

Private Chat & Calls (DEEP)

Phase 2 Deep Dive

NexGate / QBIT SPARK | Version 1.0 *1:1 DMs · Group Chats · Voice Calls · Video Calls · Ejabberd · WebRTC*

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1. Scope

This document covers only private communication features in Phase 2:

- IN SCOPE:
- 1:1 private DMs (personal + shop commerce DMs)

Group chats – private + public (up to 500 members)

Group join model (consent DM + invite link)

Voice calls (1:1 + group)

Video calls (1:1 + group)

Audio ↔ video switching during calls

Screen sharing

Chat states (typing, recording voice note)

Message interactions (edit, delete, react, forward, reply)

Message receipts (sent, delivered, read)

Presence (online, offline, last seen)

Multi-device support

Offline delivery

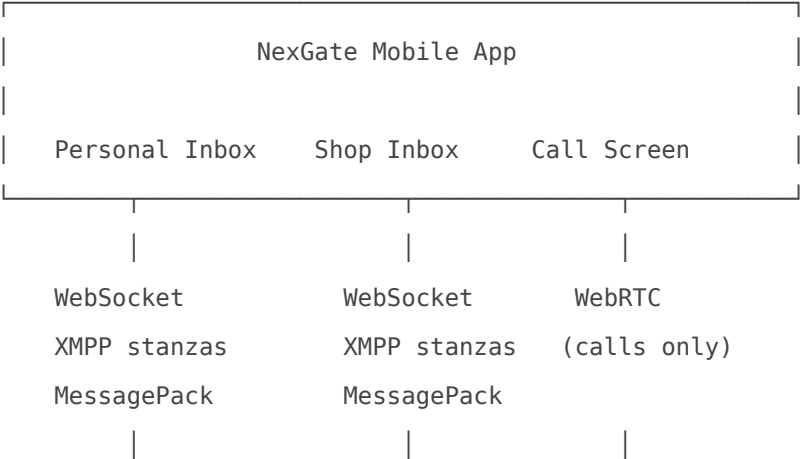
- OUT OF SCOPE:
- Broadcast channels → NOT built (VP Feed covers this – see Doc 6)

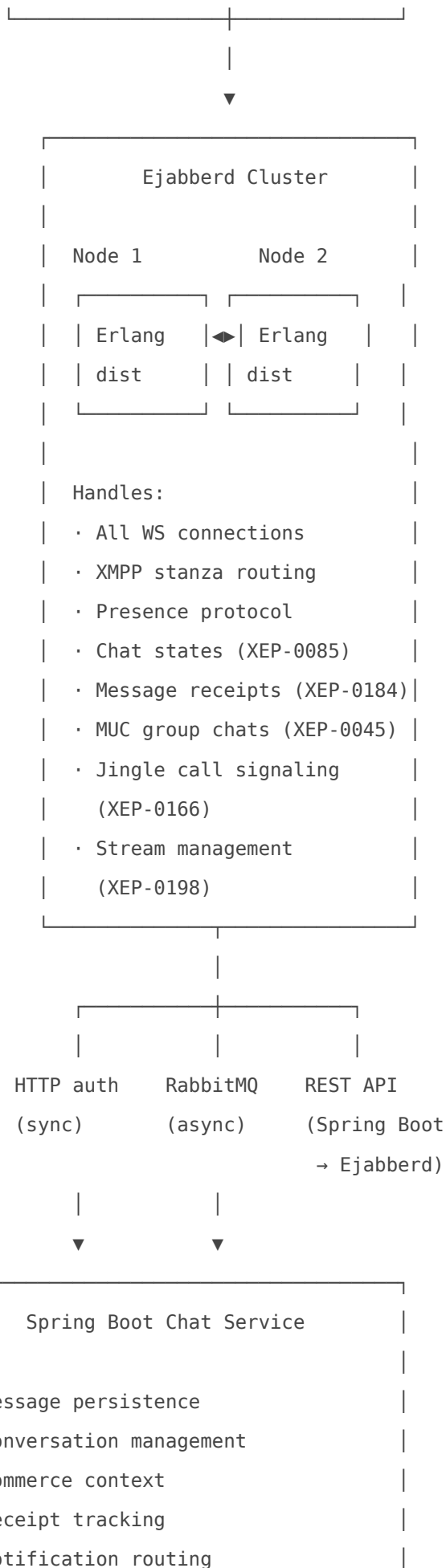
VP Live streaming → covered in VP Live doc

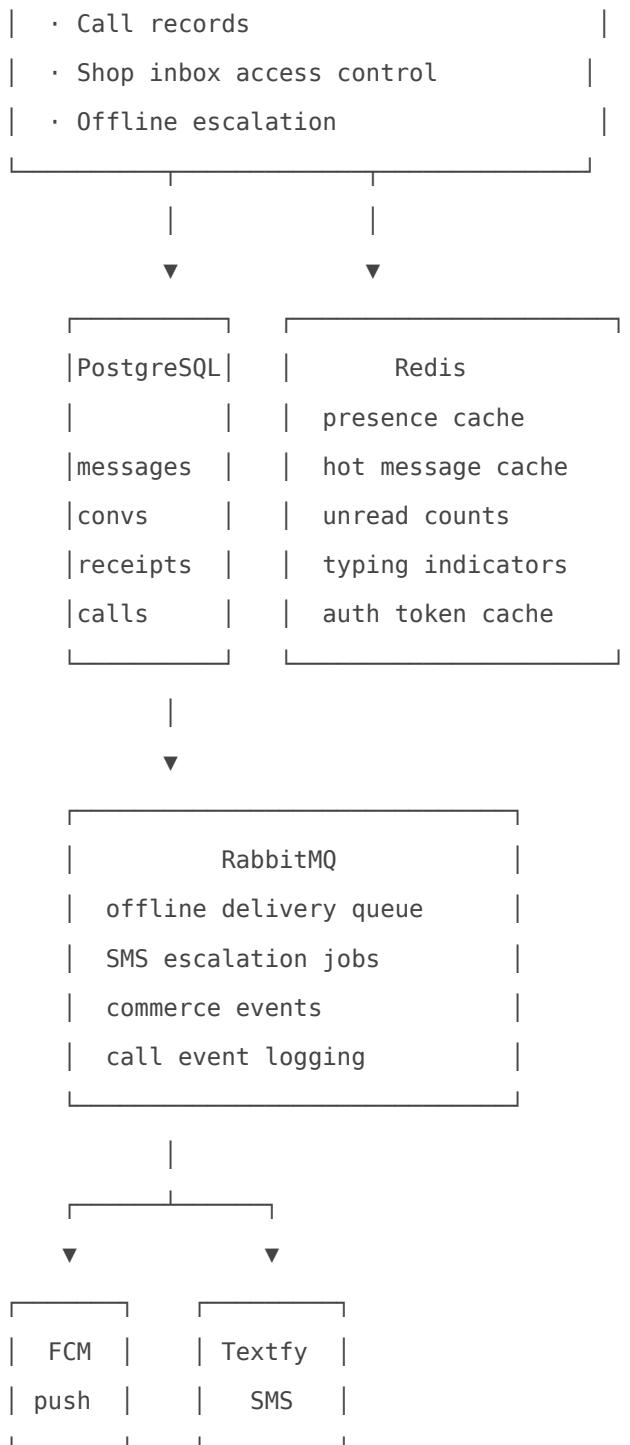
VP Audio Spaces → covered in VP Live doc

File Thunder → covered in File Thunder docs

2. Architecture Overview







Also:

Coturn TURN server (separate VPS)

→ relay for voice/video calls

→ when EA carrier NAT blocks P2P

3. XMPP & Ejabberd Fundamentals

JID — Every Entity Has an Address

In XMPP every connected entity has a JID (Jabber ID)

Works like an email address for messaging

Personal user (full JID):

kibuti@nexgate.com/android

| | |

user domain resource (device)

Personal user (bare JID):

kibuti@nexgate.com

(without device – used for addressing)

Shop identity:

techstore@shops.nexgate.com

(the shop – not the person behind it)

System bot:

system@nexgate.com

(order updates, notifications)

Group chat room:

group-abc@conference.nexgate.com

Multi-device – same user, multiple resources:

kibuti@nexgate.com/android ← phone

kibuti@nexgate.com/tablet ← tablet

Both receive messages simultaneously

READ on one → Ejabberd notifies other to clear notification

XEPs — XMPP Extension Protocols

XMPP base protocol = just message/presence/iq stanzas

XEPs add specific capabilities on top

XEPs enabled for NexGate private chat:

XEP-0045	Multi-User Chat (MUC) → group chats up to 500 members
XEP-0085	Chat State Notifications → typing indicators, recording indicators
XEP-0184	Message Delivery Receipts → sent / delivered ticks
XEP-0198	Stream Management → reliable delivery on bad networks → reconnect without losing messages → ACK at stanza level
XEP-0166	Jingle → voice and video call signaling
XEP-0357	Push Notifications → FCM/APNs bridge when user offline
XEP-0333	Chat Markers → read receipts (blue ticks)
XEP-0280	Message Carbons → sync messages across multiple devices

Three Stanza Types — Everything Is One of These

```

<!-- 1. Message – send content -->
<message from="kibuti@nexgate.com"
  to="juma@nexgate.com"
  type="chat"
  id="msg-001">
  <body>Habari yako!</body>
</message>

<!-- 2. Presence – announce availability -->
<presence from="kibuti@nexgate.com/android">

```

```
<show>available</show>
<status>I am here</status>
</presence>

<!-- 3. IQ (Info/Query) – request/response -->
<iq type="get" id="req-001">
  <query xmlns="jabber:iq:roster"/>
  <!-- asking for contact list -->
</iq>
```

4. Ejabberd Cluster — Two Nodes

Why Two Nodes Over One

Running a single Ejabberd node works technically. But one node means one point of failure. If that container crashes or the VPS reboots during a deployment — every connected user loses their session, every active call drops, every in-flight message is lost.

