORTCONFIG is Before Gateway ID, then sort
by routing gateway's Lowest Rate per Second. Gateway which disabled Sort by
Lowest Rate per Second is prior than enabled.

If the above 2 parameter values are the same, then sort by
SS_GATEWAYFEERATEROUTEBEFOREASR.

Step 6 Sort by Gateway ID

---End

 NOTE

If caller phone or mapping gateway set first routing strategy or second routing strategy, then
uses all available routing gateways, according to selected strategy.

3.2.4 Package Billing Order

Step 1 Use Free Duration.

See time span of the free duration, the longer the later.

See invalid time, the earlier the prior.

See effective date, the earlier the prior.

Step 2 Use Free Money Amount

See rent type, prior is "Day > Month > Year".

See invalid time, the earlier the prior.

See effective date, the earlier the prior.

Step 3 Use Account Money

---End

3.2.5 Parameter Description

3.2.5.2 VOS3000 Parameter

Parameter Name	Default Value	Parameter Description
EXTERNAL_WEB_SEND_PHONE_ONLINE	Off	Interface: phone online/offline transfer
EXTERNAL_MAX_CDR_PENDING_SIZE	10000	Interface: queue size of resend cdr 1000-100000
EXTERNAL_SEND_CDR	Off	Interface: send cdr
SERVER_ALARM_CUSTOMER_BALANCE_MAX_SIZE	1000	Number of accounts in “Navigation > Alarm management > Alarm settings > Balance alarm”
SERVER_ALARM_DATABASE_IGNORE_ERROR_CODE		Database error code without warning
SERVER_ALARM_DISABLE	Off	Off: enable alarm. On: disable alarm.
SERVER_ALARM_E164S		Default e164 setting in “Navigation > Alarm management > Alarm Setting”
SERVER_ALARM_EMAIL		Default email setting in “Navigation > Alarm management > Alarm Setting”
SERVER_ALARM_EMAIL_DELAY	300	Unit: second. Interval for email alarm.

Parameter Name	Default Value	Parameter Description
SERVER_ALARM_ENABLE_EMAIL	Off	Default email alarm setting in "Navigation > Alarm management > Alarm Setting"
SERVER_ALARM_ENABLE_VOICE	Off	Voice alarm
SERVER_BILLING_CUSTOMER_DEFAULT_FEE_RATE		Default fee rate group for HW interface
SERVER_BILLING_FEE_PRECISION	0.0000000	Billing money accuracy 0-1000
SERVER_BILLING_FEE_UNIT	0.0000000	Billing money unit 0-1000
SERVER_BILLING_FORWARD_PREFIX		Billing prefix for Call Transfer
SERVER_BILLING_FREE_E164S		Service Number for free, no time limit
SERVER_BILLING_FREE_TIME	0	Unit: second. Charged Duration will cut this value, but won't change conversation time.
SERVER_BILLING_GATEWAY_ROUTE_PREFIX		Routing gateway additional prefix, e.g. set 99,88 then if callee is 99123, will be charged as 123 in CDR

Parameter Name	Default Value	Parameter Description
SERVER_BILLING_HOLD_TIME_PRECISION	1000	Unit: million second. Time precision, e.g. set 50, if duration is 21.049s, then uses 21s, else if duration is 21.050, then uses 22s.
SERVER_BILLING_NO_CDR_E164S		See SERVER_BILLING_FREE_E164S, but these numbers won't create CDR
SERVER_BILLING_PREVENT_OVERDRAFT_ADVANCE_TIME	1	Account anti overdraft advance the number of minutes each time (minutes) 1-15
SERVER_BILLING_PROFIT_CALCULATE	<Call charges>Sub<Call expense>	Call profit calculation
SERVER_BILLING_RECORD_SERVER_HANG_UP	On	Valid when SERVER_BILLING_RECORD_ZERO_HOLD_TIME is on. If SERVER_BILLING_RECORD_SERVER_HANG_UP is off, hold time is 0 and hang up by server, the CDR won't be saved.
SERVER_BILLING_RECORD_ZERO_HOLD_TIME	On	If system is under attack, turn off this to reduce process load

Parameter Name	Default Value	Parameter Description
SERVER_BILLING_REC ORD_ILLEGAL_CALL	On	Illegal call: caller's IP cannot find in mapping gateways and caller's number cannot find in phones.
SERVER_BILLING_USE_ ROUTING_GATEWAY_P REFIX		
SERVER_CALL_HELPER _E164		Call helper number, can be used with SERVER_BILLING_FREE_E164S
SERVER_CDR_FILE_WR ITE_INTERVAL	None	Additional write call record file, the new file created time interval (seconds) 60-86400
SERVER_CDR_FILE_WR ITE_MAX	2048	Additional write call record file, the maximum number of reserved file 10-4096
SERVER_CDR_REAL_TI ME_REPORT_SERVER		Additional send call record to server address
SERVER_DATABASE_V ERSION		Version of data
SERVER_DISPLAY_CHA RT_GATEWAY_SIZE	10	Gateway number in "Gateway Performance" and "Period Connect analysis"
SERVER_DISPLAY_CUS TOMER_ALARM_MONEY	20	In "account management", if current balance below the value, color of the row will be changed

Parameter Name	Default Value	Parameter Description
SERVER_DISPLAY_MON EY_PRECISION	3	Money Precision, e.g. set value 3, then 1 will be shown as 1.000.
SERVER_DNS_UPDATE _INTERVAL	600	Unit: second. Used for "Domain management"
SERVER_GATEWAY_RO UTE_PREFIX		Routing additional routing prefix, separated by commas.
SERVER_IPPBX_DEFAU LT_LANGUAGE	chinese	Default language of ip pbx service
SERVER_LOGIN_FAILED _DISABLE_TIME	120	Time of disable user login when failed several times(seconds) 30-7200
SERVER_E164_NON_ST ANDARD_PREFIX		If caller is phone and SERVER_E164_INTERN ATIONAL_CALLEE_REB UILD is on, before routing, callee number will omit this prefix. This parameter won't change callee number.
SERVER_MAIL_CUSTO MERE_BILL	Off	SMTP Send Account Consumption Email Automatically. "Account management > Customer Information > Email"
SERVER_MAIL_REPORT _DAY_OF_WEEK	Monday	"Account management > Customer Information > Email > Report Sending Mode > Weekly"

