

# Create trunk for KX-HTS32V on Elastix, Number 512 221 2810

The screenshot shows the Elastix PBX Configuration interface. On the left is a sidebar menu with categories like System, Agenda, Email, Fax, PBX, and IM. The main content area is titled 'Edit SIP Trunk' for trunk 5122212810. It includes a 'Delete Trunk' button and a note 'In use by 1 route'. The 'General Settings' section contains fields for Trunk Name (5122212810), Outbound CallerID (hidden), CID Options (Allow Any CID), Maximum Channels, Asterisk Trunk Dial Options (with an Override checkbox), Continue if Busy (with a checkbox to always try next trunk), and Disable Trunk (with a Disable checkbox). Below this is the 'Dial Number Manipulation Rules' section with a dropdown menu set to '(pick one)'. The 'Outgoing Settings' section shows Trunk Name (5122212810) and PEER Details with fields for host (192.168.1.101), username (5122212810), secret (Password1234), type (friend), and fromuser (5122212810).

# Input details from Elastix into KX-HTS32, No 512 221 2810

The screenshot shows the HTS Web Maintenance Console interface. The top header includes 'HTS Web Maintenance Console 001.60031', a language dropdown set to 'English (UK)', and a 'Save & Logout' button. The left sidebar shows a tree view with 'PBX Configuration' expanded to '3.Trunk' and '7.SIP Trunk Property' selected. The main content area is titled 'SIP Trunk Property' and shows configuration details for 'SIP Carrier - 1'. The 'Common' tab is active. Fields include: SIP Registration Status (Not Registered), Provider Name (5122212810), SIP Server Name (empty), SIP Server IP Address (192.168.1.254), SIP Server Port Number (5060), SIP Service Domain (empty), and Subscriber Number (empty). The 'Account' section has User Name (5122212810), Authentication ID (5122212810), and Authentication Password (Password1234). The 'Register' section has Register Ability set to 'Enable'. The 'Calling Party' section has Header Type set to 'From Header', Host Name (empty), Format set to 'Without "+"', and Send CLIP set to 'Enable'.

# Create trunk for KX-HTS32V on Elastix, Number 660 386 4201

The screenshot shows the Elastix PBX Configuration interface. On the left is a navigation menu with categories like System, Agenda, Email, Fax, PBX, and IM. The main content area is titled 'Edit SIP Trunk' for trunk 6603864201. It includes sections for 'General Settings' and 'Dial Number Manipulation Rules'. The 'General Settings' section contains fields for Trunk Name (6603864201), Outbound CallerID (hidden), CID Options (Allow Any CID), Maximum Channels, Asterisk Trunk Dial Options (with an Override checkbox), Continue if Busy (with a checkbox to check to always try next trunk), and Disable Trunk (with a Disable checkbox). The 'Dial Number Manipulation Rules' section has a table with columns for '(prepend)', '+ prefix', and 'match pattern', and buttons for '+ Add More Dial Pattern Fields' and 'Clear all Fields'. Below this is a 'Dial Rules Wizards' dropdown set to '(pick one)' and an 'Outbound Dial Prefix' field. The 'Outgoing Settings' section includes fields for Trunk Name (6603864201) and PEER Details (host=192.168.1.101, username=6603864201, secret=Password1234, type=friend, fromuser=6603864201). A 'Delete Trunk 6603864201' button is visible at the top left of the configuration area.

# Input details from Elastix into KX-HTS32, No 660 386 4201

The screenshot shows the HTS Web Maintenance Console interface. The header includes 'HTS Web Maintenance Console' with the phone number 001.60031, a language dropdown set to 'English (UK)', and a 'Save & Logout' button. The main content area is titled 'SIP Trunk Property' and shows the configuration for trunk 6603864201. The configuration is organized into several sections: 'Not Registered' (Provider Name: 6603864201, SIP Server Name: [empty], SIP Server IP Address: 192.168.1.254, SIP Server Port Number: 5060), 'Account' (User Name: 6603864201, Authentication ID: 6603864201, Authentication Password: Password1234), 'Register' (Register Ability: Enable), and 'Calling Party' (Header Type: From Header, Host Name: [empty], Format: Without "+", Send CLIP of caller when call from SIP carrier is forwarded to same SIP carrier: Enable). A left-hand navigation menu shows 'PBX Configuration' and 'Maintenance' sections.