Two nodes change the picture completely:

Single node:

- Node 1 crashes
- 100% of users disconnected
- all active calls dropped
- messages in-flight lost
- users notice immediately

Two nodes:

- Node 1 crashes
- 50% of users reconnect to Node 2 (seconds)
- Node 2 was already running – no cold start
- active calls on Node 2 unaffected
- Ejabberd cluster detects Node 1 gone
- routes everything to Node 2 automatically
- most users experience a brief reconnect
- not a full outage

Two nodes also doubles the connection capacity:

One node → ~500k-1M concurrent connections

Two nodes → ~1M-2M concurrent connections

Same cost increase as one extra container

Launch Plan — Same VPS, Two Containers

For NexGate launch, both nodes run on the same Hetzner VPS. This is the right starting point:

- Cheaper — one VPS bill not two
- Simpler — same Docker network, zero latency between nodes
- Enough — two containers on one VPS still gives redundancy against container crashes and restarts
- Learning — operate cluster on familiar single VPS first
- △□ VPS hardware failure → both nodes gone
(acceptable risk at launch stage)

When to move to two VPS:

NexGate has paying users depending on uptime
VPS hardware failure = real revenue loss
At that point: Option B (two VPS) is worth the cost

What is Erlang Dist?

This is the mechanism that makes the two containers feel like one system.

Erlang was designed in 1986 for telecom — specifically for telephone switches that could never go down even when individual machines failed. The solution Ericsson built was **Erlang Distribution**: multiple Erlang nodes connected over a network, sharing a process registry, able to send messages between processes on different machines as if they were local.

Normal programming:

Process on Machine A cannot directly talk to
process on Machine B
Need: HTTP, gRPC, message queue, shared DB
Always an extra hop

Erlang distribution:

Process on Node 1 sends message to process on Node 2
Directly — like calling a local function

No extra infrastructure

No Redis, no RabbitMQ for this

Just: `node1_process ! { message_to, node2_process }`

Erlang runtime handles delivery across the network

Applied to Ejabberd:

Every connected user = one Erlang process (~2KB RAM)

Kibuti connected to Node 1 = process on Node 1

Juma connected to Node 2 = process on Node 2

Kibuti sends "Habari" to Juma:

Node 1 (Erlang):

"Find Juma's process"

Check local process registry → not here

Check Node 2 via Erlang dist → FOUND

Send message directly to Juma's process on Node 2

Node 2 delivers to Juma's WebSocket

No Redis pub/sub

No RabbitMQ for this routing

No extra network hops

Microsecond latency between nodes

This is why Ejabberd routes at 2M concurrent

where Spring Boot WS needs Redis pub/sub

Erlang Cookie — The Cluster Password

Before two Erlang nodes trust each other

they must prove they belong to the same cluster

The shared secret = Erlang Cookie

Node 1 starts → "my cookie is: nexgate_erlang_cookie_xyz"

Node 2 starts → "my cookie is: nexgate_erlang_cookie_xyz"

Same cookie → they trust each other → cluster formed

Unknown node attempts to join:

```
"my cookie is: wrong_cookie"  
→ rejected → cannot join cluster
```

Rules:

- Same cookie on ALL nodes – mandatory
- Long random string – not a simple word
- Stored in HashiCorp Vault → injected as env variable
- Never committed to git
- Rotate periodically like any secret

How Nodes Discover and Join Each Other

Step 1 – EPMD (Erlang Port Mapper Daemon):

- Each Erlang node registers with EPMD on port 4369
- EPMD is like a local DNS for Erlang nodes
- "I am ejabberd@ejabberd-node1, listening on port X"

Step 2 – Node 2 finds Node 1:

- Node 2 asks EPMD on Node 1's host:
- "Where is ejabberd@ejabberd-node1?"
- EPMD responds with port number
- Node 2 connects directly

Step 3 – Cookie handshake:

- Node 2: "here is my cookie hash"
- Node 1: validates → matches → accept
- Erlang dist connection established

Step 4 – Join cluster:

- `ejabberdctl join_cluster ejabberd@ejabberd-node1`
- Nodes sync:
 - MUC room state
 - User session registry
 - Mnesia tables (Ejabberd internal DB)
- Cluster ready ☐

In Docker – hostname is critical:

- Container hostname must match Erlang node name
- `ejabberd@ejabberd-node1` → container hostname: `ejabberd-node1`

Mismatch = nodes cannot find each other

Docker Compose — Two Nodes on Same VPS

```
ejabberd-node1:
  image: ghcr.io/processone/ejabberd:latest
  container_name: ejabberd-node1
  hostname: ejabberd-node1          # must match ERLANG_NODE
  restart: unless-stopped
  environment:
    - ERLANG_NODE=ejabberd@ejabberd-node1
    - ERLANG_COOKIE=${EJABBERD_ERLANG_COOKIE} # from Vault
  ports:
    - "5222:5222"    # XMPP TCP
    - "5280:5280"    # WebSocket
    - "5285:5285"    # REST API (internal)
    - "1883:1883"    # MQTT
    - "4369:4369"    # EPMD (Erlang port mapper)
  volumes:
    - ./ejabberd/ejabberd.yml:/home/ejabberd/conf/ejabberd.yml
    - ./ejabberd/node1/data:/home/ejabberd/database
    - ./ejabberd/node1/logs:/home/ejabberd/logs
  networks:
    - nexgate-internal

ejabberd-node2:
  image: ghcr.io/processone/ejabberd:latest
  container_name: ejabberd-node2
  hostname: ejabberd-node2          # different hostname
  restart: unless-stopped
  environment:
    - ERLANG_NODE=ejabberd@ejabberd-node2
    - ERLANG_COOKIE=${EJABBERD_ERLANG_COOKIE} # same cookie
  ports:
    - "5223:5222"    # different host ports
    - "5281:5280"
    - "5286:5285"
    - "4370:4369"
  volumes:
```

```
- ./ejabberd/ejabberd.yml:/home/ejabberd/conf/ejabberd.yml
- ./ejabberd/node2/data:/home/ejabberd/database
- ./ejabberd/node2/logs:/home/ejabberd/logs

depends_on:
  - ejabberd-node1

networks:
  - nexgate-internal
```

How Traefik Load Balances Between Nodes

Traefik sits in front of both nodes:

chat.nexgate.com → Traefik → Node 1 or Node 2

Critical: WebSocket needs sticky sessions

Once a user connects to Node 1 – they must
always go to Node 1 for that session
(the WS connection lives on that node)

Traefik labels:

```
sticky.cookie: true
sticky.cookie.name: "ejabberd_node"
```

First connection:

```
User hits chat.nexgate.com
Traefik picks Node 1 (round robin)
Sets cookie: ejabberd_node=node1
User connects WebSocket to Node 1
```

Subsequent requests same session:

```
Cookie present: ejabberd_node=node1
Traefik always routes to Node 1
WebSocket session stable ☐
```

Node 1 crashes:

```
Cookie points to dead node
Traefik detects Node 1 unhealthy
Routes to Node 2
User reconnects (brief disconnect)
Node 2 was already running → fast reconnect
```

What Happens When One Node Goes Down

Scenario: ejabberd-node1 container crashes

Immediately:

- ~50% of users lose WebSocket connection
- Their apps detect disconnect
- Exponential backoff reconnect starts

Within seconds:

- Apps reconnect to chat.nexgate.com
- Traefik detects Node 1 unhealthy
- Routes all new connections to Node 2
- Users reconnect to Node 2

Stream Management (XEP-0198):

- Short disconnects (< 5 min): session resumable
- Users reconnect → Ejabberd resends missed stanzas
- No messages lost

Longer outage:

- RabbitMQ offline queue holds messages
- FCM push notifications already fired
- When user reconnects → queue drains
- Messages delivered

Calls during crash:

