mod_com_amd

About

mod_com_amd is a FreeSWITCH™ commercial module used to analyze call progress and identify if it was answered by a human or a machine.

Introduction

FreeSWITCH™ provides a licensed commercial Answering Machine Detection module. A key technology for autodialers is the ability to detect live human pickup and answering machine. Since there is no indication/hardware signal when a call is answered by an answering machine or voicemail system, autodialer systems have to analyze incoming audio in order to make a prediction. Currently, there is no algorithm that can achieve 100% accuracy.

Getting Licenses

All FreeSWITCH™ purchases must be done through our billing system. Go to http://freeswitch.com, click on the top right menu called Account and then on Register.
After registering, go to login page at https://freeswitch.com/clientarea.php and provide your credentials. Once logged in the system, you'll see a menu on top of the page called Products, click and then click on Order New Products. Direct link is https://freeswitch.com/cart.php.
On the cart page, scroll down your screen and you'll see AMD product description. Click on the Order Now button. Direct link is https://freeswitch.com/cart.php?a=confproduct&i=1
On the next screen you need to type how many licenses you need. A license is needed for each created channel that you need to activate the detection. If you have 5 simultaneous calls and want to detect who’s answering you’ll need 5 ports/channels. Click on Add to Cart after selecting the number of licenses.
Now you can proceed to payment, and the end of the page there’s a Complete Order button. The payment method is PayPal, if you don’t have paypal account please proceed to PayPal’s registration page and follow the instructions.

If you need to activate licenses for different servers you need to buy them separately. For example, if you need 5 licenses to server A and more 10 to server B, add 5 licenses to the cart and then add 10 more again, you’ll receive two activation codes, one for each server.
After the payment, PayPal will notify FreeSWITCH™ billing system about the new payment and the license(s) will be sent to your email. This process takes a few minutes, if you do not receive the email in a few minutes please contact us at consulting@freeswitch.org or at #freeswitch at irc.freenode.net. For more information on how to contact us please visit IRC page.

Installing

Updated Instructions for FreeSWITCH 1.8

Install the package

```bash
apt-get install freeswitch-mod-com-amd
```

Uncomment or create in autoload_configs/modules.conf.xml

```xml
<load module="mod_com_amd"/>
```

Validate license
freeswitch-license-validator

freeswitch product licencing tool
You will require one or more sales codes to activate licences

Enter a sales code, or a blank line to end: 0abd19753bf28fbb88a5d74
Enter a sales code, or a blank line to end:

Sales codes to be activated:
0abd19753bf28fbb88a5d74
OK (Y/N)? y

Success. The file licences.zip contains valid licences.
Unzip this to /etc/freeswitch/

Unzip the key

unzip licences.zip

Archive: licences.zip
inflating: 0abd19753bf28fbb88a5d74.conf

Copy key to /etc/freeswitch (do not put the zip file in there)

cp 0abd19753bf28fbb88a5d74.conf /etc/freeswitch

Shutdown freeswitch

systemctl stop freeswitch

Start license server

freeswitch-license-server

Scanning /etc/freeswitch/0abd19753bf28fbb88a5d74.conf for licences
Purchase code 0abd19753bf28fbb88a5d74
10 channels of AMD

Start freeswitch again

systemctl start freeswitch

Check that amd is communicating with license server

fs_cli -x amd_info

Success checking AMD/0
AMD license available => 10
AMD license allocated => 0

Do some testing in dialplan. Download test wav file Answering_Machine.wav.
Checking for available license

notice how we check if license are available by checking a regex against zero using the amd_info api

<condition field="$expand regex $[amd_info]|available => 0" expression="false">

if regex is false, then proceed with amd usage, else if true, do anti actions to notify customer with a custom sound file

<anti-action application="playback" data="${sounds_dir}/All_Circuits_Busy.mp3"/>
<anti-action application="hangup"/>
Older Instructions for Freeswitch 1.6

We have packaged our commercial modules into a single installer. All products currently offered (mod_com_amd, mod_com_g729, mod_com_g719, mod_com_g728) are all installed and take up very little space.

