**C H A P T E R 36**

# Sites Settings

**Impacted GUIs: Cisco Unified Communications Manager Business Edition 3000 First Time Setup Wizard and Cisco Unified Communications Manager Business Edition 3000 Administrative Interface**

**Tip** You can also add the Site settings to the Cisco-provided .xls data configuration file.

Sites are the geographical locations where users (employees) work.

- Central Site—In most cases, the central site is the location where the majority of users work; in most cases, the company headquarters is the central site. In all cases, the Cisco Unified Communications Manager Business Edition 3000 server is located at the central site. The central site is mandatory; therefore, you cannot delete it. You can have only 1 central site.

- Remote Sites—Remote sites, which are optional, are branch offices that work with the central site; a WAN link or Internet connection and routers must exist between the central and remote sites. You must have dedicated subnets for remote sites. You can have up to 9 remote sites.

- Teleworker Site—A teleworker site, which is optional, is a site that is for workers that do not work only at the central site or branch offices; teleworkers use VPN connections to connect to the central site, and no router is required to contact the central site because their Internet connection provides access to the central site. You can have 1 teleworker site.

You can configure multiple branches for a central site. If you have multiple sites, you first configure the central site and then the remote sites. The following topics contain information about the tabs and settings that display on the Sites page (**Connections > Sites**):

- Configuring Central Site, page 36-2
- Configure Remote Site, page 36-9
- Configure Remote User Site, page 36-15
- Configuring Logical Partitioning, page 36-18

Table 36-1 describes the fields under the Sites tab, which display only in the Cisco Unified Communications Manager Business Edition 3000 First Time Setup Wizard.

*Table 36-1      Site Configuration Fields in the Cisco Unified Communications Manager Business Edition 3000 First Time Setup Wizard*

| Field | Description |
|-------|-------------|
| Site Options | • **Configure Central Site Only (Single Site)**—Select if you only have one site.<br>• **Configure Central Site and One or More Branch Sites (Multisite)**—Select to configure the central site and one or more branch offices.<br>• **Number of Branch Sites**—Select the number of branches you want to configure. |
| Remote Users | **Support Offsite Phones for Teleworkers**—Select to allow offsite phones for users who do not work at the central or branch offices. |

# Configuring Central Site

This section describes how to configure a central site. The central site is the location where the majority of users work; in all cases, the Cisco Unified Communications Manager Business Edition 3000 server is located at the central site. In most cases, the company headquarters is the central site.

The Central Site can be configured during the Cisco Unified Communications Manager Business Edition 3000 First Time Setup Wizard.

Table 36-2 describes the settings for the Central site.

*Table 36-2      Configure Central Site*

| Field | Description |
|-------|-------------|
| **General Tab** | |
| **Basic Site Information** | |
| Name | Enter a name for your central site. |
| Description | Enter the description for your central site. |

*Table 36-2      Configure Central Site (continued)*

| Field | Description |
|-------|-------------|
| Local Mobile Phone Prefixes | Enter the local mobile phone prefixes for the central site. You can enter multiple local mobile phone prefixes separated by commas (,).<br><br>The local mobile phone prefix for a site consists of the first few digits of the mobile phone number, and it uniquely determines the service provider and the local zone or region of the site where the call is considered local.<br><br>Local mobile phone prefix must have a length that is valid in the country where you will set up the Cisco Unified Communications Manager Business Edition 3000. For example, in India, you must enter the first four digits of the mobile phone numbers as the local mobile phone prefix. In China, you must enter the first seven digits of the mobile phone numbers as the local mobile phone prefix.<br><br>You need to specify the local mobile phone prefixes only for some countries such as India, China, and so on. |
| **Internal Networks** | |

*Table 36-2        Configure Central Site (continued)*

| Field | Description |
|---|---|
| Subnet Mask | The subnet and subnet masks allow you to associate your phone(s) and gateway(s) with a site. As a phone/gateway registers with the Cisco Unified Communications Manager Business Edition 3000 server, the server learns the IP address of the phone/gateway. The server then searches the subnet and masks for every site to determine the best match. The phone/gateway gets placed in that site. If no match is found, by default:<br><br>• The phones are considered to be placed in your central site<br><br>• The gateways are considered to be placed in an unknown site<br><br>**Note**    The site at which a gateway is placed determines whether your PSTN calls are routed through the that gateway, based on your Gateway Usage settings.<br><br>By default, the subnet address 192.168.1.0 with a subnet mask of 24 displays on the Central Site page. You may update this information if it does not apply to your setup.<br><br>**Note**    The subnet mask value is the same which you have updated in the **System Settings > Network (Step 1 of 2)** in the Cisco Unified Communications Manager Business Edition 3000 First Time Setup Wizard.<br><br>Enter the subnet address and specify the number of subnet masks. Enter subnet addresses and masks that are unique for each site.<br><br>To add more subnets, click the **Plus** icon. To delete subnets, click the **Minus** icon. |
| **Media Access Allowed** | |
| Allow Access to Conference Bridge | Check this check box to allow access to the conference bridge, which is an internal component of the Cisco Unified Communications Manager Business Edition 3000 software that is used for conferences. |

