


```

== Using SIP RTP CoS
-- Executing [22168434@incoming:1] Answer("SIP/202.101.42.32-09ce0e30", "") in new stack
-- Executing [22168434@incoming:2] Background("SIP/202.101.42.32-09ce0e30", "demo-thanks")
in new stack
-- <SIP/202.101.42.32-09ce0e30> Playing 'demo-thanks.gsm' (language 'en')
-- Auto fallthrough, channel 'SIP/202.101.42.32-09ce0e30' status is 'UNKNOWN'

```

???? extensions.conf ??

[general]

```

...
autofallthrough=no

```

????? dialplan

```

...
exten => 22168434,n,Background(demo-thanks)
exten => 22168434,n,WaitExten()    <===== here
exten => _101,1,Playback(vm-nobodyavail)
...

```

[PIAF]????????OOH323/(null)-cc0dws322

Ans: ?????? H323 ?????????????? modules.conf?????????

noload => chan_oo323.so

?? Asterisk ???

????

- [How To Debug and Troubleshoot VOIP::Voip-Info.org](http://Voip-Info.org)