### Downloading and Running Installer

```bash
cd /tmp
wget http://files.freeswitch.org/amd/fs-latest-installer
chmod +x fs-latest-installer
./fs-latest-installer
```

This will guide you thru some installation steps. We try to detect the most common places your FreeSWITCH™ installation may be located. If we can't detect it then we will prompt for information about where your FreeSWITCH™ is installed. You can optionally you can provide the data as arguments to the fs-latest-installer binary:

### Installer arguments

```
./fs-latest-installer <bin_path> <mods_path> <config_path>
```

Once you accept the EULA, you'll see output similar to this:

### Installing

```
Thank You!
Running Installer
Stopping license server
Installing new license server
Installing new FreeSWITCH module
Installing activation utility
Creating license directory
Running ldconfig...
Cleaning up...

Now you can activate your product license(s) by running /usr/local/freeswitch/bin/validator

Thank you

Done.
```

Once complete you'll need to activate your licenses using the validator outlined in the installation completion/thank you message. You'll use the same validator application for any of our commercial products that are available from our online store.

Once you start the validator you'll see output similar to this:
Licencing tool - Installing new licence.

freeswitch product licencing tool

You will require one or more sales codes to activate licences
Enter a sales code, or a blank line to end: 8c87ca579c46258a5e2ee9f98ca3f93ca4752d26
Enter a sales code, or a blank line to end:

Sales codes to be activated:
8c87ca579c46258a5e2ee9f98ca3f93ca4752d26

OK (Y/N)? Y
Success. The file licences.zip contains valid licences.
Unzip this to /etc/freeswitch/

The resulting licences.zip will be in the current working directory, Unzip and copy the .conf files into /etc/freeswitch:

Unzip Licenses

unzip licences.zip
Archive: licences.zip
  inflating: 8c87ca579c46258a5e2ee9f98ca3f93ca4752d26.conf

cp 8c87ca579c46258a5e2ee9f98ca3f93ca4752d26.conf /etc/freeswitch

Hardcode Path Here

/etc/freeswitch is a hard coded path the freeswitch_license_server will auto-scan up restart... You MUST put your <HEX>.conf file in /etc/freeswitch regardless of where you've installed FreeSWITCH.

To make the license process aware of your new licenses please run:

Reloading licence server

pkill -HUP freeswitch_license_server

This will make the license server rescan the directory adding any additional licenses to your list of resources its capable of licensing. We just activated a license for mod_com_amd which when loading the module will look like this:

Loading mod_com_amd

freeswitch@internal> amd_info
AMD license available => 20
AMD license allocated => 0
freeswitch@internal> amd_available
true
freeswitch@internal> amd_count
20
As you can see its seeing the licenses that were just installed, you can use the various api calls registered to manage your licenses.

Default Configuration

The default mod_com_amd configuration will be suitable in most cases, but if you need to tweak the configuration go to /usr/local/freeswitch/conf/autoload_configs and open amd.conf.xml:

```
<configuration name="amd.conf" description="AMD Configuration">
    <!-- AMD -->
    <settings>
        <!-- silent_threshold: The level of volume to consider talking or not talking, same scale as used in mod_conference -->
        <param name="silent_threshold" value="256"/>
        <!-- silent_initial: Time in ms for there to be silence after answer in order to result in "silent-initial" with status of person -->
        <param name="silent_initial" value="4500"/>
        <!-- silent_after_intro: Time in ms after an initial non silent greeting in order to result in silent-after-intro with status of person -->
        <param name="silent_after_intro" value="1000"/>
        <!-- silent_max_session: Time in ms of total silence before we allow detection to complete -->
        <param name="silent_max_session" value="200"/>
        <!-- noise_max_intro: Time in ms length of initial intro over which in order to result in max-intro with status of person -->
        <param name="noise_max_intro" value="1250"/>
        <!-- noise_min_length: Time in ms minimum to be considered a word -->
        <param name="noise_min_length" value="120"/>
        <!-- noise_inter_silence: Time in ms of silence to be considered a word break -->
        <param name="noise_inter_silence" value="30"/>
        <!-- noise_max_count: If we have more than this many noise hits (words) result will be "max-count" with status of machine -->
        <param name="noise_max_count" value="6"/>
        <!-- total_analysis_time: total time in ms that we will try to analyze a call -->
        <param name="total_analysis_time" value="5000"/>
        <!-- debug: set to 1 to get more debug information -->
        <param name="debug" value="1"/>
    </settings>
</configuration>
```

Overriding Config Values via Dialplan

```
<action application="voice_start" data="silent_threshold=96,
silent_initial=3500,
silent_after_intro=1500,
silent_max_session=200,
noise_max_intro=2250,
noise_min_length=175,
noise_inter_silence=50,
noise_max_count=10,
total_analysis_time=5000,
debug=1"/>
```

Examples

Let's figure out how can we use the mod_com_amd.

