

# VoIP

- Asterisk VoIP
  - Fix - asterisk.ctl: No such device or address
  - Fix - This module requires Asterisk chan\_motif & res\_xmpp to be installed and loaded
  - Hangup All Calls Asterisk CLI
  - Monitor PJSip Trunk Registration with Bash Script
  - Originate Call from Command Line - Asterisk
  - Upgrade Asterisk core in Asterisk Now 1.6 > 1.8
  - Use Asterisk CLI in Bash Script
- Call Detail Records (CDR)
  - Install WebCDR - Open Source GIT CDR Front End
    - Disable Auth access to API in WebCDR
- Capture VoIP traffic with TCPDUMP and Examine in Wireshark
- Enable TFTPBoot for Hardware SIP Phones
- FOP2
  - Activate-Revoke FOP2 License
  - File does not exist: /var/www/html/fop2/js/lang\_..js
  - Fix - Failed to connect to asterisk - localhost:5038 - FOP2
  - FOP2 Checkdir.php Custom Pop\_Up
  - FreePBX + FOPv2 Integration - For Older FreePBX
  - Install FOP2
  - Integrate FOP2 with FreePBX
- FreePBX VoIP
  - Configure Google Voice Motif + Asterisk SIP for Google Voice - FreePBX 13
  - Customize Images on FreePBX
  - Disable "If this is correct, press 1" from VoiceMail Blasting - FreePBX
  - Fix FreePBX can't Connect to Asterisk
  - Locked out of FreePBX GUI
  - PIN conference is no longer optional - FreePBX 12 + Asterisk 1.8
  - Pre and Post Backup Hooks to add Database to FreePBX Backup Module
  - Remove 0 Second Calls FreePBX - Asterisk
  - Set TimeZone for Voicemail in FreePBX - Asterisk
- Install LAME 3.99
- Obi - Google Voice
  - Block Private - Unknown - Restricted - ObiTalk - Google Voice
- PBX In a Flash (PIAF)
  - Disable IPTables PIAF
  - PIAF Default Root MYSQL PWD
- SHMZ 6.6
  - Upgrade FreePBX Distro 6 - 10 - As far as possible - Script
- VoIP Phones
  - Configure Panasonic KX-TGP500 - FreePBX - Asterisk - SIP DECT Wireless Phone
  - Configure YeaLink W52P-H
  - PolyCom
    - PolyCom SoundPoint 321 - Configuration
  - Port Landline to Google Voice