Parameter Name	Default Value	Parameter Description
SERVER_MAIL_SMTP_FROM	127.0.0.1	Email sender's IP
SERVER_MAIL_SMTP_FROM_USER	Linknat Limiteddemo	Smtip sender
SERVER_MAIL_SMTP_SERVER	127.0.0.1	SMTP Server IP
SERVER_MAIL_SYSTEM_NOTICE		System notification email address
SERVER_MAX_CDR_PENDING_LIST_LENGTH	100000	Length of cdr queue limit 10000-100000
SERVER_MAX_INFO_ON_E_MESSAGE	10000	Max Send Number per time when Data Changed
SERVER_MGC_EQUIPMENT_IPS		Reliable equipment ip
SERVER_MGC_EXTERNAL_IPS		External interface device ip
SERVER_NTP_SERVER	time-a.nist.gov	Network time server (snip)
SERVER_PASSWORD_LENGTH	8	Default Length of Password
SERVER_PASSWORD_TERMINAL_ADDITIONAL_CHARACTERS		Additional characters for phone and gateway random passwords. Default: 0-9
SERVER_PAY_DELAY_CUSTOMER_EXPIRE_DAY	365	Unit: day. Extend the validity after recharge

Parameter Name	Default Value	Parameter Description
SERVER_PAY_NORMAL_EXPIRE_ACTIVE_DISCOUNT	0	Deduct Original Amount Percent, when Normal Recharge to Expired Account(%)
SERVER_PAY_NORMAL_EXPIRE_ACTIVE_FEE	0.0	Deduct Fee per Day, when Normal Recharge to Expired Account(Dollar)
SERVER_PAY_PHONE_CARD_EXPIRE_ACTIVE_DISCOUNT	0	Deduct Original Amount Percent, when use phone card to Recharge Expired Account(%)
SERVER_PAY_PHONE_CARD_EXPIRE_ACTIVE_FEE	0.0	Deduct Fee per Day, when use phone card to Recharge Expired Account(Dollar)
SERVER_PAY_PHONE_CARD_PAY_RELATE_TO_AGENT	Off	Charge Agent Accounts, which Use Lower Rate, at the same time, when customers use Phone Card to do charge.
SERVER_PHONE_AS_CALLEE_MONEY_VERIFY	on	Verify callee phone account balance
SERVER_PHONE_CARD_AUTO_UNBIND_HOUR_IN_DAY	3	Checking time for phonecard account automatic unbind 0--23
SERVER_PHONE_CARD_AUTO_UNBIND_REMAIN_MONEY	None	Remain money for phonecard automatic unbind
SERVER_PHONE_CARD_CONSUMPTION_PRECISION	0.0000000	Phone card billing precision 0-1000

Parameter Name	Default Value	Parameter Description
SERVER_PHONE_CARD_CONSUMPTION_UNIT	0.0000000	Phone card billing unit 0-1000
SERVER_PHONE_DEFAULT_LANGUAGE	1	Call Helper Default Language Code
SERVER_PSTN_PREFIX	0	PSTN Call Prefix
SERVER_QUERY_CDR_DENY_TIME		No CDR Query Time(24 hour) e.g. 18,19,20,21,22,23
SERVER_QUERY_CDR_MAX_DAY_INTERVAL	31	Maximum Interval for CDR Inquiry(Day)
SERVER_QUERY_MAX_ONE_PAGE_SIZE	200000	Maximum Number of Data per Page
SERVER_QUERY_MAX_SIZE	30000000	Data Query Limit(Item)
SERVER_QUERY_NON_PAGABLE_MAX_LINES	100000	Non payable table, maximum lines per page 1000-200000
SERVER_QUERY_ONE_PAGE_SIZE	10000	Number of Data per Page(Item)
SERVER_REPORT_AGENT_INCOME	On	Automatically generates agent income report
SERVER_REPORT_CLEARING_CUSTOMER_FEE	Off	Auto Generate Clearing Account Details Report
SERVER_REPORT_CLEARING_CUSTOMER_IO	Off	Auto Generate Account Clearing Balance Report
SERVER_REPORT_CLEARING_CUSTOMER_LOCATION_FEE	Off	Auto Generate Clearing-Account Area Details Report

Parameter Name	Default Value	Parameter Description
SERVER_REPORT_CLEARING_GATEWAY_FEE	Off	Auto Generate Clearing Gateway Details Report
SERVER_REPORT_CUSTOMER_FEE	On	Auto Generate Revenue Details Report
SERVER_REPORT_CUSTOMER_IO	Off	Auto Generate Account Balance Report
SERVER_REPORT_CUSTOMER_LOCATION_FEE	On	Auto Generate Account Area Detail Report
SERVER_REPORT_GATEWAY_CROSS_LOCATION_ASR_ACD	Off	Automatically generate gateway cross area analysis report
SERVER_REPORT_GATEWAY_FEE	On	Auto Generate Gateway Bill Report
SERVER_REPORT_GATEWAY_MAPPING_ASR_ACD	Off	Auto Generate Mapping Gateway Connect Analysis Report
SERVER_REPORT_GATEWAY_MAPPING_LOCATION_ASR_ACD	On	Automatically generate mapping gateway area analysis report
SERVER_REPORT_GATEWAY_ROUTING_ASR_ACD	Off	Auto Generate Routing Gateway Connect Analysis Report
SERVER_REPORT_GATEWAY_ROUTING_LOCATION_ASR_ACD	On	Automatically generate routing gateway area analysis report
SERVER_REPORT_PHONE_CARD_E164_FEE	On	Automatically generate bind number bill report
SERVER_REPORT_PHONE_CARD_FEE	On	Automatically generate phone card bill report

