

Calls & Meetings API Documentation

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Base URL: `http://localhost:8765/api/v1` (local) — `server.port=8765`

Short Description: Real-time voice and video inside the chat system. Two surfaces sit here: **Calls**, which ring somebody now, and **Meetings**, which are scheduled or open rooms people join. Both are signalling only — this API issues room credentials and records what happened; the media itself never touches the backend, it goes phone → LiveKit SFU → phone.

Hints:

- Every endpoint needs `Authorization: Bearer <token>`.
- A join token is for **one room, one identity, and expires in 10 minutes** (`app.livekit.token-ttl-seconds=600`). Fetch it when you are about to connect, not when you render a list.
- Refusals are `409` with a machine-readable `code`. Branch on the code, never on the prose.
- The server owns call state and learns it from **LiveKit webhooks**, not from clients. A phone that dies mid-call never tells anyone it left.
- An unanswered call becomes `MISSED` after **45 seconds** (`app.livekit.ring-timeout-seconds=45`).

Standard Response Format

Success responses use the Globe Response Builder envelope:

```
{
  "success": true,
  "httpStatus": "OK",
  "message": "Call started",
  "action_time": "2026-08-20T10:30:45",
  "data": { }
}
```

Refusals — read this before writing a client

A business refusal is **HTTP 409 CONFLICT**, and the envelope's `message` carries the **code**, with the code repeated inside `data`:

```
{
  "success": true,
  "httpStatus": "CONFLICT",
  "message": "BUSY",
  "action_time": "2026-08-20T10:30:45",
  "data": { "code": "BUSY", "reason": "That person is already in a call" }
}
```

Note `success` stays `true` — the envelope reports transport, not business outcome. **Treat any 409 as a refusal and read `data.code`.**

Part 1 — Calls

`api/v1/calls` — invite, accept and end are REST; the incoming alert is SSE plus push; media never comes near this controller.

The shape of a call

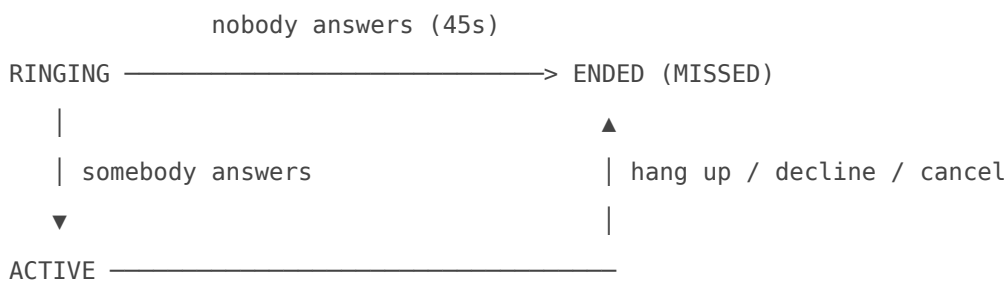
CALLER	BACKEND	CALLEE
-- POST /calls ----->		
{kind, one target}	create call (RINGING)	
	mint caller token	
<-- 200 CallTokenResponse ----		
	== SSE call.incoming =====>	(device online)
	== push notification =====>	(device asleep)
	both carry the CALLEE token	
---- connect to LiveKit ----->	[SFU]<---- connect -----	
	LiveKit webhook: participant joined	
	state -> ACTIVE	

```
| -- POST /{id}/end ----->| outcome COMPLETED |
|                               |== SSE call.ended =====>|
|                               | writes CALL_EVENT to thread |
```

Two things worth saying out loud in a demo:

1. The `call.incoming` event **carries the callee's token in the payload**. A ringing phone that had to fetch one first would add a round trip to answer latency. `POST /{callId}/join` exists only for a callee that learned about the call some other way, or cold-started.
2. A call always ends up attached to a conversation — that is where its `CALL_EVENT` message goes and what the permission check reads — but the caller never has to create one first.

Call state machine



`CallState` only ever advances. States: `RINGING`, `ACTIVE`, `ENDED` (plus `SCHEDULED` and `OPEN`, which belong to meetings).

`CallOutcome` is set once, on entry to `ENDED`, and copied into the thread's `CALL_EVENT`:

Outcome	Meaning
COMPLETED	reached <code>ACTIVE</code> — somebody answered
MISSED	rang out past the 45s timeout
DECLINED	callee explicitly refused
CANCELLED	caller hung up before it was answered
FAILED	could not be set up at all
BUSY	callee already in another call

`CANCELLED` and `MISSED` are deliberately separate — "I gave up" and "you never picked up" are different facts, and collapsing them loses the one the caller cares about.

POST /api/v1/calls — Start a call

Rings somebody. Returns the caller's own room credentials.