*Table 36-2    Configure Central Site (continued)*

| Field | Description |
|---|---|
| Allow Access to Music On Hold Server | Check this check box to allow access to music on hold, which plays music on the phone when the user is on hold.<br><br>You set the audio source for music on hold in the usage profile. You upload audio source .wav files for music on hold on the Music On Hold page in the Cisco Unified Communications Manager Business Edition 3000 Administrative Interface (**System Settings > Music On Hold**). |
| Allow Access to Transcoder | Check this check box to allow access to a transcoder, which is a component of the Cisco Unified Communications Manager Business Edition 3000 software that can take the media stream from a codec and converts it from one compression type to another compression type. |
| **Call Settings Tab** | |
| **Call Privileges Information** | |
| Highest Privilege Allowed | Select the calling privileges for this site. The order goes from lowest privilege to highest privilege. For example, if you do not want the users at the sites to be able to make international calls but the users can make long distance calls, select Long Distance Calls.<br><br>This value applies to the entire site, so select a value that accommodates all users. Because this setting applies to the entire site, you restrict whether some users can make certain types of calls by updating the Highest Level of Calls Allowed setting in the usage profile and then applying the usage profile to the users. If the value for the Highest Privilege Allowed does not match the value that you set for the Highest Level of Calls, the lowest level that is configured for the settings get applied to the users. |
| **Emergency Calling Information** | |

***Table 36-2        Configure Central Site (continued)***

| Field | Description |
|-------|-------------|
| Allow Emergency Calls from this site | Check this check box to allow making emergency calls from this site. Emergency calls are made to the center that addresses emergencies for your municipality. You can restrict whether individual users can make emergency calls by checking or unchecking the Emergency Calls check box in a usage profile (**Users/Phones > Usage Profiles**) and then applying that usage profile to the user (**Users/Phones > Users**).<br><br>If this remote site is not part of the Local Exchange Carrier (LEC), do not check this check box. In most cases, for the teleworker site, uncheck this check box. |
| Additional Emergency Services Numbers | Enter the additional emergency services numbers.<br><br>Additional Emergency Numbers are treated as emergency calls (for example, a police station number in the city to contact during emergency).<br><br>To add multiple additional emergency services numbers, ensure that you separate each number with commas. |
| Emergency Location ID Numbers (ELINs) | Enter the emergency location ID number within your respective site or location.<br><br>To add multiple emergency location ID numbers, ensure that you separate each ELIN with commas.<br><br>⚠️<br>**Caution**  For the emergency services to correctly identify the site location as the source of emergency calls, you must specify at least one ELIN. This ELIN phone number must be unique to a site and be registered with its location. |
| **PSTN Access Information** | |
| Gateway Usage | This setting allows you to control the use of gateways placed in different sites for routing PSTN calls made from phones placed in this site.<br><br>Chosen gateways are ordered based on gateway description. Calls are distributed in a top-down order among this chosen gateways group.<br><br>Please understand your country's regulations thoroughly before you configure this. Configuring this incorrectly may lead to toll bypass rules being violated. |

*Table 36-2 Configure Central Site (continued)*

| Field | Description |
|---|---|
| All Gateways | This option will configure all gateways registered to existing sites in the system to be used for routing PSTN calls made from phones placed in this site. |
| | Click **Show All Gateways** to display all the gateways available for use. |
| | Click **Hide All Gateways** to hide the gateways from the displayed list. |
| Local Gateways Only | This option will enforce only gateways placed in this site to be used for routing PSTN calls made from phones placed in this site. |
| | Click **Show Local Gateways** to display the local gateways available for use. |
| | Click **Hide Local Gateways** to hide the local gateways from the displayed list. |
| Custom | This option will allow you to choose gateways from a list of gateways added to your system to be used for routing outbound PSTN calls made from phones placed in this site. You may choose to order the selected gateways you wish to use. |
| **Call Quality Tab** | |
| **Call Between Sites Information** | |
| Bandwidth Between Sites | Select the amount of span that you are leasing from your service provider for your Internet and intranet connectivity. |
| | Note    Sites which operate with a network bandwidth less than T1 to a central site are not supported for point to point video calls. |
| Video: Enable Video Calls Between Sites | Check box to enable video calls between sites. |
| | Note    Point-to-point video is not supported within the teleworker site or between the teleworker site and any other site connected to the teleworker site. |

