## Linux

## FreePBX Asterisk RTP and Call Monitoring Script

We created this script to assist with constant monitoring of RTP streams and calls in Asterisk/FreePBX. We typically save it to the root of a FreePBX system, name it "chanmonitor" and run chmod +x /chanmonitor to make it executable. It runs the following commands every two seconds and displays it in a formatted fashion.

To monitor just PJSIP:

```
#!/bin/bash
while true
do
    clear
    echo -e "\e[33mCurrent PJSIP Peers / Contacts:\e[39m"
    asterisk -rx 'pjsip show contacts'
    echo -e "\e[33mCurrent PJSIP Statistics / Active Calls:\e[39m"
    asterisk -rx 'pjsip show channelstats'
    sleep 2
done
```

To monitor SIP & PJSIP:

asterisk -rx 'sip show peers' asterisk -rx 'sip show channelstats' asterisk -rx 'pjsip show contacts' asterisk -rx 'pjsip show channelstats'

#!/bin/bash

```
while true
do
    clear
    echo -e "\e[33mCurrent Chan_SIP Peers:\e[39m"
    asterisk -rx 'sip show peers'
    echo -e "\e[33mCurrent Chan_SIP RTP Statistics / Active Calls:\e[39m"
    asterisk -rx 'sip show channelstats'
    echo ""
    echo -e "\e[33mCurrent PJSIP Peers / Contacts:\e[39m"
    asterisk -rx 'pjsip show contacts'
    echo -e "\e[33mCurrent PJSIP Statistics / Active Calls:\e[39m"
    asterisk -rx 'pjsip show contacts'
    echo -e "\e[33mCurrent PJSIP Statistics / Active Calls:\e[39m"
    asterisk -rx 'pjsip show contacts'
    echo -e "\e[33mCurrent PJSIP Statistics / Active Calls:\e[39m"
    asterisk -rx 'pjsip show channelstats'
    sleep 2
done
```

It looks like this when you run it.

## Linux

Current PJSIP Peers / Contacts:									
Contact:	<aor contacturi<="" td=""><td></td><td></td><td>&gt; <hash< td=""><td>&gt; <statu< td=""><td>s&gt; <rtt(ms)< td=""><td>.&gt;</td><td></td></rtt(ms)<></td></statu<></td></hash<></td></aor>			> <hash< td=""><td>&gt; <statu< td=""><td>s&gt; <rtt(ms)< td=""><td>.&gt;</td><td></td></rtt(ms)<></td></statu<></td></hash<>	> <statu< td=""><td>s&gt; <rtt(ms)< td=""><td>.&gt;</td><td></td></rtt(ms)<></td></statu<>	s> <rtt(ms)< td=""><td>.&gt;</td><td></td></rtt(ms)<>	.>		
							=		
Contact:	201/sip:201098.153	3.58.190:1217;x-	ast-orig-host	d740a87	efb Avail	24.923	3		
Contact:	202/sip:202098.153	3.58.190:1230;x-	ast-orig-host	c Ofeacac	99e Avail	21.963	3		
Contact:	203/sip:203@98.153	3.58.190:1213;x-	ast-orig-host	: b673194	01d Avail	23.850			
Contact:	204/sip:204@98.153	3.58.190:1233;x-	ast-orig-host	: be80d34	7el Avail	24.246	5		
Contact:	Contact: 205/sip:205@98.153.58.190:64070;x-ast-orig-hos 54bbc31379 Avail 23.005								
Contact:									
Contact:	209/sip:209098.153					23.361			
Contact:	Flowroute/sip:0449	96083@us-west-or	.sip.flowrout	: 47cfa74	d15 Avail	26.462	2		
Objects found: 8									
Current PJSIP Statistics / Active Calls:									
ReceiveTransmit									
BridgeId C	hannelId (	UpTime Codec.	Count Lo	ost Pct	Jitter C	ount Lost	Pct Jitt	er RTT	
1c2f7528 F	lowroute-0000000a (	00:00:17 ulaw	899		0.000	523 0	0 0.0	00 0.058	
Objects found: 1									

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