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????? Incoming Calls ???????????? sip.conf ?? call-limit=xx ????????????

1. ??????? call-limit ???????????
2. ???????? call-limit ??????? SIP ???? 480?????? 503 ???????? chan_sip.c ??????????? asterisk?
3. ?? call-limit ?????? chan_sip.c ??????????????????

? DP ??? call-limit ????

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?? /etc/asterisk/extensions_custom.conf

```
; Below is used to limit the maximum number of incoming calls for trunk ccnet
;
[my-from-trunk-call_limit-ccnet]
exten => _.,1,Set(DIAL_TRUNK=2) ;2 is trunk# for ccnet, you can get it on FreePBX
exten => _.,n,Set(GROUP()=OUT_${DIAL_TRUNK})
exten => _.,n,Noop(Calls usage(activated): ${GROUP_COUNT(OUT_${DIAL_TRUNK})})
exten => _.,n,GotoIf($["${OUTMAXCHANS_${DIAL_TRUNK}}foo" = "foo"]?nomax)
exten => _.,n,GotoIf($[ ${GROUP_COUNT(OUT_${DIAL_TRUNK})} > ${OUTMAXCHANS_${DIAL_TRUNK}} ]?chanfull)
exten => _.,n(nomax),Noop(Passing the call to from-trunk)
exten => _.,n,Goto(from-trunk,${EXTEN},1)
exten => _.,n(chanfull),Noop(max channels used up)
exten => _.,n,Hangup(42) ;42 is causecode, which will return 503 response code
exten => h,1,Hangup
```

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- [Asterisk Hangupcase](#)
- [Hangup Cause Code](#)