*Table 36-2     Configure Central Site (continued)*

| Field | Description |
|---|---|
| Bandwidth Allocation for Audio and Video<br><br>⚠️<br>**Caution**  The number of video calls is expected to be small. Because bandwidth is usually limited between sites, the system does not reserve video bandwidth for infrequent video calls because this bandwidth can be used for the audio-only calls. Thus, if a large number of video calls are made (relative to the number of video calls between sites as shown on the sites page), audio and video quality can suffer between the sites. If you encounter poor quality due to a large number of video calls, then you may find it necessary to disable video to and from site. | Select the percentage of span that you want to make available for combined audio and video between sites; for example, between the central and remote site. The rest of the span gets used for data communication between sites.<br><br>You can use the slider to select the value or enter in the field provided. |
| Quality/Quantity Tradeoff | Determine whether the quality of (audio and video) calls or the number of total calls is more important for calls between the sites; for example, between the central and teleworkers site or central and branch sites. Slide the scale to accommodate your decision.<br><br>You can use the slider to select the value.<br><br>**Note**  Point-to-point video is not supported within the teleworker site or between the teleworker site and any other site connected to the teleworker site. |
| Approximate Call Capacity to Other Sites | This setting, which is read only, lists the maximum number of calls (audio and video) that can occur between the sites. The number changes if you update any of the settings in the Calls Between Sites section. |
| **Calls Within Sites** | |
| Quality/Quantity Tradeoff | Determine whether the quality of calls (audio and video) or the number of total calls is more important for calls within the site that you are configuring. Slide the scale to accommodate your decision.<br><br>You can use the slider to select the value. |

# Configure Remote Site

This section describes how to configure a branch site. Remote sites represent physical locations where your branch offices are located. Teleworker site represents a virtual location used to place users that do not work at the central site or any remote site. You can configure multiple branch/remote sites in your system.

Table 36-3 describes the settings for remote sites, including branch offices and the teleworker site.

*Table 36-3        Configure Central and Branch Sites*

| Field | Description |
|---|---|
| **General Tab Information** | |
| **Basic Site Information** | |
| Name | Enter the name of your Remote Site. |
| Description | Enter a description for your Remote Site. |
| Time Zone | From the drop-down list box, select the time zone for the site. |
| **Internal Networks** | |

*Table 36-3      Configure Central and Branch Sites (continued)*

| Field | Description |
|---|---|
| Subnet Mask | The subnet and subnet masks allow you to associate your phone(s) and gateway(s) with a site. As a phone/gateway registers with the Cisco Unified Communications Manager Business Edition 3000 server, the server learns the IP address of the phone/gateway. The server then searches the subnet and masks for every site to determine the best match. The phone/gateway is placed in that site. If no match is found, by default: |
| | • The phones are considered to be placed in your Central Site |
| | • The gateways are considered to be placed in an unknown site |
| | **Note** Site at which a gateway is placed decides whether your PSTN calls are routed through the that gateway, based on your Gateway Usage settings. |
| | By default, subnet address 192.168.1.0 with a subnet mask of 24 displays on the central site page. You may update this information if it does not apply to your setup. |
| | **Note** The subnet mask value is the same which you have updated in the **System Settings > Network (Step 1 of 2)** in the Cisco Unified Communications Manager Business Edition 3000 First Time Setup Wizard. |
| | Enter the subnet address and specify the number of subnet masks. Enter subnet addresses and masks that are unique for each site. |
| | To add more subnets, click the **Plus** icon. To delete subnets, click the **Minus** icon. |
| **Media Access Allowed** | |
| Allow Access to Conference Bridge | Check this check box to allow access to the conference bridge, which is an internal component of the Cisco Unified Communications Manager Business Edition 3000 software that is used for conferences. |