Parameter Name	Default Value	Parameter Description
SERVER_REPORT_PHONE_FEE	On	Auto Generate Phone Bill Report
SERVER_REPORT_TIME_ZONE_LOWER		Minimum accounts time zone(milliseconds)
SERVER_REPORT_TIME_ZONE_UPPER		Maximum accounts time zone(milliseconds)
SERVER_SUPPORT_EXPIRE_NOTIFY	Off	Technical support expire reminder
SERVER_SIP_LOAD_BALANCE_SERVER		Load Balancing Server IP
SERVER_SMAP_RESERVE_TIME	None	SMAP Reserve Duration (seconds) 60-86400 None: disable.
SERVER_SOFTSWITCH_CLUSTER		IP List of Softswitch Cluster
SERVER_SOFTSWITCH_ENDPOINT_EXPIRE	3600	Unit: second. Terminal Registration Expiry Time
SERVER_SOFTSWITCH_ENDPOINT_NAT_EXPIRE	120	Unit: second. Terminal Registration Expiry Time
SERVER_SUPPORT_EXPIRE_NOTIFY	On	Technical Support Expire Reminder
SERVER_TRACE_FILE_LENGTH	40960	Size of Debug file
SERVER_TRACE_MASK	ERROR	Set Display of Debug Information

Parameter Name	Default Value	Parameter Description
SERVER_TRACE_TO_FILE	On	Output Debug Information into File
SERVER_VERIFY_CLEARING_CUSTOMER	Off	Check Callee's Phone Account On: clearing account's balance must over SERVER_VERIFY_CLEARING_CUSTOMER_REMAIN_MONEY_LIMIT.
SERVER_VERIFY_CLEARING_CUSTOMER_REMAIN_MONEY_LIMIT	0.0	Clearing Account Remain Money Limit 0-10000000
SERVER_VERIFY_CLEARING_CUSTOMER_TIME	Off	If SERVER_VERIFY_CLEARING_CUSTOMER is on, Check Clearing Account Available Time
WEB_PHONEBOOKCALLBACKACCESSNUMBER_CARD		Access Number for Web Directory billing by Phone Card
WEB_PHONEBOOKCALLBACKACCESSNUMBER_PHONE		Access Number for Web Directory billing by Phone Number

3.2.5.3 Softswitch Parameter

Table 3-2 H323 Parameter

Parameter Name	Default Value	Parameter Description
SS_H245_PORT_RANGE	10000,39999	H245 port range
SS_H323_DTMF_METHOD	H.245 alphanumeric	Default DTMF send mode

Parameter Name	Default Value	Parameter Description
SS_H323_NUMBERING_PLAN	UnknownPlan(0)	Default value in “Routing Gateway > Additional settings > Protocol > H323”
SS_H323_NUMBER_TYPE	UnknownType(0)	Default value in “Routing Gateway > Additional settings > Protocol > H323”
SS_H323_PROGRESS_INDICATOR	ProgressInbandInformationAvailable(8)	Default value in “Mapping Gateway > Additional settings > Protocol > H323”
SS_H323_SCREENING_INDICATOR	None	Default value in “Routing Gateway > Additional settings > Protocol > H323”
SS_H323_STOP_SWITCH_AFTER_OLC	Off	Default value in “Routing Gateway > Additional settings > Protocol > H323”
SS_H323_TIMEOUT_ALERTING	120	Unit: second. Default value in “Routing Gateway > Additional settings > Protocol > H323”
SS_H323_TIMEOUT_CALLPROCEEDING	20	Unit: second. Default value in “Routing Gateway > Additional settings > Protocol > H323”

Parameter Name	Default Value	Parameter Description
SS_H323_TIMEOUT_CAL LPROCEEDING_OLC	20	Unit: second. Default value in “Routing Gateway > Additional settings > Protocol > H323”
SS_H323_TIMEOUT_SET UP	5	Unit: second. Default value in “Routing Gateway > Additional settings > Protocol > H323”

Table 3-3 SIP Parameter

Parameter Name	Default Value	Parameter Description
SS_SIP_AUTHENTICATI ON_CODE	Unauthorized(401)	Return code for SIP authentication, when Register message or Invite message without Proxy-Authorization or Authorization.
SS_SIP_AUTHENTICATI ON_RETRY	6	SIP authentication retry time, when received 401 or 407
SS_SIP_AUTHENTICATI ON_TIMEOUT	10	Time for SIP Authentication If caller failed to get authentication within the time, Softswitch will reject the call.
SS_SIP_E164_DISPLAY_ FROM	Ignore	Mode of sip display information

Parameter Name	Default Value	Parameter Description
SS_SIP_NAT_KEEP_ALIVE_MESSAGE	HELLO	Content of NAT Keep Message
SS_SIP_NAT_KEEP_ALIVE_PERIOD	30	NAT Keep Message's Period(seconds) 10-86400
SS_SIP_NAT_KEEP_ALIVE_SEND_INTERVAL	500	NAT Keep Message's Send Interval(million seconds)
SS_SIP_NAT_KEEP_ALIVE_SEND_ONE_TIME	3000	NAT Keep Message's Quantity per Time
SS_SIP_NO_TIMER_REINVITE_INTERVAL	7200	Maximum Conversation Time for Non-TIMER SIP Caller If SIP caller doesn't support "timer", softswitch will stop the call when the time is up.
SS_SIP_PUBLISH_EXPIRE	300	Routing gateway public update timeout default duration (seconds) 30-7200
SS_SIP_RESEND_INTERVAL	0.5,1,2,4,4,4,4,4,4	Resend SIP Message Interval (Second) If got no response or confirm within the time, Softswitch will resend SIP message. If exceeded the retry times, Softswitch will stop sending and regard as call failure, then try another gateway or hang up.

