

bridge

Bridge a new channel to the existing one. Generally used to route an incoming call to one or more endpoints. Multiple endpoints can be dialed simultaneously or sequentially.

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```
{
  "bridge": {
    "strategy": "multiple",
    "pickup": "mygroup",
    "parameters": ["a=1", "instant_ringback=true"],
    "endpoints": [{
      "name": "gw_name1",
      "type": "sipGateway",
      "dialString": "&reg0.$1",
      "parameters": ["absolute_codec_string='PCMA, G729'"]
    },
    {
      "name": "1000",
      "type": "device"
    },
    {
      "name": "1001",
      "type": "user",
      "domainName": "10.10.10.144",
      "parameters": ["d=3"]
    }
  ]
}
```

Value	Description
strategy	• multiple - no limit to concurrency, first one to answer wins. • failover - no limit to failover number.
pickup	pickup group name. The pickup endpoint is a dummy channel that never answers to which you can originate from anywhere you can place calls.

parameters	Set variables to all endpoints. Some useful variables: • instant_ringback=true - ringback will not wait for indication before sending ringback tone to calling party. • ignore_early_media=true - Ignore early media from the endpoint. • ignore_early_media=ring_ready - The same as ignore_early_media=true but also send a SIP 180 to the inbound leg when the first SIP 183 is caught. • call_timeout=20 - Controls how long (in seconds) to ring the endpoint. Default is 60 seconds. • leg_timeout=15 - Can be used inside endpoints parameters only. • leg_delay_start=15 - Specifies a wait time in seconds before the leg is called. Can be used inside endpoints parameters only. • origination_caller_id_number - Sets the origination CallerID number. • absolute_codec_string - Sets the absolute codec string to use (nothing will be appended). • sip_renegotiate_codec_on_reinvite=true - Allow SDP codec change with re-INVITE.
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Endpoints

sipGateway

Value	Description
name	SIP gateway name.
dialString	Dialed number to this endpoint. ®0.\$1 = destination number
parameters	Set variables to this endpoint.

user

Value	Description
name	User name.
domainName	User domain name.
parameters	Set variables to this endpoint.

device

Value	Description
name	Device name in current domain.
parameters	Set variables to this endpoint.