*Table 36-3       Configure Central and Branch Sites (continued)*

| Field | Description |
|---|---|
| Allow Access to Music On Hold Server | Check this check box to allow access to music on hold, which plays music on the phone when the user is on hold. |
| | You set the audio source for music on hold in the usage profile. You upload audio source .wav files for music on hold on the Music On Hold page in the Cisco Unified Communications Manager Business Edition 3000 Administrative Interface (**System Settings > Music On Hold**). |
| Allow Access to Transcoder | Check this check box to allow access to a transcoder, which is a component of the Cisco Unified Communications Manager Business Edition 3000 software that can take the media stream from a codec and converts it from one compression type to another compression type. |
| **Call Settings Tab** | |
| **Calls Privileges Information** | |
| Highest Privilege Allowed | Select the calling privileges for this site. The order goes from lowest privilege to highest privilege. For example, if you do not want the users at the sites to be able to make international calls but the users can make long distance calls, select Long Distance Calls. |
| | This value applies to the entire site, so select a value that accommodates all users. Because this setting applies to the entire site, you restrict whether some users can make certain types of calls by updating the Highest Level of Calls Allowed setting in the usage profile and then applying the usage profile to the users. If the value for the Highest Privilege Allowed does not match the value that you set for the Highest Level of Calls, the lowest level that is configured for the settings get applied to the users. |
| **Emergency Calling Information** | |

*Table 36-3      Configure Central and Branch Sites (continued)*

| Field | Description |
|---|---|
| Allow Emergency Calls from this site | Check this check box to allow making emergency calls from this site. Emergency calls are made to the center that addresses emergencies for your municipality. You can restrict whether individual users can make emergency calls by checking or unchecking the Emergency Calls check box in a usage profile (**Users/Phones > Usage Profiles**) and then applying that usage profile to the user (**Users/Phones > Users**). |
| | If this remote site is not part of the Local Exchange Carrier (LEC), do not check this check box. In most cases, for the teleworker site, uncheck this check box. |
| Additional Emergency Services Numbers | Enter the additional emergency services numbers. |
| | Additional Emergency Numbers are treated as emergency calls (for example, a police station number in the city to contact during emergency). |
| | To add multiple additional emergency services numbers, ensure that you separate each number with commas. |
| Emergency Location ID Numbers (ELINs) | Enter the emergency location ID number within your respective site or location. |
| | To add multiple emergency location ID numbers, ensure that you separate each ELIN with commas. |
| | ⚠️ **Caution**  For the emergency services to correctly identify the site location as the source of emergency calls, you must specify at least one ELIN. This ELIN phone number must be unique to a site and be registered with its location. |
| **PSTN Access Information** | |
| Gateway Usage | This setting allows you to control the use of gateways placed in different sites for routing PSTN calls made from phones placed in this site. |
| | Chosen gateways are ordered based on gateway description. Calls are distributed in a top-down order among this chosen gateways group. |
| | Please understand your country's regulations thoroughly before you configure this. Configuring this incorrectly may lead to toll bypass rules being violated. |

*Table 36-3        Configure Central and Branch Sites (continued)*

| Field | Description |
|-------|-------------|
| All Gateways | This option will configure all gateways registered to existing sites in the system to be used for routing PSTN calls made from phones placed in this site. |
| | Click **Show All Gateways** to display all the gateways available for use. |
| | Click **Hide All Gateways** to hide the gateways from the displayed list. |
| Local Gateways Only | This option will enforce only gateways placed in this site to be used for routing PSTN calls made from phones placed in this site. |
| | Click **Show Local Gateways** to display the local gateways available for use. |
| | Click **Hide Local Gateways** to hide the local gateways from the displayed list |
| Custom | This option will allow you to choose gateways from a list of gateways added to your system to be used for routing outbound PSTN calls made from phones placed in this site. You may choose to order the selected gateways you wish to use. |
| **Call Quality Tab** | |
| **Call Between Sites Information** | |
| Bandwidth Between Sites | Select the amount of span that you are leasing from your service provider for your Internet and intranet connectivity. |
| | **Note**    Sites which operate with a network bandwidth less than T1 to a central site are not supported for point-to-point video calls. |
| Video: Enable Video Calls Between Sites | Check box to enable video calls between sites. |
| | **Note**    Point-to-point video is not supported within the teleworker site or between the teleworker site and any other site connected to the teleworker site. |