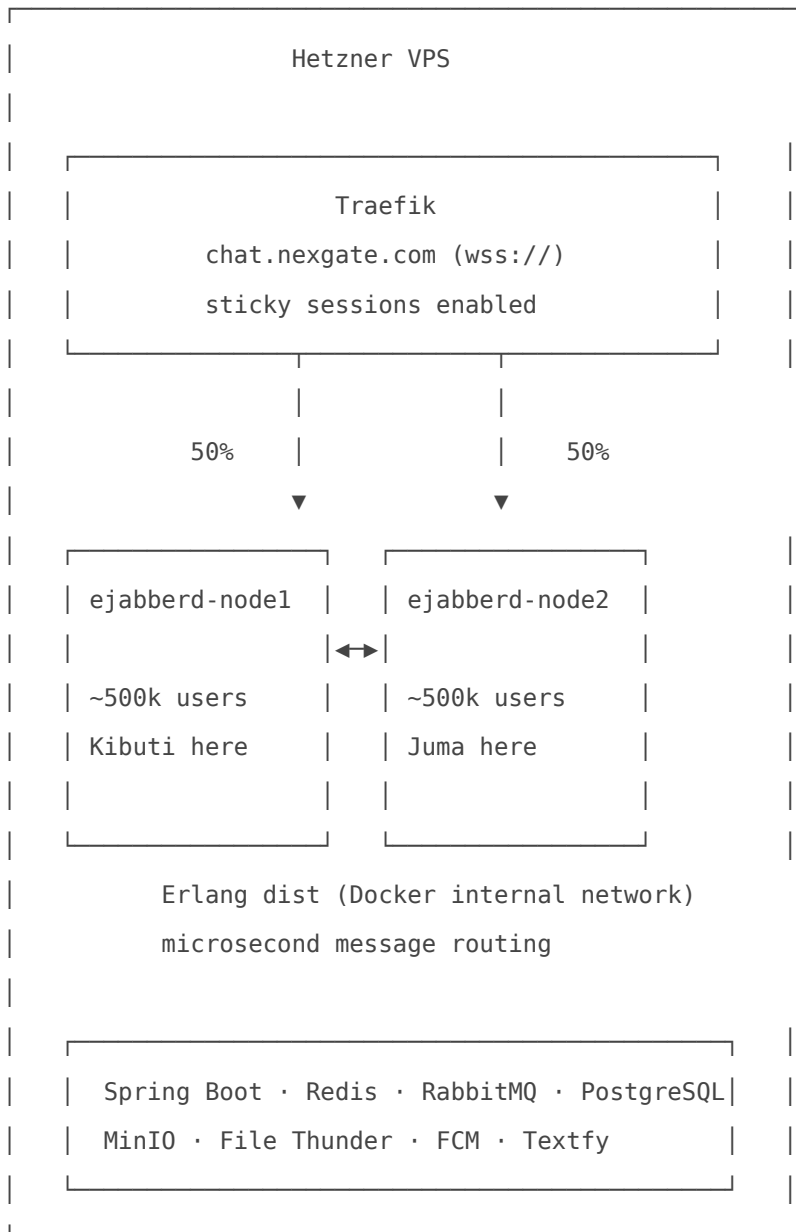
- WebRTC audio/video continues flowing
- (P2P or Coturn – not through Ejabberd)
- Signaling channel dropped
- Active calls: audio continues but
- call management (mute, end) needs reconnect

Node 2 (other 50%):

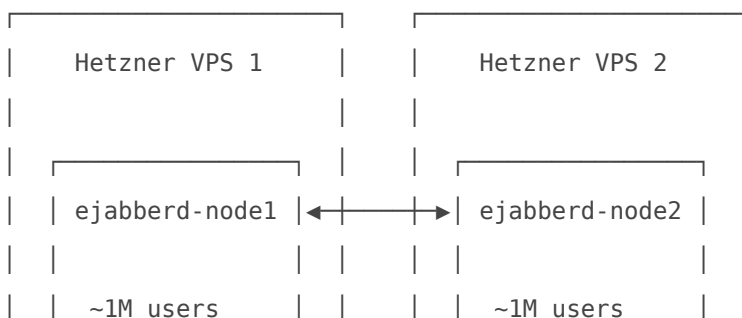
- Completely unaffected
- No interruption for their users
- Their calls continue perfectly

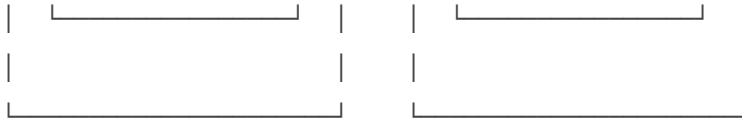
Cluster Architecture — Visual

Same VPS (Launch):



Two VPS (Growth stage):





Erlang dist via Hetzner private network
(free bandwidth between Hetzner VPS)
low latency – same data center region

Mnesia — Ejabberd's Internal Database

Ejabberd uses Mnesia (Erlang's built-in DB)
for internal operational data:

What Mnesia stores:

- Active user sessions (who is connected where)
- MUC room membership + state
- Presence subscriptions (roster)
- Offline message buffer (short-term)

In a cluster:

- Mnesia replicates across nodes automatically
- Node 1 and Node 2 share same Mnesia data
- Node 1 updates session table → Node 2 sees it
- This is how Node 2 knows Juma is on Node 2
- and can route Kibuti's message correctly

Mnesia is NOT:

- A replacement for PostgreSQL
- Where NexGate messages are stored
- Where conversation history lives
- (that is all PostgreSQL via Spring Boot)

Mnesia is purely Ejabberd internal
NexGate Spring Boot never touches Mnesia

Scale Path Summary

Launch:

- 1 VPS
- 2 Docker containers (node1 + node2)

Erlang dist over Docker internal network
Traefik sticky sessions
~1M concurrent capacity
Zero redundancy against VPS hardware failure
☐ Right for launch

Growth:

2 VPS (Hetzner private network)
1 container per VPS
Erlang dist over private network (low latency)
True hardware redundancy
~2M concurrent capacity
VPS failure → other VPS serves all users
☐ Right when uptime = revenue

WeChat EA scale:

3-5 VPS nodes
Each node handles ~500k-1M users
Erlang cluster routes everything
Geographic distribution possible
☐ Right when NexGate is EA infrastructure

5. Connection Lifecycle

Full Connect ? Reconnect ? Disconnect Flow

App launches / user logs in:

|

▼ POST /auth/login (Main Backend)

Receive two tokens:

REST JWT → HTTP API calls (7 days)

XMPP Token → Ejabberd connection (24 hours)

|

▼ Connect WebSocket

wss://chat.nexgate.com/ws

Header: Authorization: Bearer {XMPP_TOKEN}

|

```
▼ XMPP stream opened
<stream:stream to="nexgate.com"
    version="1.0">
    |
    ▼ Ejabberd → Spring Boot (sync HTTP auth)
POST /internal/ejabberd/auth
{ username: "usr-kibuti", token: "XMPP_TOKEN" }
Spring Boot:
    Check Redis cache first (< 5ms if cached)
    Validate JWT signature
    Check user not suspended
    Return 200 or 401
    |
    ▼ Auth success – Ejabberd sends features
<stream:features>
    <sm xmlns="urn:ietf:params:xml:ns:xmpp-session"/>
    <!-- Stream Management XEP-0198 -->
</stream:features>
    |
    ▼ Client enables Stream Management
<enable xmlns="urn:ietf:params:xml:ns:xmpp-sm:3"
    resume="true"/>
Ejabberd: <enabled id="session-abc" resume="true"/>
    |
    ▼ Client sends presence (I am online)
<presence/>
    |
Ejabberd:
    Registers kibuti@nexgate.com/android as ONLINE
    Publishes RabbitMQ: chat.presence.online
    Spring Boot: drain offline queue for kibuti
    |
Connection established ☐
App shows conversations, unread counts
```

Stream Management — Why It Matters for EA

```
Problem without Stream Management:
    Network drops (very common on EA mobile)
```

TCP connection breaks

In-flight messages LOST

User reconnects – no idea what was missed

XEP-0198 Stream Management solution:

Every stanza gets a sequence number

Client ACKs received stanzas:

`<a xmlns="..." h="5"/>` ← "I received up to stanza 5"

Server ACKs received stanzas the same way

On reconnect:

Client sends: `<resume id="session-abc" h="5"/>`

Ejabberd knows: client got up to stanza 5

Ejabberd resends: stanzas 6, 7, 8 (unacknowledged)

Zero message loss

For EA mobile networks:

Connection drops constantly (3G → 2G → WiFi)

Stream Management means users never miss messages

Even on unstable connections

Critical for NexGate commerce DMs

(missing an order negotiation message = lost sale)

Reconnection Strategy

Connection drops detected:

|

App: exponential backoff reconnect

Attempt 1: wait 1 second

Attempt 2: wait 2 seconds

Attempt 3: wait 4 seconds

Attempt 4: wait 8 seconds

Max wait: 30 seconds

|

On reconnect:

If session resumable (< 5 minutes offline):

Resume: `<resume id="session-abc" h="last_ack"/>`

Ejabberd resends missed stanzas

No message loss ☐

If session expired (> 5 minutes offline):

Full re-auth with XMPP token

Fetch conversation list from REST API

Missed messages come from MAM (message archive)

or from RabbitMQ offline queue drain

6. 1:1 Private DMs

Sending a Text Message

[Kibuti types "Habari" – taps Send]

|

▼ App sends XMPP message stanza:

```
<message from="kibuti@nexgate.com/android"
  to="juma@nexgate.com"
  type="chat"
  id="msg-abc-123">
  <body>Habari</body>
  <request xmlns="urn:xmpp:receipts"/>
  <!-- Request delivery receipt XEP-0184 -->
  <nexgate xmlns="urn:nexgate:meta">
    <conv_id>conv-789</conv_id>
    <temp_id>local-xyz</temp_id>
    <level>NORMAL</level>
  </nexgate>
  <!-- NexGate custom namespace for app metadata -->
</message>
```

|

App shows message as: pending ☐

|

▼

[Ejabberd Node 1 – Kibuti's node]

Receives stanza

ACKs via Stream Management:

 ← "I got your stanza"