Request (`StartCallRequest`) — exactly **one** target must be present:

Field	Type	Required	Description
<code>conversationId</code>	UUID	one-of	Calling from an open thread
<code>targetShopId</code>	UUID	one-of	Calling a shop cold from its page; resolves or creates the commerce thread
<code>targetAccountId</code>	UUID	one-of	Calling a person you may not have messaged yet
<code>kind</code>	enum	yes	<code>AUDIO</code> or <code>VIDEO</code>
<code>shopId</code>	UUID	no	Present when calling as a shop (different from <code>targetShopId</code>). Validated against the caller's real contexts, never trusted

```
curl -s -X POST $BASE/api/v1/calls \
  -H "Authorization: Bearer $TOKEN" -H 'Content-Type: application/json' \
  -d '{"targetAccountId":"...","kind":"AUDIO"}'
```

Response (`CallTokenResponse`):

Field	Type	Description
<code>callId</code>	UUID	
<code>roomName</code>	string	LiveKit room
<code>token</code>	string	One room, one identity, 10-minute TTL
<code>url</code>	string	LiveKit websocket URL
<code>kind</code>	enum	<code>AUDIO</code> / <code>VIDEO</code>
<code>state</code>	enum	<code>RINGING</code>

`kind` is fixed for the call's lifetime. A video call that drops to audio is still a `VIDEO` call that degraded, and the **client** decides that, not the server.

POST /api/v1/calls/{callId}/join — Answer

For a callee that learned about the call over SSE or a cold start. A push-woken client already has its token in the payload and **skips this round trip entirely**.

Param	In	Required	Description
<code>callId</code>	path	yes	
<code>shopId</code>	query	no	Answering as a shop

Returns `CallTokenResponse`.

POST /api/v1/calls/{callId}/end — Hang up, decline, or cancel

All three are one verb. Which of the three it was is decided by the call's state and who asked — not by the client naming it — so a client cannot report a call it declined as one it completed.

Param	In	Required
<code>callId</code>	path	yes
<code>shopId</code>	query	no

Returns `CallResponse`.

POST /api/v1/calls/{callId}/participants — Pull somebody in

Adds a person to a call already running. **Initiator only, personal calls only.**

Param	In	Required
<code>callId</code>	path	yes
<code>accountId</code>	query	yes

Refuses with `CALL_T00_LARGE` past `app.livekit.group-max-participants` — a hard refusal rather than ringing the first N, because silently dropping members of a group call is worse than saying no:

nobody can tell who was left out.

GET /api/v1/calls — Call history

The Calls list. **Keyed on who was ON the call** rather than on conversations, so a call somebody added you to appears here even though its thread is not one of yours.

Param	In	Default
shopId	query	— (personal)
limit	query	30

Returns `CallResponse[]`.

GET /api/v1/calls/{callId} — One call

Returns `CallResponse`. **Carries no token** — a client that wants to join asks for one, so a stale poll cannot hand out live credentials.

CallResponse

Field	Type
callId	UUID
conversationId	UUID
kind	AUDIO / VIDEO
state	RINGING / ACTIVE / ENDED
outcome	see table above
initiatorJid	string
createdAt, answeredAt, endedAt	ISO-8601
durationSeconds	long

Call refusal codes

Code	When
------	------

AMBIGUOUS_TARGET	zero or more than one target given
NO_COUNTERPARTY	nobody on the other end to ring
BUSY	callee already in another call
ALREADY_IN_CALL	the caller is
BLOCKED	blocked in either direction
CALL_TOO_LARGE	past the participant cap
CALL_ENDED	joining or acting on a finished call
NOT_A_PARTICIPANT	you were never on it
NOT_INITIATOR	only the initiator may add participants
NOT_A_PERSONAL_CALL	add-participant is personal-calls-only
NOT_INVITED	not on the guest list

Part 2 — Meetings

api/v1/meetings — scheduled and open calls. Separate from calls because **the verbs differ**: you schedule and join a meeting; you ring and answer a call.

A meeting is a `CallEntity` too, so it shares the state machine — but it is joined rather than answered, so it **never rings and can never be MISSED**.

SCHEDULED —(its window opens)—> OPEN —(host ends it)—> ENDED

OPEN means the room is live for its window — and **stays open when empty**, since somebody looking in at 16:58 must not end the 17:00 meeting.

Meeting flow

HOST	BACKEND	INVITEES
-- POST /meetings ----->	state SCHEDULED	
{title, scheduledAt, invitees}		
	== SSE meeting.scheduled ==>	(once, for the
calendar)		
	[MeetingScheduleJob: window opens]	

```

|                                     | state OPEN                                     |
|                                     |== SSE meeting.starting ==>| (NO token – see below)
|                                     |                                     |
|-- POST /{id}/join ----->|<---- POST /{id}/join -----|
|<-- CallTokenResponse ----- |----> CallTokenResponse --->|
|                                     |                                     |
|-- POST /{id}/end ----->| state ENDED                                     |

```

`meeting.starting` **carries no token**, unlike `call.incoming`. A meeting is joined when the user decides to, not by a phone deciding for them.

Roles — a bandwidth decision, not ceremony

Role	May
<code>HOST</code>	speak, promote, demote, end
<code>SPEAKER</code>	publish; shares the publisher cap
<code>LISTENER</code>	subscribe only

A speaker costs an uplink and one decode on every other device; a listener costs one downstream stream and no uplink at all. That is why the participant cap counts **speakers** and lets listeners run far higher — and why the role is **enforced in the token grant** rather than asked of the client.