*Table 36-3        Configure Central and Branch Sites (continued)*

| Field | Description |
|---|---|
| Bandwidth Allocation for Audio and Video<br><br>⚠️ **Caution**  The number of video calls is expected to be small. Because bandwidth is usually limited between sites, the system does not reserve video bandwidth for infrequent video calls because this bandwidth can be used for the audio-only calls. Thus, if a large number of video calls are made (relative to the number of video calls between sites as shown on the sites page), audio and video quality can suffer between the sites. If you encounter poor quality due to a large number of video calls, then you may find it necessary to disable video to and from the sites. | Select the percentage of span that you want to make available for combined audio and video between sites; for example, between the central and remote site. The rest of the span gets used for data communication between sites.<br><br>You can use the slider to select the value or enter in the field provided. |
| Quality/Quantity Tradeoff | Determine whether the quality of (audio and video) calls or the number of total calls is more important for calls between the sites; for example, between the central and teleworkers site or central and branch sites. Slide the scale to accommodate your decision.<br><br>You can use the slider to select the value.<br><br>**Note**    Point to point video is not supported within the teleworker site or between the teleworker site and any other site connected to the teleworker site. |
| Approximate Call Capacity to Other Sites | This setting, which is read only, lists the maximum number of calls (audio and video) that can occur between the sites. The number changes if you update any of the settings in the Calls Between Sites section. |
| **Calls Within Sites** | |
| Quality/Quantity Tradeoff | Determine whether the quality of calls (audio and video) or the number of total calls is more important for calls within the site that you are configuring. Slide the scale to accommodate your decision.<br><br>You can use the slider to select the value. |

**Administration Guide for Cisco Unified Communications Manager Business Edition 3000**

# Configure Remote User Site

This section describes how to configure a remote user site. Remote user site represents a virtual location used to place users/teleworkers that do not work at the central site or any remote site.

The remote user site uses the Time zone and Local Area Code(s) of the central site. All the phones in the remote user site will display the system time zone. To place PSTN calls outside the central site, prefix the long distance dial code. In countries that have toll bypass regulation, ensure that the PSTN calls from remote user site adheres to their country regulation.

Table 36-4 describes the settings for a remote user.

*Table 36-4       Configure Remote User for Teleworkers Site*

| Field | Description |
|---|---|
| **General** | |
| Name | Enter the name for your Remote User Site. |
| Description | Enter a description for your Remote User Site. |
| **Media Access Allowed** | |
| Allow Access to Conference Bridge | Check this check box to allow access to the conference bridge, which is an internal component of the Cisco Unified Communications Manager Business Edition 3000 software that is used for conferences. |
| Allow Access to Music On Hold Server | Check this check box to allow access to music on hold, which plays music on the phone when the user is on hold.<br><br>You set the audio source for music on hold in the usage profile. You upload audio source .wav files for music on hold on the Music On Hold page in the Cisco Unified Communications Manager Business Edition 3000 Administrative Interface (**System Settings > Music On Hold**). |
| Allow Access to Transcoder | Check this check box to allow access to a transcoder, which is a component of the Cisco Unified Communications Manager Business Edition 3000 software that can take the media stream from a codec and converts it from one compression type to another compression type. |
| **Call Settings Tab** | |
| **Call Privileges Information** | |

*Table 36-4*        *Configure Remote User for Teleworkers Site*

| Field | Description |
|---|---|
| Highest Privilege Allowed | Select the calling privileges for this site. The order goes from lowest privilege to highest privilege. For example, if you do not want the users at the sites to be able to make international calls but the users can make long distance calls, select Long Distance Calls. |
| | This value applies to the entire site, so select a value that accommodates all users. Because this setting applies to the entire site, you restrict whether some users can make certain types of calls by updating the Highest Level of Calls Allowed setting in the usage profile and then applying the usage profile to the users. If the value for the Highest Privilege Allowed does not match the value that you set for the Highest Level of Calls, the lowest level that is configured for the settings get applied to the users. |
| **Emergency Calling Information** | |
| Allow Emergency Calls from this site | Check this check box to allow making emergency calls from this site. Emergency calls are made to the center that addresses emergencies for your municipality. You can restrict whether individual users can make emergency calls by checking or unchecking the Emergency Calls check box in a usage profile (**Users/Phones > Usage Profiles**) and then applying that usage profile to the user (**Users/Phones > Users**). |
| | If this remote site is not part of the Local Exchange Carrier (LEC), do not check this check box. In most cases, for the teleworker site, uncheck this check box. |
| Additional Emergency Services Numbers | Enter the additional emergency services numbers. |
| | Additional Emergency Numbers are treated as emergency calls (for example, a police station number in the city to contact during emergency). |
| | To add multiple additional emergency services numbers, ensure that you separate each number with commas. |