App: pending → sent ✓

|
Is Juma online?
YES → Juma on Node 2:
 Erlang distributed message → Node 2
 Node 2 delivers to Juma's WS
NO → Store in offline queue
 XEP-0160 offline storage
 OR RabbitMQ (NexGate custom)

|
Fire RabbitMQ: chat.message.inbound
(async, does not block delivery)

|
▼

[Spring Boot Chat Service – async]
Write to PostgreSQL
Write to Redis hot cache
Resolve offline escalation if needed

Receiving a Message

[Juma's app – connected on Node 2]
|
Ejabberd Node 2 pushes stanza:
<message from="kibuti@nexgate.com"
 to="juma@nexgate.com"
 type="chat"
 id="msg-abc-123">
 <body>Habari</body>
 <request xmlns="urn:xmpp:receipts"/>
</message>
|
Juma's app:
 Displays message in conversation
 Automatically sends delivery receipt:

<message from="juma@nexgate.com"
 to="kibuti@nexgate.com">
 <received xmlns="urn:xmpp:receipts"
 id="msg-abc-123"/>

```
</message>
|
Ejabberd routes receipt to Kibuti
Kibuti's app: sent ✓ → delivered ✓✓
|
Juma opens conversation:
<message from="juma@nexgate.com"
  to="kibuti@nexgate.com">
  <displayed xmlns="urn:xmpp:chat-markers:0"
    id="msg-abc-123"/>
</message>
|
Kibuti's app: delivered ✓✓ → read ✓✓ (blue)
```

Rich Content Cards in DMs

NexGate extends XMPP with custom namespaces
for rich content (product cards, events etc)

Product card message:

```
<message from="kibuti@nexgate.com"
  to="juma@nexgate.com"
  type="chat"
  id="msg-card-001">
<body>Angalia bidhaa hii</body>
<nexgate-card xmlns="urn:nexgate:cards">
  <type>PRODUCT</type>
  <ref_id>prod-123</ref_id>
  <snapshot>
    <name>Samsung A15</name>
    <price>450000</price>
    <currency>TZS</currency>
    <image_url>...</image_url>
    <shop_name>TechStore</shop_name>
  </snapshot>
</nexgate-card>
</message>
```

Receiving app:

Detects nexgate-card element
Renders rich card UI instead of plain text
Tappable → deep links to product page

7. Group Chats

MUC — Multi User Chat

Group chats in Ejabberd use XEP-0045 (MUC)
Each group = a MUC room with its own JID:
group-abc@conference.nexgate.com

Two group types in NexGate:

PRIVATE: closed, controlled membership
not discoverable in search
default when creating a group

PUBLIC: open, anyone can join via link
discoverable in NexGate search
explicit choice by creator

Creating a Group — Technical Flow

User creates group in app:

```
|
  ▼ POST /chat/groups/create (Spring Boot)
{
  name: "Business Friends",
  type: "PRIVATE",           ← or PUBLIC
  description: "Dar founders discussion"
}
```

Spring Boot:

Create conversation record (type: GROUP)

Call Ejabberd REST API:

```
POST /api/create_room
```

```
{
  name:      "group-abc",
  service: "conference.nexgate.com",
  options: {
    persistent:      true,
    public:          false,  ← PRIVATE
    members_only:    true,
    allow_private_messages: false
  }
}
```

Creator auto-joined as OWNER

Generate invite link token

Return: { groupId, inviteLink }

Group Join Model — Two Mechanisms

NexGate principle:

Nobody ends up in a group
without actively choosing to join

No forced direct add

Two consent-based mechanisms only

Mechanism 1 — Consent DM Invitation

Admin selects people from contacts/followers/
commerce relationships (NOT strangers)

Each selected person receives a DM:

Spring Boot → Ejabberd REST API:

POST /api/send_message

```
{
  from: "system@nexgate.com",
  to:   "juma@nexgate.com",
  extra: {
    type:      "GROUP_INVITATION",
    group_id:  "group-abc",
    group_name: "Business Friends",
  }
}
```

```

    group_type:    "PRIVATE",
    member_count:  47,
    description:   "Dar founders discussion",
    invited_by:    "Kibuti Mwangi",
    expires_in:    "48h"
  }
}

```

XMPP stanza (NexGate custom namespace):

```

<message from="system@nexgate.com"
  to="juma@nexgate.com"
  type="chat">
  <body>You have been invited to join a group</body>
  <nexgate-group-invite xmlns="urn:nexgate:group:1">
    <group_id>group-abc</group_id>
    <group_name>Business Friends</group_name>
    <group_type>PRIVATE</group_type>
    <member_count>47</member_count>
    <invited_by>Kibuti Mwangi</invited_by>
    <expires_at>2026-07-15T10:00:00Z</expires_at>
  </nexgate-group-invite>
</message>

```

Recipient app renders:

Group Invitation	
Kibuti Mwangi invited you to join:	
Business Friends	
47 members · Private Group	
"Dar founders discussion"	
[Accept & Join]	[Decline]

If Accept:

App sends: POST /chat/groups/group-abc/join

Spring Boot: ejabberdctl add_member group-abc juma

Juma is now a group member ☺

If Decline:

POST /chat/groups/group-abc/decline

Not added ☐

Kibuti NOT notified (privacy)

If Ignored (48h passes):

Invitation auto-expired

Auto-declined silently ☐

Mechanism 2 — Invite Link

Admin generates link:

nexgate.app/join/abc-xyz-def

Person taps link → sees group preview:

PRIVATE GROUP:

App calls: GET /chat/groups/preview/abc-xyz-def

Shows preview + [Request to Join] button

Spring Boot creates join request

Admin sees request in group management

Admin approves → member ☐

Admin declines → person not notified

PUBLIC GROUP:

App calls: GET /chat/groups/preview/abc-xyz-def

Shows preview + [Join Group] button

Tap → instant member ☐

No approval needed

Link settings (stored in DB):

expires_at: nullable (never if null)

max_joins: nullable (unlimited if null)

revoked_at: nullable (active if null)

On revoke:

Old token deleted

New token generated

Old link: "This invite link is no longer valid"

Sending a Group Message

```
<!-- Member sends to group -->
<message to="group-abc@conference.nexgate.com"
        type="groupchat"
        id="gmsg-001">
  <body>Hello everyone!</body>
</message>

<!-- Ejabberd MUC reflects back with room JID -->
<message from="group-abc@conference.nexgate.com/Kibuti"
        type="groupchat"
        id="gmsg-001">
  <body>Hello everyone!</body>
</message>
```

- Ejabberd MUC fan-out:
- Receives from Kibuti
 - Broadcasts to ALL room members simultaneously
 - Each member's WS gets the stanza
 - Erlang handles fan-out natively
 - No Redis pub/sub needed
 - Spring Boot persists via RabbitMQ event

Group Roles — Ejabberd MUC Mapping

Ejabberd MUC role	NexGate role	Permissions
owner	OWNER	everything delete group transfer ownership
admin	ADMIN	manage members delete any message pin messages change group info
moderator	MODERATOR	mute members remove members
participant	MEMBER	send messages

visitor	READ_ONLY	delete own messages view only (announcement mode)
---------	-----------	---

Fan-out Strategy

Groups up to 500 members:

- Ejabberd MUC native fan-out
- All members get stanza in real time
- Erlang handles it – no extra logic

Groups approaching 500:

- Recommend switching to PUBLIC group
- with announcement mode (admins only post)
- Better for large audiences
- No broadcast channels needed
- (VP Feed covers mass content distribution)

Group Invite Link DB Schema

group_invite_links

link_id	UUID	
group_id	UUID	
token	TEXT	unique random token
created_by	UUID	admin userId
expires_at	TIMESTAMPTZ	nullable
max_joins	INT	nullable
join_count	INT	default 0
revoked_at	TIMESTAMPTZ	nullable
created_at	TIMESTAMPTZ	

8. Chat States — Typing & Recording

XEP-0085 — Built Into Ejabberd

No custom backend code needed
Ejabberd routes chat state stanzas automatically
Spring Boot never sees them (not persisted)
Pure real-time ephemeral signals

All States and When App Sends Them

State	App sends when	Recipient sees
composing	user starts typing	"Kibuti is typing..."
paused	user stopped typing (3s no keystroke)	indicator disappears
active	user opened conversation but not typing	no indicator
inactive	user left conversation screen (10s elapsed)	no indicator
gone	user closed conversation	no indicator
recording	user holding mic button	"Kibuti is recording..."