Parameter Name	Default Value	Parameter Description
SS_SIP_ROUTING_GATEWAY_INVITE_USE_CONTACT	Off	Use number from request-line as callee and keep original number in To field when send invite to callee
SS_SIP_SESSION_TIMEOUT_EARLY_HANGUP	0	SIP Timer no reinvite (update) Early Hang up(Second)
SS_SIP_SESSION_TTL	600	Detecting SIP Connected Status Interval(Second) If SIP caller supports "session-timer", within the time Softswitch will detect connect status according to the retry times. If got no confirm message, Softswitch will regard as call finish, then hang up.
SS_SIP_SESSION_UPDATE_SEGMENT	2	SIP Timer reinvite (update) Interval 2--10
SS_SIP_STOP_SWITCH_AFTER_SDP	On	Stop Switch Gateway After Receive SDP
SS_SIP_TIMEOUT_INVITE	10	Unit: second. Default value in "Routing Gateway > Additional settings > Protocol > SIP"
SS_SIP_TIMEOUT_RINGING	120	Unit: second. Default value in "Routing Gateway > Additional settings > Protocol > SIP"

Parameter Name	Default Value	Parameter Description
SS_SIP_TIMEOUT_SESSION_PROGRESS	20	Unit: second. Default value in “Routing Gateway > Additional settings > Protocol > SIP”
SS_SIP_TIMEOUT_SESSION_PROGRESS_SDP	120	Unit: second. Default value in “Routing Gateway > Additional settings > Protocol > SIP”
SS_SIP_TIMEOUT_TTY_RING	20	Unit: second. Default value in “Routing Gateway > Additional settings > Protocol > SIP”
SS_SIP_USER_AGENT_EXPIRE	Auto Negotiation	SIP Registration Expiration Time to Other Server(seconds) 20-7200
SS_SIP_USER_AGENT_PRIVACY	Ignore	Privacy Setting for Register User Ignore: No Privacy field Id: contain Privacy: id None: contain Privacy: none
SS_SIP_USER_AGENT_RETRY_DELAY	60	Resend Interval for SIP Registration when Failed(Second) 30--600
SS_SIP_USER_AGENT_SEND_UNREGISTER	On	Send Cancel Register Message
SS_SIP_USER_AGENT_STOP_SWITCH_AFTER_INVITE_TIMEOUT	Off	Stop Switch Gateway After INVITE Timeout

Table 3-4 System Parameter

Parameter Name	Default Value	Parameter Description
SS_ACCOUNT_INDICATION_METHOD	Off	Insufficient balance alarm mode
SS_ACCOUNT_INDICATION_MONEY	10	Insufficient balance alarm threshold
SS_ACCOUNT_INDICATION_TIME	100	Insufficient remaining duration alarm threshold(minutes)
SS_AUTHENTICATION_FAILED_SUSPEND	180	Disable duration after exceed max terminal password authentication retry times(seconds) 60-3600
SS_AUTHENTICATION_MAX_RETRY	6	Max terminal password authentication retry times 0-999
SS_BLACK_LIST_CALLER_CONCURRENT_EXPIRE	86400	Malicious call dynamic caller black list expired duration(seconds)
SS_BLACK_LIST_CALLER_CONCURRENT_LIMIT	None	Malicious call dynamic caller black list concurrency limit
SS_BLACK_LIST_CALLER_MALICIOUS_CALL_LIMIT	None	Malicious call dynamic caller black list max call times
SS_BLACK_LIST_CALLER_MALICIOUS_CALL_CHECK_INTERVAL	600	Malicious call dynamic caller black list monitor cycle(seconds)

Parameter Name	Default Value	Parameter Description
SS_BLACK_LIST_CALLER_MALICIOUS_CALL_EXPIRE	3600	Malicious call dynamic caller black list expired duration(seconds)
SS_BLACK_LIST_MALICIOUS_CALL_CHECK_INTERVAL	600	Malicious call dynamic callee black list monitor cycle(seconds)
SS_BLACK_LIST_MALICIOUS_CALL_EXPIRE	3600	Malicious call dynamic callee black list expired duration(seconds)
SS_BLACK_LIST_MALICIOUS_CALL_LIMIT	None	Malicious call dynamic callee black list max call times
SS_BLACK_LIST_NO_ANSWER_EXPIRE	2	No answer call dynamic black list expired duration(days)
SS_BLACK_LIST_NO_ANSWER_LIMIT	None	No answer call dynamic black list continuous call times
SS_BLACK_LIST_NO_ANSWER_PERIODS		No answer call dynamic black list monitor period
SS_CALLCAPTURERTP_PORT	40000	Local Port for Call Monitor(Even Number)
SS_CALL_FAILED_INDICATION	None	Prompt phone call failed
SS_CALL_FORWARD_USING_ORIGINAL_CALLER	On	Use the Original Caller as Forward Display Number
SS_CALL_HOLD_KEY	*9	Call hold key
SS_CALL_PICK_UP_KEY	*4	Call pickup key

Parameter Name	Default Value	Parameter Description
SS_CALL_REMAIN_TIME INDICATION_BOUNDAR Y	1	Time of prompt phone remaining duration (minutes). During the call, if remain time less than the value, system will give prompt to both side.
SS_CALL_REMAIN_TIME _INDICATION	Off	Prompt phone remaining duration
SS_CALL_REPORT_IP		Call state notify address
SS_CALL_REPORT_POR T	8000	Call state notify port
SS_CALL_REPORT_RET RY	3	Call state notify retry times
SS_CALL_REPORT_RET RY_INTERVAL	6	Call state notify retry interval
SS_CALL_SERVICE_CA NCEL_KEY	***	Call transfer cancel key
SS_CALL_STATE_REPO RT	Off	Http call status notification
SS_CALL_TRANSFER_A SK_DISPLAY	Original caller	Ask call transfer display
SS_CALL_TRANSFER_A SK_KEY	*2	The inquisitorial call transfer start key
SS_CALL_TRANSFER_C ANCEL_KEY	**	Call transfer cancel key
SS_CALL_TRANSFER_E ND_KEY	#	Confirm button for call transfer
SS_CALL_TRANSFER_N ORMAL_DISPLAY		

