Nexmo doesn't seem to handle it well if your first specified codec is L16. Try to set `absolute_codec_string` to PCMU and see if that helps.

Also note that there is no need to include custom SIP headers such as api_key, api_secret, and answer_url when you make an outbound call.

Since you mentioned also needing inbound - keep in mind that when you use Nexmo's built-in "Forward to SIP" setting for each number in the dashboard, the dialed number will not be passed as a SIP variable and you have no way of knowing it once you receive the SIP invite. One way to get around this is to have your application buy & update numbers via the Nexmo API and set a custom SIP address per Nexmo DID, for example: `nexmo_12121115555@your-server.com` and then have a dialplan such as:

```xml
<extension name="IncomingNexmo">
    <condition field="destination_number" expression="^nexmo_(\d+)$">
        <action application="info"/>
        <action application="set" data="nexmo.forwarded_for=$1"/>
        <action application="lua" data="nexmo.handler.lua"/>
    </condition>
</extension>
```

The `nexmo_forwarded_for` session variable will now expose to you the dialed Nexmo phone number allowing your application or XML dialplan to use it.