Stanzas

```
<!-- User starts typing -->
<message from="kibuti@nexgate.com/android"
  to="juma@nexgate.com"
  type="chat">
  <composing xmlns="http://jabber.org/protocol/chatstates"/>
</message>

<!-- User paused typing -->
<message from="kibuti@nexgate.com/android"
  to="juma@nexgate.com"
  type="chat">
  <paused xmlns="http://jabber.org/protocol/chatstates"/>
</message>

<!-- User is recording voice note -->
<message from="kibuti@nexgate.com/android"
  to="juma@nexgate.com"
```

```
        type="chat">
    <composing xmlns="http://jabber.org/protocol/chatstates"/>
    <recording xmlns="urn:nexgate:states"/>
    <!-- NexGate custom namespace for recording state -->
</message>

<!-- User stopped recording (sent or cancelled) -->
<message from="kibuti@nexgate.com/android"
        to="juma@nexgate.com"
        type="chat">
    <paused xmlns="http://jabber.org/protocol/chatstates"/>
</message>
```

Throttling — Don't Spam the Network

Wrong (naive) approach:

- Send composing stanza on every single keystroke
- 100 keystrokes = 100 stanzas
- Wastes bandwidth – bad for EA data bundles

Correct approach:

- User starts typing → send composing once
- Keep typing → resend composing every 3 seconds
- User stops → wait 3 seconds → send paused
- Total: ~1 stanza per 3 seconds while typing
- Much more efficient

Mobile dev implements this with a timer:

- startTypingTimer() → fires composing once
- resetTimer() on each keystroke
- onTimerExpire() → send paused

Group Chat States

In group chats:

- Same stanzas – sent to room JID instead of personal JID
- Ejabberd MUC broadcasts to all room members

UI handling when multiple people type:

```
1 person:  "Juma is typing..."
2 people:  "Juma and Amina are typing..."
3+ people: "3 people are typing..."
```

App collects composing events from room

Tracks: Set<userId> currentlyTyping

Renders string based on set size

9. Message Receipts

Three Tick States

✓ Sent	Server received + stored stanza Stream Management ACK received
✓✓ Delivered	Recipient device received stanza XEP-0184 receipt returned
✓✓ Read (blue)	Recipient opened conversation XEP-0333 chat marker returned

XEP-0184 — Delivery Receipt

```
<!-- Kibuti requests receipt in original message -->
<message id="msg-001" ...>
  <body>Habari</body>
  <request xmlns="urn:xmpp:receipts"/>
</message>

<!-- Juma's app automatically responds on delivery -->
<message from="juma@nexgate.com"
  to="kibuti@nexgate.com">
  <received xmlns="urn:xmpp:receipts"
    id="msg-001"/>
<!-- id references the original message -->
```

```
</message>
```

Kibuti's app receives this:

- updates message msg-001 status: DELIVERED
- shows ✓✓

XEP-0333 — Chat Markers (Read Receipt)

```
<!-- Juma opens conversation – app sends displayed marker -->
<message from="juma@nexgate.com"
  to="kibuti@nexgate.com"
  type="chat">
  <displayed xmlns="urn:xmpp:chat-markers:0"
    id="msg-001"/>
  <!-- marks msg-001 and all previous as read -->
</message>
```

Kibuti's app receives this:

- updates msg-001 and all before: READ
- shows ✓✓ blue

Group Message Receipts

In groups: receipts work per member

Message sent to group of 5:

- Each member's delivery → individual receipt
- All 5 delivered → show ✓✓

Read receipts in groups:

- Show count: "Read by 3"
- Tap to see who read it
- (WhatsApp same pattern)

Spring Boot aggregates:

- Stores each receipt in message_receipts table
- Computes: delivered_count, read_count
- Returns to sender on request

11. Presence System

How Presence Works in XMPP

Presence is built into XMPP protocol
No custom implementation needed
Ejabberd handles all presence routing

User connects:

Sends: <presence/>
Ejabberd broadcasts to all contacts
who have presence subscription

User disconnects:

Ejabberd auto-sends: <presence type="unavailable"/>
All subscribed contacts notified

This is fully automatic

Spring Boot only needs to:

Listen to RabbitMQ presence events
Update last_seen_at in PostgreSQL
Cache presence in Redis (for fast lookup)

Presence States

```
<!-- Available (online) -->
<presence from="kibuti@nexgate.com/android"/>

<!-- Away (phone locked / app backgrounded) -->
<presence from="kibuti@nexgate.com/android">
  <show>away</show>
</presence>

<!-- Do Not Disturb -->
<presence from="kibuti@nexgate.com/android">
  <show>dnd</show>
```

```
<status>In a meeting</status>
</presence>

<!-- Offline -->
<presence from="kibuti@nexgate.com/android"
          type="unavailable"/>
```

Last Seen

When user goes offline:

Ejabberd fires: chat.presence.offline (RabbitMQ)

Spring Boot:

Update users.last_seen_at = now

Remove presence:{userId} from Redis

When contact opens chat with offline user:

App requests: GET /chat/users/{userId}/presence

Spring Boot returns:

```
{ status: "offline", lastSeenAt: "2026-07-02T08:30:00Z" }
```

App shows: "Mwisho kuonekana leo saa 2:30"

Privacy settings (Spring Boot enforces):

EVERYONE → anyone can see last seen

CONTACTS → only conversation partners

NOBODY → hide last seen from all

Online Indicator in Conversation

How app shows "online" in DM header:

Option 1 – Subscribe to presence (XMPP native):

App sends presence subscription to contact

Contact auto-notified when they come online

Ejabberd handles real-time push

Option 2 – Poll on conversation open:

GET /chat/users/{userId}/presence

Check Redis: presence:{userId} exists? → online

Simple, no subscription management

NexGate recommendation: Option 2

Simpler to implement

No subscription state to manage

Polling on conversation open is fine

(user only cares when they're IN the conversation)

12. Voice Calls — Deep Dive

Complete Component Map

Voice Call	
Signaling:	Ejabberd Jingle (XEP-0166) "who calls who, exchange network info"
Discovery:	STUN (built into Ejabberd) "find your public IP behind NAT"
Relay:	Coturn TURN server "relay audio when P2P impossible" EA carrier NAT blocks most P2P
Transport:	WebRTC PeerConnection "actual audio stream between devices"
Codec:	Opus "compress audio for EA networks" "adaptive 6kbps (2G) → 64kbps (WiFi)"
Encryption:	SRTP (built into WebRTC) "all audio encrypted end to end"

ICE — How Devices Find Each Other

ICE = Interactive Connectivity Establishment

The algorithm that finds the best path between devices

Step 1 – Gather candidates (both devices do this):

Local candidate:

192.168.1.5:54321 ← local network IP

STUN candidate:

41.188.xxx.xxx:54321 ← public IP (Vodacom/Airtel IP)

Found by asking STUN server: "What is my public IP?"

TURN candidate:

turn.nexgate.com:3478 ← relay fallback

Step 2 – Exchange candidates:

Both share their candidate lists

Via Ejabberd Jingle stanzas

Step 3 – Try connections (priority order):

1. Direct local network (same WiFi) → fastest
2. Direct P2P via public IPs → good
3. TURN relay → always works, higher latency

Step 4 – Use best working path:

Call starts on winning candidate

Can switch mid-call if network changes

EA reality:

Direct local → rarely (different networks)

Direct P2P → sometimes (depends on carrier)

TURN relay → most common on Vodacom/Airtel/Tigo

Full Voice Call Sequence

[Kibuti taps "Call Juma"]

|

▼ App: GET /chat/calls/turn-credentials

Spring Boot generates HMAC TURN credentials:

{

iceServers: [

{ urls: "stun:chat.nexgate.com:3478" },

```

    { urls: "turn:turn.nexgate.com:3478",
      username: "usr-kibuti:1751500000",
      credential: "hmac_sha1_token" }
  ]
}

```

|

▼ App initializes WebRTC PeerConnection

Config: iceServers from above

Add audio track:

```

Opus codec
echoCancellation: true
noiseSuppression: true
autoGainControl: true

```

|

▼ App creates SDP offer

WebRTC generates offer describing:

```

Codecs supported (Opus preferred)
Audio capabilities
Security parameters (DTLS)

```

|

▼ App sends Jingle session-initiate

```

<iq from="kibuti@nexgate.com/android"
  to="juma@nexgate.com"
  type="set"
  id="call-001">
  <jingle xmlns="urn:xmpp:jingle:1"
    action="session-initiate"
    sid="sid-abc-123"
    initiator="kibuti@nexgate.com/android">
    <content name="audio" creator="initiator">
      <description xmlns="urn:xmpp:jingle:apps:rtp:1"
        media="audio">
        <payload-type id="111"
          name="opus"
          clockrate="48000"
          channels="2"/>
      </description>
      <transport xmlns="urn:xmpp:jingle:transports:ice-udp:1"
        ufrag="abc"