Parameter Name	Default Value	Parameter Description
SS_CALL_TRANSFER_NORMAL_KEY	*1	Start button for call transfer
SS_CALL_TRANSFER_REMOTE_RING_PASSTHROUGH	On	Send color ring back tone when call transfer
SS_CALL_TRANSFER_WAIT_ACCESS_KEY	*3	The pickup call transfer start key
SS_CALL_TRANSFER_WAIT_ACCESS_TIMEOUT	120	Pickup call transfer timeout (seconds)
SS_CDR_RECORD_ILLEGAL	On	Record illegal call
SS_CDR_RECORD_NONCONNECT	Off	When saving CDR as TXT, contains CDR, which hold time is 0s
SS_CDR_RECORD_TO_FILE	Off	Save CDR as TXT
SS_DEFAULT_LOCAL_IP		Default Local Address
SS_DTMF_TIMEOUT	5	Waiting time for Automatic Call Transfer
SS_ENDPOINT_EXPIRE	300	Terminal register expire time
SS_ENDPOINT_REGISTRER_REPLACE	On	Allow replace the current registered users when terminal registration.
SS_ENDPOINT_EXPIRE	3600	Terminal registration expiry time(seconds) 60-86400

Parameter Name	Default Value	Parameter Description
SS_ENDPOINT_NAT_EXPIRE	300	Terminal registration expiry time(seconds) 60-86400
SS_ENDPOINT_REGISTRER_REPLACE	On	Allow replace the current registered users when terminal registration
SS_ENDPOINTREGISTER_RETRY	6	Max retry times when terminal registration
SS_EXTERNAL_REWRITE_TYPE	None	External rewrite method
SS_ENDPOINTREGISTER_SUSPEND	180	Disable duration after exceed retry times for when terminal registration
SS_ENDPOINTTIME_TOLIVE	60	Interval for Lightweight Registration of Terminal(seconds)
SS_GATEWAY_ACD_CALCULATE	Off	External rewrite method
SS_GATEWAY_ACD_SERVE_SEPARATE	10	Section for gateway's acd routing(calculated as the step size) 5-24
SS_GATEWAY_ACD_SERVE_TIME	600	Length for gateway's acd routing(seconds) 300-86400
SS_GATEWAY_ASR_CALCULATE	Off	Real time computing asr
SS_GATEWAY_ASR_SERVE_SEPARATE	10	Section for gateway's asr routing(calculated as the step size) 5-24

Parameter Name	Default Value	Parameter Description
SS_GATEWAY_ASR_RE SERVE_TIME	600	Length for gateway's asr routing(seconds) 300-86400
SS_GATEWAY_ASR_RO UTE_SORT_CONFIG	Before line usage	Position for routing gateway's asr routing
SS_GATEWAY_FEE_RA TE_ROUTE_BEFORE_AS R	Off	Rate routing priority over asr routing
SS_GATEWAY_FEE_RA TE_ROUTE_SORT_CON FIG	Before line usage	Position for routing gateway's rate routing
SS_GATEWAY_SWITCH _LIMIT	None	Times limit for Routing Gateway Auto-Switch
SS_GATEWAY_SWITCH _STOP_AFTER_RTP_ST ART	On	Stop Switch Gateway when RTP Start
SS_GATEWAY_SWITCH _STOP_AFTER_USER_B USY	On	Callee busy stop switch
SS_GATEWAY_SWITCH _UNTIL_CONNECT	Off	Switch Gateway Until Connect
SS_GC_INTERVAL	3600	Interval for Clean Useless Data of Softswitch(seconds)
SS_LOCAL_RING_DEFA ULT	localring	Default local ring
SS_LRN_SERVER_IP		Lrn server ip
SS_LRN_SERVER_PORT	5060	Lrn server port 1-65535

Parameter Name	Default Value	Parameter Description
SS_MAPPING_GATEWAY_TIMEOUT	None	Mapping Gateway Default Timeout
SS_MAX_CALL_DURATION	None	Max Conversation Limitation(seconds)
SS_MEDIA_CHECK_TIMEOUT	2	No Media Data Hang Up Duration(minutes)
SS_MEDIA_PROXY_BEHIND_NAT	On	Forward RTP for Registered Terminals behind NAT
SS_MEDIA_PROXY_BETWEEN_NET	Off	Forward RTP for Calls between Different Networks
SS_MEDIA_PROXY_MODE	Auto	Media Proxy
SS_MEDIA_PROXY_PORT_RANGE	10000,39999	RTP Port for Forwarding Voice, use Comma to Separate the Start and End port
SS_MEDIA_PROXY_SAME_NAT	On	Forward RTP for Registered Terminals in the same NAT
SS_MOBILE_E164_LENGTH	11	Length limit for Mobile Number separated by commas (Maximum 31), uses 0 for allow all the length
SS_NON_STANDARD_PREFIX		Non standard e164 prefix, separated by commas.
SS_NO_BILLING_TO_PHONE	Off	Free billing to phone

Parameter Name	Default Value	Parameter Description
SS_PHONE_ONLINE_REPORT	Off	Interface: phone online/offline notice
SS_PHONE_SERVICE_IP		Phone service ip
SS_PHONE_SERVICE_PORT	5062	Phone service port 1-65535
SS_REDIRECT_OFFLINE_PHONE_TO_GATEWAY	Off	If phone is offline, try to use routing gateway for routing.
SS_REDIRECT_SERVER		External billing and routing control redirect server address (Sip 3xx)
SS_REDIRECT_SERVER_AVAILABLE_TO_PHONE	Off	External billing and routing control also take effect when callee is phone
SS_REPLY_UNAUTHORIZED	On	Respond to Unauthorized Registration or Call
SS_RTP_ENCRYPT_V2	XOR,RC4,AES128	Version 2 rtp encryption algorithm (XOR,RC4,AES128)
SS_SELF_SERVICE_URL	http://\$ipaddress/\$language/directlogin.jsp?name=\$loginname&password=\$password	Phone self service url
SS_TCP_CLOSE_RESET	Off	Close TCP connection in Direct Reset mode.
SS_TIMEOUT_CALL_FORWARD_NO_ANSWER	20	Time for Call Forwarding, When No Reply(seconds)
SS_TIMEOUT_PHONE_NO_ANSWER	120	Time for Hang Up, When No Reply(seconds)