Access

<code>CallAccess</code>	Meaning
<code>INVITED</code>	a guest list — behaves like a group call arranged in advance (default)
<code>OPEN</code>	anyone holding a join link — but still a NexGate account, because a participant without a JID is unaddressable and unblockable

`POST /api/v1/meetings` — Schedule a meeting

Request (CreateMeetingRequest):

Field	Type	Required	Description
title	string	yes	A call needs no name because nobody labels "phoning Asha"; a meeting does, because everybody labels "Friday pricing review"
description	string	no	
scheduledAt	ISO-8601	no	Absent means now
endsAt	ISO-8601	no	When the room closes itself. Approximate by nature — not a promise about when people stop talking
kind	enum	no	AUDIO / VIDEO
access	enum	no	INVITED (default) or OPEN
conversationId	UUID	no	Anchored to a thread → writes its CALL_EVENT there; standalone → recorded in GET /api/v1/calls instead
inviteeAccountIds	UUID[]	no	

Returns CallResponse with state: SCHEDULED .

POST /api/v1/meetings/{meetingId}/join — Join

Returns CallTokenResponse . The token's grants encode the caller's role.

POST /api/v1/meetings/{meetingId}/links — Create a join link

Role-scoped: a speaker link and a listener link are different links.

Param	In	Default
-------	----	---------

meetingId	path	—
role	query	LISTENER

Response:

```
{ "token": "...", "role": "LISTENER", "expiresAt": "2026-09-19T..." }
```

Links live **30 days**. You cannot mint a `HOST` link — that refuses with `CANNOT_SHARE_HOST`.

POST /api/v1/meetings/join/{token} — Join by link

Path takes the link token. Still requires an authenticated NexGate account. Returns `CallTokenResponse`.

PUT /api/v1/meetings/{meetingId}/participant s/role — Promote / demote

Param	In	Required
jid	query	yes
role	query	yes (<code>HOST</code> / <code>SPEAKER</code> / <code>LISTENER</code>)

Host only.

POST /api/v1/meetings/{meetingId}/end — End it

Returns `CallResponse` with `state: ENDED`.

Meeting refusal codes

Code	When
<code>TITLE_REQUIRED</code>	no title
<code>NOT_A_MEETING</code>	the id is a call, not a meeting
<code>NOT_HOST</code>	host-only verb
<code>NOT_INVITED</code>	<code>INVITED</code> access and you are not on the list
<code>MEETING_ENDED</code>	joining a finished meeting
<code>LINK_EXPIRED</code>	past the 30-day link lifetime
<code>SPEAKER_LIMIT</code>	stage is full; join as a listener
<code>CANNOT_SHARE_HOST</code>	you may not mint a HOST link
<code>BLOCKED</code>	blocked in either direction

Part 3 — The real-time channel

GET `/api/v1/events` (SSE)

One stream covers **every context the caller may act for**, so the context switcher and the permission check read the same source and cannot disagree.

Client note: the browser `EventSource` API cannot set an `Authorization` header, so **web must use fetch-based streaming**, not `EventSource`. Putting the token in the query string would leak credentials into access logs and proxy history — a poor trade for a few lines of client code.

Resume with `Last-Event-ID` (header wins) or `?lastEventId=`.

Call and meeting events

Event	Carries a token?	Payload
<code>call.incoming</code>	yes	<code>callId</code> , <code>conversationId</code> , <code>kind</code> , <code>from</code> , <code>roomName</code> , <code>url</code> , <code>token</code>
<code>call.updated</code>	no	
<code>call.ended</code>	no	<code>callId</code> , <code>conversationId</code> , <code>outcome</code> , <code>durationSeconds</code>
<code>meeting.scheduled</code>	no	<code>meetingId</code> , <code>title</code> , <code>scheduledAt</code> , <code>hostJid</code> , <code>conversationId?</code>
<code>meeting.starting</code>	no , on purpose	

`call.ended` goes to **everyone, including whoever hung up** — their other devices are still ringing otherwise.

Signalling fires `AFTER_COMMIT` throughout: alerting someone about a call whose row rolled back would ring a phone for a call that does not exist. SSE reaches a device that already holds a connection; push wakes one that does not. **Both go out** — a device with a live stream ignores the duplicate, and guessing wrong means a call nobody hears.

`POST` `/api/v1/calls/livekit/webhook` — internal

LiveKit → backend. **This is what actually moves a call to `ACTIVE` and to `ENDED`.** Not for clients; signed by LiveKit.

Configuration

Property	Default	Meaning
<code>app.livekit.url</code>	<code>ws://127.0.0.1:7880</code>	SFU websocket
<code>app.livekit.token-ttl-seconds</code>	<code>600</code>	Join token lifetime
<code>app.livekit.ring-timeout-seconds</code>	<code>45</code>	Unanswered → <code>MISSED</code>
<code>app.livekit.group-max-participants</code>	<code>8</code>	Group call / meeting speaker cap

The `CallTimeoutJob` runs every 15 seconds against the 45-second timeout, so a missed call settles within a minute. The client stops its own ringing at 45s regardless.

Revision #1

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