*Table 36-4        Configure Remote User for Teleworkers Site*

| Field | Description |
|---|---|
| Emergency Location ID Numbers (ELINs) | Enter the emergency location ID number within your respective site or location.<br><br>To add multiple emergency location ID numbers, ensure that you separate each ELIN with commas.<br><br>⚠<br>**Caution**    For the emergency services to correctly identify the site location as the source of emergency calls, you must specify at least one ELIN. This ELIN phone number must be unique to a site and be registered with its location. |
| **PSTN Access Information** | |
| Gateway Usage | This setting allows you to control the use of gateways placed in different sites for routing PSTN calls made from phones placed in this site.<br><br>Chosen gateways are ordered based on gateway description. Calls are distributed in a top-down order among this chosen gateways group.<br><br>Please understand your country's regulations thoroughly before you configure this. Configuring this incorrectly may lead to toll bypass rules being violated. |
| All Gateways | This option will configure all gateways registered to existing Sites in the system to be used for routing PSTN calls made from phones placed in this site.<br><br>Click **Show All Gateways** to display all the gateways available for use.<br><br>Click **Hide All Gateways** to hide the gateways from the displayed list |
| Local Gateways Only | This option will enforce only gateways placed in this site to be used for routing PSTN calls made from phones placed in this site.<br><br>Click **Show Local Gateways** to display the local gateways available for use.<br><br>Click **Hide Local Gateways** to hide the local gateways from the displayed list. |
| Custom | This option will allow you to choose gateways from a list of gateways added to your system to be used for routing outbound PSTN calls made from phones placed in this site. You may choose to order the selected gateways you wish to use. |
| **Call Quality Tab** | |

*Table 36-4      Configure Remote User for Teleworkers Site*

| Field | Description |
|---|---|
| **Call Between Sites Information** | |
| Bandwidth Between Sites | Select the amount of span that you are leasing from your service provider for your Internet and intranet connectivity. |
| Video: Enable Video Calls Between Sites.<br><br>**Note**    Point to point video is not supported within the teleworker site or between the teleworker site and any other site connected to the teleworker site. | Check box to enable video calls between sites.<br><br>**Note**    Disabled for Teleworker sites. |
| Bandwidth Allocation for Audio and Video<br><br>**Note**    Point to point video is not supported within the teleworker site or between the teleworker site and any other site connected to the teleworker site. | Select the percentage of span that you want to make available for audio between sites; for example, between the central and remote site. The rest of the span gets used for data communication between sites.<br><br>You can use the slider to select the value or enter in the field provided. |
| Quality/Quantity Tradeoff | Determine whether the quality of calls or the number of total calls is more important for calls between the sites; for example, between the central and teleworkers site or central and branch sites. Slide the scale to accommodate your decision.<br><br>You can use the slider to select the value. |
| Approximate Call Capacity to Other Sites | This setting, which is read only, lists the maximum number of calls that can occur between the sites. The number changes if you update any of the settings in the Calls Between Sites section. |
| **Calls Within Sites** | |
| Quality/Quantity Tradeoff | Determine whether the quality of calls or the number of total calls is more important for calls within the site that you are configuring. Slide the scale to accommodate your decision.<br><br>You can use the slider to select the value. |

# Configuring Logical Partitioning

Some countries have regulations on PSTN access to prevent toll bypass when PSTN networks are bridged with data networks; for example, India. In Cisco Unified Communications Manager Business Edition 3000, you can control how phones in different sites participate during PSTN calls. You can configure Logical Partitioning on Cisco Unified Communications Manager Business Edition 3000 if installed in countries enforcing the Telecom Regulations.

Table 36-5 describes the settings that display on the Edit Logical Partitioning page.

*Table 36-5        Settings on Edit Logical Partitioning page*

| Field | Description |
|---|---|
| **PSTN Access Restrictions** | |
| Default (Local Site Access only) | Choose this to prevent a phone in a remote site participating in a call happening between phone(s) in the local site and PSTN (using the gateway of the local site). |
| Custom (Choose Allowed Sites) | In the grid showing all your sites, check the check boxes to allow PSTN access between a pair of sites. This setting is bidirectional and allows phones from the selected pair of sites to participate in the same call connected to PSTN. |
| | **Note**    When you add a new site, the Add Site page displays all the sites. You can select the site(s) to which phones from the selected site(s) and the new site can participate in the same call connected to PSTN. |