Parameter Name	Default Value	Parameter Description
SS_TRACE_CALL_FILE_SIZE	16	Call signaling trace file size limit (MB) 16-2048
SS_TRACE_FILE_LENGTH	40960	Size of Softswitch's Debug File(KB)
SS_TRACE_MASK	ERROR	Set Display of Debug Information
SS_TRACE_REGISTER_FILE_SIZE	16	Registration signaling trace file size limit (MB) 16-2048
SS_TRACE_REGISTER_MAX_TRANSMIT	5	Registration Track Maximum Data Size(MB)
SS_TRY_PROTECT_ROUTE_DELAY	None	Protect rout enable time (seconds) 0-180
SS_TRACETOFILE	On	Output Debug Information into File
SS_UNBOUND_INDICATION	Off	Prompt whether the Phone Card is Binded
SS_USE_CALLER_PHONE_DISPLAY	Off	SS_USE_CALLER_PHONE_DISPLAY
SS_VALUE_ADDED_CODECS	g729a,g729,g723,g711a,g711u	Audio codecs prior to value added (g729a,g729,g723,g711a,g711u)
SS_VALUE_ADDED_IP		Value Added Server IP
SS_VALUE_ADDED_PORT	5055	Value Added Server Port
SS_VIRTUAL_IPS		DMZ Settings

3.2.5.4 Audio Service Parameter

Parameter Name	Default Value	Parameter Description
IVR_CALLBACK_KEEP_LINE_RING_TIME	5	Alerting time for callback caller reservation 0-120
IVR_CALLBACK_KEEP_LINE_TIME	30	Used for callback line keep. See IVR_CALLBACK_KEEP_LINE_RING_TIME
IVR_CALL_REPORT_IP		Send IVR second line's call state. UDP request format: Call ID, Serial Number, Call State, Caller Number, Callee Number, Forward Number, Menu ID, Menu Name. Response format: Call ID, Serial Number. Call State: Ringing(180/183)/OK/Bye
IVR_CALL_REPORT_PORT	8000	Report UDP Port
IVR_CALL_REPORT_RETRY	6	Call State Notify Retry Times
IVR_CALL_REPORT_RETRY_INTERVAL	3	Call State Notify Retry Interval
IVR_CODEC_PRIORITY	g729a,g729,g723,g711a,g711u	Voice Codecs Priority (g729a,g729,g723,g711a,g711u)
IVR_DEFAULT_ERROR_AUDIO	defaulterror	Default Error Message Voice

IVR_DEFAULT_LANGUAGE	chinese	Use Default Language when IVR don't know Client's Language
IVR_ENABLE_CARD_LINE_RESTRICT	On	Each Phone Card cannot use IVR service at the same time.
IVR_ENABLE_PARSE_INBAND	Off	Inband DTMF Analysis
IVR_ENABLE_PARSE_SECOND_INBAND	Off	Second Line Inband DTMF Analysis
IVR_MEDIA_CHECK_TIME_OUT	2	No media data hang up duration(minutes) 1-120
IVR_PARSE_DTMF_MODE	Auto	DTMF Analysis Mode
IVR_RESERVE_CALL_KEEP_TIME	300	Reservation callee number keep time(seconds) 10-1800
IVR_RINGING_TIMEOUT	120	Time for IVR Hang Up, When No Reply(seconds)
IVR_RTP_PORT	40000,47999	Media Port Range
IVR_SETUP_TIMEOUT	20	Invite Timeout Duration
IVR_SHORTE164_AUTO_RECORD_ENABLE	Off	Auto Save Number Function
IVR_SHORTE164_AUTO_RECORD_LENGTH	4	Auto Save Number Suffix
IVR_SHORTE164_AUTO_RECORD_MAX_NUMBER	5	Auto Save Number Amount

IVR_SIP_NO_TIMER_MAX_SESSION_TIME	7200	Maximum Conversation Time for Non-Timer SIP Caller
IVR_SIP_RESEND_INTERVAL	3	Interval for Resend SIP Message(seconds)
IVR_SIP_SEND_RETRY	6	Times of Resend SIP Message
IVR_SIP_SESSION_RETRY	6	Retry Times for Detecting SIP Connected Status
IVR_SIP_SESSION_TTL	600	Interval for Detecting SIP Connected Status(seconds)
IVR_SOFTSWITCH_IP		IP of Softswitch
IVR_SOFTSWITCH_SIP_PORT	5060	Port of SIP Softswitch
IVR_SOFTSWITCH_AVAILABLE_IP		Access IP List, separated by commas
IVR_TRACE_FILE_SIZE	40960	Size of Softswitch's Debug File(KB) 4096-4096000
IVR_TRACE_MASK	ERROR	Set Display of Debug Information
IVR_TRACE_TO_FILE	On	Output Debug Information into File
IVR_WEB_CALLBACK_SAMPLE_TIME_CODEC	g729a	Codec for Call Both Side
IVR_ALARM_CALLER_E164		Voice Alarm Caller Number

IVR_ALARM_CONFIRM_KEY		Voice Alarm Confirm Key
IVR_ALARM_PERIOD	5	Voice Alarm Period(minutes)
IVR_ALARM_PRE_AUDIO	alarmpreaudio	Voice Alarm Pre-Prompt Audio
IVR_ALARM_RETRY	6	Voice Alarm Retry Times
IVR_ALARM_RETRY_INTERVAL	20	Voice Alarm Retry Interval
IVR_VOICEMAIL_EXPIRE_DAY	7	Voice Mail Preservation Days
IVR_VOICEMAIL_MAX_NUMBER	10	Voice Mail Max Items
IVR_VOICEMAIL_MAX_TIME	60	Voice Mail Recording Length(seconds)
IVR_VOICEMAILWELCOME	voicemailwelcome	Default Audio for Voice Mail Access

3.3 Softswitch recorded cdr file in txt

When the softswitch's parameter "SS_CDRRECORDTOFILE" is set to On, the softswitch will record cdr into txt file in the directory cdr beneath the installation directory.