```

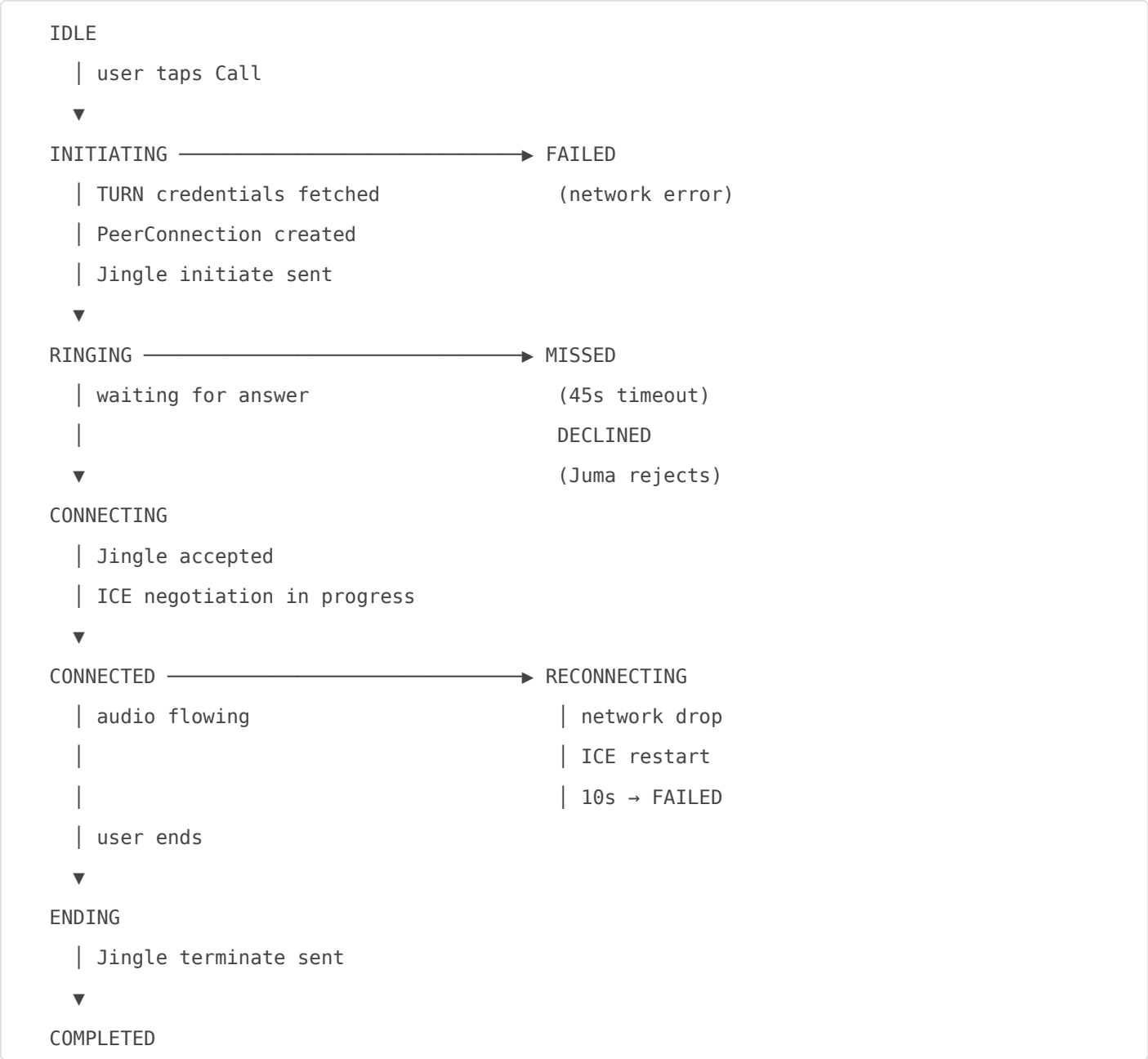
```

        pwd="password123">
    <candidate component="1"
        foundation="1"
        type="host"
        ip="192.168.1.5"
        port="54321"
        priority="2130706431"/>
    <candidate component="1"
        foundation="2"
        type="srflx"
        ip="41.188.xxx.xxx"
        port="54321"
        priority="1694498815"/>
</transport>
</content>
</jingle>
</iq>
|
▼ Ejabberd routes to Juma
Ejabberd fires RabbitMQ: chat.call.initiated
Spring Boot:
    Creates call record (status: RINGING)
    If Juma offline → FCM HIGH priority:
        { type: INCOMING_CALL, callId: "sid-abc-123",
          callerName: "Kibuti", callType: VOICE }
    |
[Juma's phone rings – incoming call screen]
Juma taps Answer
|
▼ Juma: get TURN credentials
Initialize PeerConnection (same config)
Set remote description (Kibuti's SDP)
Create SDP answer
Gather own ICE candidates
|
▼ Juma sends Jingle session-accept
<iq from="juma@nexgate.com/android"
    to="kibuti@nexgate.com/android"
    type="set">

```

```
<jingle action="session-accept"
      sid="sid-abc-123">
  <!-- Juma's SDP answer + ICE candidates -->
</jingle>
</iq>
|
▼ Ejabberd routes to Kibuti
Kibuti's app:
  Sets remote description (Juma's SDP)
  ICE negotiation completes
  Best path selected (likely TURN on EA networks)
  |
CALL LIVE ☐☐
Opus audio flowing between devices
  |
RTCP monitors quality every 200ms:
  Reports: packet loss, jitter, RTT, bandwidth
  Opus adapts bitrate automatically:
    64kbps → 32kbps → 16kbps → 8kbps → 6kbps
  Never drops – always degrades gracefully
  |
Kibuti taps End
  |
  ▼ Jingle session-terminate
<iq type="set">
  <jingle action="session-terminate"
        sid="sid-abc-123">
    <reason><success/></reason>
  </jingle>
</iq>
|
Ejabberd fires RabbitMQ: chat.call.ended
Spring Boot:
  Update call record:
    status: COMPLETED
    ended_at: now
    duration_seconds: 247
    relay_used: true
    end_reason: NORMAL
```

Call State Machine



Opus Codec Ladder

Network	Bitrate	What it sounds like
WiFi / 4G strong	64 kbps	HD voice, crystal clear
4G normal	32 kbps	Clear, natural voice
3G	16 kbps	Good, slight compression
2G / Edge	8 kbps	Robotic but intelligible
Barely alive	6 kbps	Minimum – still connected

Opus switches between these automatically
based on RTCP feedback every 200ms
Mobile dev configures nothing – it just works

Key Opus features for EA:

inbandfec: true	→ Forward Error Correction recovers from packet loss without retransmit
usedtx: true	→ Discontinuous Transmission silence = no packets sent saves bandwidth during pauses
stereo: false	→ Mono only for calls half the bitrate vs stereo

13. Video Calls — Deep Dive

Additional Components vs Voice

Voice call +

- Video codec (H.264 primary)
- Camera capture (front/rear switchable)
- Video rendering (remote + local preview)
- Higher bandwidth requirement
- Higher CPU on device
- More Coturn relay bandwidth if P2P fails

H.264 — Why for EA

H.264 (AVC) chosen because:

Hardware acceleration:

- Every phone since 2013 has H.264 hardware encoder
- Including Tecno Spark, Infinix Hot (dominant in EA)
- Hardware encoder = GPU does the work
- Battery impact: LOW

CPU: barely used

Software encoding (VP8, VP9, AV1):

CPU does all encoding work

On low-end EA phones: hot, slow, battery drain

10 minutes of video = significant battery cost

Users notice and complain

H.264 at low bitrates:

360p @ 400kbps → works on 3G

240p @ 150kbps → works on 2G

Quality acceptable for face-to-face conversation

Video Resolution Ladder

Device tier + network → resolution selected:

Device	Network	Resolution	FPS	Bitrate
Any	WiFi	720p	30	1.5 Mbps
Any	4G strong	480p	24	800 kbps
Any	3G	360p	15	400 kbps
Any	2G	240p	10	150 kbps
Any	Very poor	AUDIO ONLY	–	Opus only

Degradation order (call never drops):

1. Reduce color depth
2. Reduce resolution (720→480→360→240)
3. Reduce frame rate (30→24→15→10)
4. Reduce audio bitrate
5. Disable video completely → audio only
6. Audio minimum (6kbps Opus)

Upgrade is conservative:

Wait 5 seconds of stable improved bandwidth

Then upgrade one step (e.g. 360p → 480p)

Prevents quality flapping on unstable networks

Jingle for Video — Two Content Blocks

```
<jingle action="session-initiate" sid="vid-001">

  <!-- Audio block – always present -->
  <content name="audio" creator="initiator">
    <description media="audio">
      <payload-type id="111" name="opus"
        clockrate="48000"/>
    </description>
    <transport ...>
      <candidate .../> <!-- ICE candidates -->
    </transport>
  </content>

  <!-- Video block – added for video calls -->
  <content name="video" creator="initiator">
    <description media="video">
      <!-- H.264 preferred -->
      <payload-type id="96" name="H264"
        clockrate="90000"/>
      <!-- VP8 as fallback -->
      <payload-type id="97" name="VP8"
        clockrate="90000"/>
    </description>
    <transport ...>
      <candidate .../>
    </transport>
  </content>

</jingle>
```

Camera UI Features

Mobile dev implements:

Local preview (Picture-in-Picture):

Small corner window showing your own camera

Standard in all video call UIs

Switch camera:

Front → Rear → Front toggle

WebRTC: `videoCapturer.switchCamera()`

Camera off (privacy):

`videoTrack.setEnabled(false)`

Remote sees: black screen or avatar

Audio continues

Auto-disable video on battery:

Monitor: battery < 20% AND on Coturn relay

Show warning: "Battery low – switching to audio only"

Disable video track

Continue audio call

10. Message Interactions

All message interactions use standard XMPP XEPs. Ejabberd routes stanzas automatically. Spring Boot validates rules and persists via RabbitMQ.