XML Dialplan Example

The first way you can use mod_com_amd is from the xml dialplan. When the module is loaded, three application are created:
Let's build a dialplan that when the call gets answered by a machine one message gets delivered to the voicemail:

**AMD Dialplan XML Example**

```xml
<extension name="amd_example" continue="false">
  <condition field="destination_number" expression="^5551212$">
    <action application="set" data="media_bug_answer_req=true"/>
    <action application="set" data="amd_execute_on_machine=transfer machine_detected XML default"/>
    <action application="voice_start"/>
    <action application="playback" data="/usr/local/freeswitch/sounds/en/us/callie/ivr/8000/ivr-welcome_to_freeswitch.wav"/>
    <action application="playback" data="/usr/local/freeswitch/sounds/en/us/callie/ivr/8000/ivr-welcome_to_freeswitch.wav"/>
    <action application="playback" data="/usr/local/freeswitch/sounds/en/us/callie/ivr/8000/ivr-welcome_to_freeswitch.wav"/>
    <action application="playback" data="/usr/local/freeswitch/sounds/en/us/callie/ivr/8000/ivr-welcome_to_freeswitch.wav"/>
  </condition>
</extension>

<extension name="Found machine">
  <condition field="destination_number" expression="^(machine_detected)$">
    <action application="wait_for_silence" data="300 30 5 25000"/>
    <action application="sleep" data="8000"/>
    <action application="playback" data="/usr/local/freeswitch/sounds/en/us/callie/ivr/8000/ivr-welcome_to_freeswitch.wav"/>
    <action application="info"/>
    <action application="hangup"/>
  </condition>
</extension>
```

Pay attention to the variable `amd_execute_on_machine` and the application `voice_start`. When the call gets answered, FreeSWITCH™ will route it to extension "amd_example". This extension will set the variable `amd_execute_on_machine`, which will transfer the call to destination "machine_detected" at context "default" if the call gets answered by a machine, otherwise the playback actions will get executed (dialplan lines 6, 7, 8 and 9).

If a machine answers this call, it'll be transferred to our next extension, `machine_detected`, which can be useful when you want to leave message in the customer voicemail, in this case you can use the application `wait_for_silence` to wait the end of voicemail initial greeting and then leave your message.

`mod_com_amd` sets a variable called "amd_status" and his value can be "human" or "machine". You can see the value with the "info" application (dialplan line 20).

In this first example, you can originate a call to 5551212 using:

### Originating call

```
bgapi originate {ignore_early_media=true}sofia/profile/number 5551212
```

This will make a call to `sofia/profile/number` and when answered FreeSWITCH™ will look for extension that matches 5551212 in your dialplan.

### Lua Example

If you're building your routing logic with Lua, you can use the same application and variables used earlier in the xml dialplan. Example:
dialer.lua

local dst_number = argv[1]
-- Connecting to the freeswitch API.
api = freeswitch.API()
use_amd = api:executeString("amd_available")

subscriber_session = freeswitch.Session("{ignore_early_media=true}sofia/gateway/mygw/ .. dst_number")

while (subscriber_session:ready() and not subscriber_session:answered()) do
    -- Waiting for answer.
    freeswitch.msleep(500)
end

if subscriber_session:ready() and subscriber_session:answered() then
    freeswitch.consoleLog("INFO", string.format("Number answered call %s.", dst_number))
    if use_amd == "true" then
        subscriber_session:execute("voice_start")
        -- Giving some time to AMD to work on the call.
        subscriber_session:sleep(3000)
        subscriber_session:execute("voice_stop")
        amd_detect = subscriber_session:getVariable("amd_status")
        if amd_detect == "machine" then
            freeswitch.consoleLog("INFO", "amd_status: machine")
            subscriber_session:execute("wait_for_silence", "300 30 5 25000")
            subscriber_session:hangup()
            return
        end
    -- Do your actions if human answered. Ex. Transfer to operator/user 100.
    subscriber_session:execute("bridge", "user/100")
end
else
    freeswitch.consoleLog("INFO", string.format("Cannot call %s", dst_number))
    return
end