The format of the cdr file name will be YYYYMMDDHH.txt(YYYY-year, MM-month, DD-day, HH-hour).

The softswitch will generate one cdr file per hour, as 2013103112.txt will record the the cdrs that end between 2013-10-31 12:00 and 2013-10-31 13:00.

Each line represents one cdr in the txt file. The format is tabled as below:

```
callerE164 |calleeE164 |startTime |stopTime |holdTime |endReason |endDirection
|callerGatewayId |calleeGatewayId |callerIp |calleeIp |callerAccessE164
```

|calleeAccessE164 |callerToGatewayE164 |calleeToGatewayE164 |calleeBilling
|billingMode |callerPdd |calleePdd

segment	description
callerE164	The caller id
calleeE164	The callee id
startTime	Begin time, as 2013-10-31 11:20:18
stopTime	End time, as 2013-10-31 16:34:09
holdTime	Call durarion (in milliseconds)
endReason	End reason
endDirection	Hangup side (0-caller, 1-callee, 2-server)
callerGatewayId	Calling gateway
calleeGatewayId	Called gateway
callerIp	Caller ip
calleeIp	Callee ip
callerAccessE164	Incoming caller
calleeAccessE164	Incoming callee
callerOutE164	Outbound caller
calleeOutE164	Outbound callee
calleeBilling	Billing method (0-By caller, 1-By callee)

billingMode	Charge mode (-1-bobilling, 0-phone number, 1-gateway ID, 3-phone card)
callerPdd	time elapsed from call received to call connected
calleePdd	time elapsed from call sent to routing response

3.4 Call End Reason

3.4.1 Server Side

Terminations caused by the server are all defined by VOS3000. For any questions related to this kind of causes, please consult your system supplier.

Following are the causes of this kind:

Response timeout: the called did not answer the phone before the timeout limit was reached. The timeout limit can be set either by the "Alerting" signal of Routing Gateway (when the call is processed through Routing Gateway) or by the "SS_TIMEOUT_PHONE_HANGUP" parameter in the Softswitch system (when the call is directed to a phone)

Protocol limit: the server cannot process this type of protocol

Connection timeout: No response to the SIP message was received after specified number of trials. The maximal number of trials can be specified by the "SS_SIP_RESEND_INTERVAL" and "SS_SIP_SEND_RETRY" parameters in the Softswitch system.

Busy: the number being called is busy.

Account locked: the account is disabled. It may also due to the lockdown of its agent account.

Session timeout: the caller or the called supports the SIP Timer protocol, yet it has not received the updating signal within a time limit; or, neither does the caller nor the called support the SIP Timer protocol, yet the session time exceeded the value specified by the system parameter

SS_SIP_NO_TIMER_REINVITE_INTERVAL".

Caller's number restricted: the length of the caller's number exceeds the length specified by the system parameter "SS_CALLERALLOWLENGTH".

Called number restricted: the length of the called number exceeds the length specified by the system parameter "SS_CALLERALLOWLENGTH".

Proceeding timeout: No response is received from the server within the time limit. The time limit can be specified by the "Setup" and "Callproceeding" parameters in the routing gateway.

Caller's number prefix restricted: the mapping gateway does not accept the caller's number prefix.

Unregistered: the terminal is not register, and is not allowed to make the call.

Called number locked: the called is currently locked.

Called number prefix restricted: the gateway for the caller or the called gateway does not accept the called number prefix.

Caller locked: the caller is currently locked.

Connection establishment timeout: the connection is not established within the timeout limit set by the mapping gateway. The time limit can be changed by setting the proceeding timeout parameter in the mapping gateway.

Account expired: the account is expired.

Connection limit exceeded: the maximum number of outgoing calls is reached. The maximum number can be specified in the system.

Forcible hang-up: the server disconnected the session, usually because the client chose to end the session in their user interface.

Account disabled: the account is currently disabled. Please check the status of the account it belongs to.

The called not online: There is no appropriate device to accept this call. For example, there is no matching routing gateway.

No-answer forwarding by the caller: the caller has set the no-answer forwarding

Timed forwarding: the call matches the timed forwarding criteria specified in the phone management settings.

On-busy forwarding: the call matches the on-busy forwarding criteria specified in the phone management settings.

No-answer forwarding by the called: the call matches the no-answer forwarding criteria specified in the phone management settings.

Forwarding loop: Due to the wrong configuration of users, the forwarding route has loops.

Call forwarding by the called: the call matches the call forwarding criteria specified in the phone management settings.

Do-not-disturb from the called: the called is in the do-not-disturb status

Session closed by the called: the called did not send the hang-up signal, but disconnected the TCP connection

Session closed by the caller: the caller did not send the hang-up signal, but disconnected the TCP connection

Illegal call: the call comes from an unauthorized IP address and the caller's number is not registered in the system.

No matching rate: There is no rate that matches this call.

No matching account: There is no account to bill this call.

Insufficient balance: the account has insufficient balance

Call restriction: the call is prevented by restrictions (such as <International call>) set by the phone or the gateway.

Hang-up by the called: the hang-up signal comes from the called.

Hang-up by the caller: the hang-up signal comes from the caller.

3.4.2 Client Side

3.4.2.1 H323 Device

UnknownCauseIE

UnallocatedNumber

NoRouteToNetwork

NoRouteToDestination

SendSpecialTone

MisdialledTrunkPrefix

ChannelUnacceptable

CallAwarded

Preemption

PreemptionCircuitReserved

NormalCallClearing

UserBusy

NoResponse

NoAnswer

SubscriberAbsent

CallRejected

NumberChanged

Redirection

ExchangeRoutingError

NonSelectedUserClearing

DestinationOutOfOrder

InvalidNumberFormat

FacilityRejected

StatusEnquiryResponse

NormalUnspecified

NoCircuitChannelAvailable

CallQueued

NetworkOutOfOrder

FrameModeOOS

FrameModeOperational

TemporaryFailure

Congestion

AccessInformationDiscarded
RequestedCircuitNotAvailable
PrecedenceCallBlocked
ResourceUnavailable
QoSNotAvailable
RequestedFacilityNotSubscribed
OutgoingCallsBarred
OutgoingCallsBarredInCUG
IncomingCallsBarred
IncomingCallsBarredInCUG
BearerCapNotAuthorised
BearerCapNotPresentlyAvailable
InconsistentOutgoingIE
ServiceOptionNotAvailable
BearerCapNotImplemented
ChannelTypeNotImplemented
RequestedFacilityNotImplemented
OnlyRestrictedDigitalBearerCapAvailable
ServiceOrOptionNotImplemented
InvalidCallReference
IdentifiedChannelNonExistent
CallIdentifyNotSuspendedCall
CallIdentifyInUse
NoCallSuspended
ClearedRequestedCallIdentity

UserNotInCUG
IncompatibleDestination
NonexistentCUG
InvalidTransitNetwork
InvalidMessageUnspecified
MandatoryIEMissing
MessageTypeNonexistent
MessageNotCompatible
IENonExistantOrNotImplemented
InvalidIEContents
MessageNotCompatibleWithCallState
TimerExpiry
ParameterNonexistent
UnrecognisedParamaterDiscarded
ProtocolErrorUnspecified
InterworkingUnspecified
ErrorInCauseIE

3.4.2.2 SIP Device

Multiple Choices
Moved Permanently
Moved Temporarily
Use Proxy
Alternative Service
Bad Request
Unauthorized

Payment Required

Forbidden

Not Found

Method not Allowed

Not Acceptable

Proxy authentication Required

Request Timeout

Gone

Request Entity Too Large

Request-URI Too Long

Unsupported Media Type

Unsupported URI Scheme

Bad Extension

Extension Required

Session Interval Too Small

Interval Too Brief

Temporarily Unavailable

Call/Transaction Does not Exist

Loop Detected

Too Many Hops

Address Incomplete

Ambiguous

Busy Here

Request Terminated

Not Acceptable Here

Request Pending
Server Internal Error
Not Implemented
Bad Gateway
Service Unavailable
Server Time-out
Version not Supported
Message Too Large
Busy Everywhere
Decline
Does not Exist Anywhere
Not Acceptable

3.5 First Usage

3.5.1 Whole Sales

Operation

Create Fee Rate Group
Create Fee Rate
Create Account
Create Mapping Gateway
Create Routing Gateway

Example

Customer IP: 172.16.1.11
Routing Gateway IP: 172.16.1.12

Double click "Navigation > Rate management", add rate group 0.5 and apply.



Double click "Number of rates > 0" to add fee rate.

Rate management		Rate<0.05>management X		
Rate prefix	Area prefix	Rate type	Area name	Billing rate
	25	Domestic		0.0500000

Double click "Navigation > Account management", add account test and apply.

Account management X				
Account id	Account name	Current balance	Overdraft limit	Billing rate
test	test	12,221.970	0.000	0.05

Double click "Number of gateway > 0" to add mapping gateway.

Account management		Account<test>gateway management X		
Gateway id	Lock type	Authorization type	Gateway group	Line limit
caller.128	No lock	Domestic		30.

Double click "Navigation > Routing gateway", add routing gateway caller.125 and apply.

Routing gateway X					
Gateway id	Gateway prefix	Prefix mode	Gateway group	Lock type	Line limit
callee.125	9	Continual		No lock	30

After configuration, ask customer to send SIP call, then open "Navigation > Data query > Cdr" to see call result.

4 Maintenance

About This Chapter

This chapter describes the maintenance of VOS3000 solutions.



CAUTION

If one server deployed multiple services, should start and stop in the following order.

4.1 Service Start

Step 1 vos3000

```
/etc/init.d/vos3000d start
```

Step 2 mgc

```
/etc/init.d/mgc start
```

Step 3 emp

```
/etc/init.d/empd start
```

Step 4 mediaagent

```
/etc/init.d/mediaagentd start
```

Step 5 callservice

```
/etc/init.d/callserviced start
```

Step 6 mbx3000

```
/etc/init.d/mbx3000d start
```

Step 7 webdata

```
/etc/init.d/webdatad start
```

Step 8 webserver

```
/etc/init.d/webserverd start
```

Step 9 dial

```
/etc/init.d/diald start
```

Step 10 ipcentrex

```
/etc/init.d/ipcentrex start
```

Step 11 valueadded

```
/etc/init.d/valueadded start
```

Step 12 audioplayer

```
/etc/init.d/audioplayerd start
```

Step 13 servermonitor

```
/etc/init.d/servermonitord start
```

4.2 Service Stop

Note: actually, service stop sequence is the reverse order of service start

Step 1 servermonitor

```
/etc/init.d/servermonitord stop
```

Step 2 audioplayer

`/etc/init.d/audioplayerd stop`

Step 3 valueaddeddd

`/etc/init.d/valueaddeddd stop`

.....

Step 13 vos3000

`/etc/init.d/vos3000d stop`

Thank you very Much, Please stay Connected at our Blog: www.multahost.com/blog/ for more security updates, Manuals, Training Etc.