Call this lua script using:

luarun dialer.lua 55552222

Testing Dialplan

XML for self testing the config setting... be sure to download the Answering_Machine.wav file and put in your sounds directory. With this setup, you should be able to tweak settings and get a feel for their values beyond the default.
<extension name="amd_test" continue="false">
  <condition field="destination_number" expression="^(amd_test|263)$">
    <action application="set" data="amd_execute_on_machine=transfer machine_found XML default"/>
    <action application="set" data="amd_execute_on_person=transfer person_found XML default"/>
    <action application="set" data="amd_execute_on Unsure=transfer amd Unsure XML default"/>
    <action application="voice_start"/>
    <action application="set" data="api_on_answer=uuid_displace ${uuid} start ${sounds_dir} /Answering_Machine.wav 0 mr"/>
    <action application="answer"/>
    <action application="waitforresult" data="ivr/ivr-one_moment_please.wav"/>
    <action application="sleep" data="200"/>
    <action application="playback" data="tone_stream://%(200,100,500,400,300,50,25);loops=2"/>
    <action application="sleep" data="200"/>
    <action application="log" data="CRIT AMD Result is => ${amd_status} => ${amd_result}"/>
    <action application="hangup"/>
  </condition>
</extension>

<extension name="Found Machine">
  <condition field="destination_number" expression="^(machine_found)$">
    <action application="playback" data="ivr/ivr-welcome_to_freeswitch.wav"/>
    <action application="log" data="CRIT AMD result Machine Found is => ${amd_status} => ${amd_result}"/>
    <action application="voice_stop"/>
    <action application="hangup"/>
  </condition>
</extension>

<extension name="Found Person">
  <condition field="destination_number" expression="^(person_found)$">
    <action application="playback" data="misc/if_you_are_this_person.wav"/>
    <action application="log" data="CRIT AMD result Machine Found is => ${amd_status} => ${amd_result}"/>
    <action application="voice_stop"/>
    <action application="hangup"/>
  </condition>
</extension>

<extension name="AMD Unsure">
  <condition field="destination_number" expression="^(amd_unsure)$">
    <action application="playback" data="misc/error.wav"/>
    <action application="log" data="CRIT AMD result Machine Found is => ${amd_status} => ${amd_result}"/>
    <action application="voice_stop"/>
    <action application="hangup"/>
  </condition>
</extension>

more technical info about mod_com_amd

uuid_displace have options [m]ux and [r]ead, so you can hear the ensuing found extension audio, and also induce the answering machine file onto the read channel so amd can operate on it.
mod_amd - Answering Machine Detection for FreeSWITCH

mod_amd supplies three different dialplan apps and several channel variables. It also fires events during the detection process. They are listed here for reference.

Apps:
voice_start - starts the answering machine detection on a channel
voice_stop - stops the answering machine detection on a channel
waitforresult [<file to play while waiting>] - waits for AMD to return a result, and optionally playback a file

Channel variables, results:
amd_status - Contains whatever was detected:
'person', 'machine', 'unsure'

Basically, anything other than 'machine' should be assumed to be a human

amd_result - Contains more information about what was detected:
'too-long' - detection process took longer than total_analysis_time (amd.conf.xml)
'max-count' - max noisy frames detected, probably a machine speaking
'max-intro' - max noisy frames detected during intro, probably a machine speaking
'silent-initial' - nothing heard on line, assume human
'silent-after-intro' - short burst of noisy frames followed by silence - probably human

Channel variables, control:
execute_on_machine_app - application to execute if machine detected
execute_on_machine_arg - argument to application to execute if machine detected

Events:
CUSTOM::AMD::EVENT - Action: Start Talking
CUSTOM::AMD::EVENT - Action: Stop Talking
SWITCH_MEDIA_BUG_ADD - Fired at start of detection process
SWITCH_MEDIA_BUG_REMOVE - Fired at stop of detection process

Look for variable amd_status header

USAGE
mod_amd can be used three ways:

via event socket, where an event-based script watches for AMD events and acts accordingly
via the supplied JavaScript file
via the dialplan by setting "execute_on_machine_app" and "execute_on_machine_arg"

See default.xml for simple examples