XEP Overview

Feature	XEP	Status	Ejabberd
Edit message	XEP-0308	Stable 	auto routed
Delete message	XEP-0424	Stable 	auto routed
Reactions	XEP-0444	Stable 	auto routed
Forwarding	XEP-0297	Stable 	auto routed
Reply/Quote	XEP-0461	Exp 	auto routed
Stable IDs	XEP-0359	Stable 	auto assigned

Edit — XEP-0308

```
<!-- Kibuti edits his sent message -->
```

```
<message from="kibuti@nexgate.com"
  to="juma@nexgate.com"
```

```
        type="chat"
        id="edit-002">
<body>Hello Juma, how is business today?</body>
<replace xmlns="urn:xmpp:message-correct:0"
        id="server-stable-id-abc"/>
<!-- references original by stanza-id (XEP-0359) -->
</message>
```

Rules:

- Only original sender can edit ☐
- Text messages only ☐
- Within 15 minutes of sending ☐
- Shows "Edited" label after ☐
- Commerce cards: NOT editable ☐
- System messages: NOT editable ☐

Spring Boot on receiving edit event (RabbitMQ):

- Validate author + time window
- Update messages.body
- Update messages.edited_at
- Increment messages.edit_count

Delete — XEP-0424

```
<!-- Delete for everyone -->
<message from="kibuti@nexgate.com"
        to="juma@nexgate.com"
        type="chat"
        id="retract-003">
  <apply-to xmlns="urn:xmpp:fasten:0"
        id="server-stable-id-abc">
    <retract xmlns="urn:xmpp:message-retract:1"/>
  </apply-to>
</message>
```

Delete for me:

- No stanza needed
- Local REST call only
- POST /chat/messages/{id}/delete { scope: SELF }

Recipient unaffected

Delete for everyone:

XEP-0424 retraction stanza

Within 15 minutes only

Commerce cards: NOT deletable ☐

System messages: NOT deletable ☐

Recipient sees: "This message was deleted"

Nothing hard-deleted from PostgreSQL (audit trail)

Reactions — XEP-0444

```
<!-- Add reaction -->
<message from="kibuti@nexgate.com"
  to="juma@nexgate.com"
  type="chat">

  <reactions xmlns="urn:xmpp:reactions:0"
    id="server-stable-id-abc">

    <reaction>👍/reaction>

  </reactions>
</message>

<!-- Remove reaction (empty = removed) -->
<message from="kibuti@nexgate.com"
  to="juma@nexgate.com"
  type="chat">

  <reactions xmlns="urn:xmpp:reactions:0"
    id="server-stable-id-abc">

  </reactions>
</message>
```

Rules:

One reaction per user per message ☐

Change: send new emoji (replaces) ☐

Remove: send empty reactions element ☐

Commerce cards: reactions ALLOWED ☐

System messages: reactions NOT allowed ☐

Launch emoji set: ♥️ 🍌 🍌 🍌 🍌 🍌 🍌 🍌

Forwarding — XEP-0297

```
<!-- Kibuti forwards Juma's message to Alice -->
<message from="kibuti@nexgate.com"
  to="alice@nexgate.com"
  type="chat"
  id="fwd-001">
  <body>Check this out</body>
  <forwarded xmlns="urn:xmpp:forward:0">
    <delay xmlns="urn:xmpp:delay"
      stamp="2026-07-13T10:32:00Z"/>
    <message from="juma@nexgate.com"
      to="kibuti@nexgate.com"
      type="chat">
      <body>Hello everyone!</body>
    </message>
  </forwarded>
  <nexgate-forward xmlns="urn:nexgate:forward">
    <original_sender_name>Juma Mwangi</original_sender_name>
    <forward_chain>1</forward_chain>
  </nexgate-forward>
</message>
```

Rules:

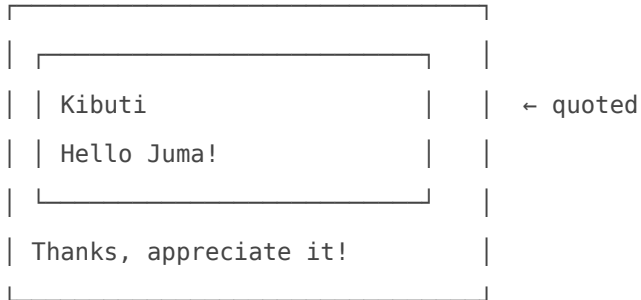
- Max 5 conversations per forward action ☐
- Chain 1: "Forwarded from Juma Mwangi"
- Chain 2-4: "Forwarded"
- Chain 5+: "Forwarded many times" (warning)
- Media: references original fileId – no re-upload ☐
- Custom price offers: NOT forwardable ☐
- Order/payment records: NOT forwardable ☐

Reply — XEP-0461

```
<!-- Reply to a specific message -->
<message from="juma@nexgate.com"
  to="kibuti@nexgate.com"
  type="chat"
  id="reply-001">
```

```
<body>Thanks, appreciate it!</body>
<reply xmlns="urn:xmpp:reply:0"
      to="kibuti@nexgate.com"
      id="server-stable-id-abc"/>
</message>
```

Renders as:



Tap quote → scrolls to original message

14. Audio ? Video Switching & Screen Share

Switch Audio ? Video During Call

Call starts as voice only
User taps camera button during call
No hang up needed – same WebRTC session

Kibuti enables camera:

- Creates video track (H.264)
- Adds to existing PeerConnection
- Sends Jingle content-add stanza:

```
<iq from="kibuti@nexgate.com/android"
  to="juma@nexgate.com"
  type="set">
  <jingle xmlns="urn:xmpp:jingle:1"
    action="content-add"
```

```
        sid="sid-abc-123">
<!-- sid = SAME session as voice call -->
<content name="video" creator="initiator">
    <description media="video">
        <payload-type id="96" name="H264"
            clockrate="90000"/>
    </description>
    <transport .../>
</content>
</jingle>
</iq>
```

Juma accepts:

Jingle action="content-accept"

Video starts flowing – same TURN relay ☐

Audio uninterrupted during upgrade ☐

Switch back (video → audio):

Jingle action="content-remove"

Removes video content block

Audio continues

Auto-downgrade (network-triggered):

RTCP detects bandwidth too low

App sends content-remove automatically

Banner: "Video disabled – poor network"

Resumes when network improves

Screen Sharing

Screen share = special video track

Instead of camera → captures device screen

Same H.264 encoding

Lower frame rate (5-15fps – screen changes slowly)

Android: MediaProjection API

iOS: ReplayKit broadcast extension

Start screen share:

User taps screen share icon during call
System permission dialog appears:
"Allow NexGate to capture your screen?"
User accepts
Screen capture starts

Jingle stanza (adds screen content block):

```
<jingle action="content-add"
  sid="sid-abc-123">
  <content name="screen" creator="initiator">
    <description media="video">
      <payload-type id="96" name="H264"
        clockrate="90000"/>
    </description>
    <transport .../>
  </content>
</jingle>
```

During screen share:

Remote side sees: screen (large) + face (PiP)
Local side sees: "Sharing screen" banner
Camera optional: can keep or disable

Stop screen share:

Jingle content-remove (screen)
Returns to normal video/audio call

EA network consideration:

Screen content is mostly static
H.264 compresses static content very well
720p screen at ~300kbps (vs 720p camera at 1.5Mbps)
Works on 3G for text/document sharing ☐

15. Group Calls

Why LiveKit for Group Calls

1:1 call:

- P2P or Coturn relay

- Two devices, one path

- No server media processing

Group call (3+ people):

- Cannot P2P to everyone simultaneously

- Kibuti uploads 1 stream to LiveKit

- LiveKit forwards to all other participants

- Each participant uploads once → downloads N-1

- SFU = Selective Forwarding Unit

LiveKit already deployed for Audio Spaces

Same Docker container

Same Coturn relay reused

Zero new infrastructure ☐

Group Call Flow

Kibuti starts group call from group chat:

|

▼ POST /chat/calls/group/start

Spring Boot:

- Create LiveKit room: group-call-{callId}

- Generate token per participant:

 - canPublish: true

 - canSubscribe: true

- Return tokens + LiveKit WS URL

|

Jingle session-initiate sent to all group members:

```
<message from="system@nexgate.com"
```

```
  to="juma@nexgate.com"
```

```
  type="chat">
```

```
<nexgate-call xmlns="urn:nexgate:call:1">
```

```
  <type>GROUP_CALL_JOIN_INFO</type>
```

```
  <call_id>call-xyz</call_id>
```

```
  <call_type>VIDEO</call_type>
```

```
  <livekit_url>wss://livekit.nexgate.com</livekit_url>
```

```
<livekit_token>eyJ...</livekit_token>
<room_id>group-call-xyz</room_id>
<expires_in>300</expires_in>
</nexgate-call>
</message>
```

Each member receives → phone rings

Members who join → connect WebRTC to LiveKit

LiveKit SFU forwards all streams ☐

[LiveKit SFU]

Kibuti stream → forwarded to Juma + Alice

Juma stream → forwarded to Kibuti + Alice

Alice stream → forwarded to Kibuti + Juma

EA Network Limits for Group Calls

Group voice (audio only, Opus):

3 people: each downloads 64kbps → works on 3G ☐

5 people: each downloads 128kbps → works on 3G ☐

8 people: each downloads 224kbps → needs 4G ▲☐

Group video (H.264 + Opus):

3 people: each downloads 800kbps → needs 4G ▲☐

4 people: each downloads 1.2Mbps → needs strong 4G ▲☐

5+ people: reduce to active speaker only ☐

Max participants shown:

Voice: up to 8 (3G compatible)

Video: up to 4 feeds simultaneously

5th+ person: audio tile only (no video feed)

Active speaker highlighted (larger tile)

Simulcast — EA Network Diversity

Each participant uploads 3 quality versions:

Low: 180p + Opus 16kbps

Medium: 360p + Opus 32kbps

High: 720p + Opus 64kbps

LiveKit delivers appropriate quality per receiver:

Receiver on 2G → low quality streams

Receiver on WiFi → high quality streams

Each receiver gets quality their network allows

Independently per stream

Result:

Good network user sees HD video

Poor network user sees low quality

Everyone stays in the call ☐

No one's bad network drops everyone else

16. Offline Handling

Three Layers of Offline Delivery

Layer 1 – Ejabberd XEP-0160 (offline storage):

User disconnects mid-session

Ejabberd stores pending stanzas

On reconnect: delivers immediately

Covers: short disconnections (seconds to minutes)

Layer 2 – RabbitMQ queue:

User has been offline longer

Spring Boot queues messages

On reconnect: Chat Service drains queue

Priority order: CRITICAL → IMPORTANT → NORMAL

Covers: hours to days offline

Layer 3 – FCM + Textfy (notifications):

Wakes device even when completely offline

User sees notification → opens app

Triggers Layer 1 + 2 delivery

Covers: device asleep, app killed

FCM for Calls (Special Case)

If Juma is offline when Kibuti calls:

Spring Boot sends FCM HIGH priority:

```
{
  type: "INCOMING_CALL",
  callId: "sid-abc-123",
  callerName: "Kibuti Mwangi",
  callerAvatar: "https://...",
  callType: "VOICE",
  turnCredentials: { ... } ← included for fast answer
}
```

|

FCM wakes Juma's phone

App shows full-screen incoming call UI
(even if app was completely killed)

|

Juma taps Answer:

App already has TURN credentials

Immediately creates PeerConnection

Sends Jingle session-accept

No extra round trip to get credentials

Faster answer time ☑

Call ringing timeout: 45 seconds

After 45s → Spring Boot marks: MISSED

→ Juma sees missed call notification

Catch-Up on Reconnect

User was offline – comes back online:

|

▼ WS connects → Ejabberd → auth success

Spring Boot receives: chat.presence.online (RabbitMQ)

|

Spring Boot:

Drain RabbitMQ offline queue for user

Check MAM (Message Archive) for any gaps

Build catch-up summary

|

App receives catch-up payload:

Missed messages pushed via WS

App shows banner:

"Umekosa ujumbe 12, maagizo 2"

[Angalia] button

Message Archive (MAM – XEP-0313):

Ejabberd stores last N days of messages

Client can query: "give me messages since X"

Covers edge cases where queue was lost

17. Multi Device

How Multiple Devices Work

Kibuti logged into:

kibuti@nexgate.com/android ← phone

kibuti@nexgate.com/tablet ← tablet

Message arrives:

Ejabberd delivers to BOTH devices

Both show the message

Both show notification

Kibuti reads on phone:

Phone sends: <displayed id="msg-001"/>

Ejabberd: sees kibuti read the message

XEP-0280 Message Carbons:

Tablet automatically receives the read marker

Tablet clears notification and marks read

Without user doing anything on tablet

This is how WhatsApp multi-device works

Ejabberd handles it natively via XEP-0280

Device Priority

If Kibuti active on phone + tablet:

Both receive messages (carbons)

If only one device active:

That device receives normally

Presence priority:

Each resource has a priority number

Higher priority = preferred delivery target

Phone: priority 10 (main device)

Tablet: priority 5 (secondary)

When both online: phone gets delivery first

Tablet gets carbon copy

Set in presence stanza:

```
<presence>
```

```
  <priority>10</priority>
```

```
</presence>
```

18. Shop Inbox in Phase 2

Shop JID — The Shop as XMPP Entity

Each NexGate shop has its own JID:

techstore@shops.nexgate.com

This is NOT Kibuti's personal JID

This is the SHOP's identity

When customer messages TechStore:

Customer sends to: techstore@shops.nexgate.com

Any authorized staff member sees it

All staff respond AS techstore@shops.nexgate.com

Customer sees "TechStore" – not individual names

Staff authentication to shop JID:

Staff logs in with own account

Switches to shop context in app

Spring Boot issues shop XMPP sub-token:

```
{ jid: "techstore@shops.nexgate.com",  
  staffId: "usr-amina",  
  role: "SUPPORT_AGENT" }
```

Ejabberd allows staff to auth as shop JID

All messages from staff appear as TechStore

Multiple Staff — Shared Inbox

TechStore has 3 staff:

Kibuti (owner – Manager role)

Amina (Support Agent)

John (Support Agent)

Customer sends message to TechStore:

Message arrives at techstore@shops.nexgate.com

Ejabberd delivers to ALL connected TechStore staff

(All three see the incoming message simultaneously)

Amina responds:

Response appears as "TechStore" to customer

Spring Boot audit log:

```
{ messageId, respondedBy: "usr-amina",  
  shopId: "shop-techstore", timestamp }
```

Kibuti and John see Amina's response in their inbox too

(full shared inbox – everyone sees everything)

Benefits:

No missed customer messages

Any staff can pick up any conversation

Owner can monitor all conversations

Customer always talks to "TechStore"

19. Security

Transport Security

WebSocket:

- wss:// (WebSocket Secure)
- TLS 1.3 termination at Traefik
- All chat traffic encrypted in transit

TURN relay:

- SRTP (Secure Real-time Transport Protocol)
- Voice/video encrypted even through Coturn
- Coturn relays encrypted packets
- Coturn cannot decrypt audio/video

XMPP tokens:

- Short-lived (24 hours)
- Signed with RS256 (asymmetric)
- Separate from REST JWT
- Stored in Vault

Internal Service Security

Ejabberd → Spring Boot:

- X-Internal-Secret header
- Secret stored in Vault
- Only Ejabberd knows this secret
- Spring Boot rejects any request without it

Spring Boot → Ejabberd:

- Admin token (Ejabberd API key)
- Stored in Vault
- Port 5285 bound to 127.0.0.1 only
- Not exposed to public internet

All inter-service secrets:

- Stored in HashiCorp Vault ☐

Rotatable without restart
Never in environment files
Never in Docker Compose plain text

Message Privacy

Server-side:

Messages stored in PostgreSQL (encrypted at rest)
Media stored in MinIO (server-side encryption)
Shop conversations isolated from personal inbox
Staff cannot access personal DMs of owner

In transit:

WSS for all WebSocket traffic
SRTP for all call media

Future (E2E encryption):

Signal Protocol integration possible
Would use OMEMO (XEP-0384) on top of XMPP
Ejabberd supports OMEMO natively
Messages encrypted on device
Server stores ciphertext only
Not in Phase 2 scope – plan for Phase 3

20. Database Schema

conversations

conversations

id	UUID	
type	ENUM	DM / GROUP / COMMERCE
owner_type	ENUM	USER / SHOP
owner_id	UUID	userId or shopId
title	TEXT	groups only
avatar_file_id	UUID	

status	ENUM	ACTIVE / ARCHIVED / BLOCKED
created_by	UUID	
created_at	TIMESTAMPTZ	
last_message_at	TIMESTAMPTZ	
last_message_preview	TEXT	

conversation_members

conversation_members

conversation_id	UUID	
user_id	UUID	
role	ENUM	OWNER / ADMIN / MODERATOR / MEMBER
joined_at	TIMESTAMPTZ	
last_read_at	TIMESTAMPTZ	
last_read_seq	BIGINT	
is_muted	BOOLEAN	
muted_until	TIMESTAMPTZ	
notifications	ENUM	ALL / MENTIONS / NONE

messages

messages

id	UUID	
conversation_id	UUID	
sender_id	UUID	
seq	BIGINT	monotonic per conversation
type	ENUM	TEXT / IMAGE / VIDEO / VOICE_NOTE / FILE / PRODUCT_CARD / CUSTOM_PRICE_OFFER / EVENT_CARD / GROUP_PURCHASE_CARD / POST_CARD / ORDER_CONFIRMATION / ORDER_STATUS_UPDATE / PAYMENT_CONFIRMATION / SYSTEM
body	TEXT	
media_ref	UUID	File Thunder fileId
context_type	ENUM	PRODUCT / ORDER / PAYMENT /

EVENT / GROUP_PURCHASE

context_ref_id	UUID	
snapshot_json	JSONB	frozen context at send time
reply_to_id	UUID	
status	ENUM	SENT / DELIVERED / READ / FAILED
level	ENUM	NORMAL / IMPORTANT / CRITICAL
edited_at	TIMESTAMPTZ	
deleted_at	TIMESTAMPTZ	
created_at	TIMESTAMPTZ	

message_receipts

message_receipts

message_id	UUID	
user_id	UUID	
status	ENUM	DELIVERED / READ
device_id	TEXT	
timestamp	TIMESTAMPTZ	

calls

calls

call_id	UUID	
caller_id	UUID	
receiver_id	UUID	
conversation_id	UUID	
type	ENUM	VOICE / VIDEO
status	ENUM	RINGING / CONNECTED / COMPLETED / MISSED / DECLINED / FAILED
started_at	TIMESTAMPTZ	
answered_at	TIMESTAMPTZ	
ended_at	TIMESTAMPTZ	
duration_seconds	INT	
relay_used	BOOLEAN	
end_reason	ENUM	NORMAL / NETWORK /

call_quality_logs

call_quality_logs

log_id	UUID	
call_id	UUID	
timestamp	TIMESTAMPTZ	
bitrate_kbps	INT	
packet_loss_pct	DECIMAL	
jitter_ms	INT	
rtt_ms	INT	
resolution	TEXT	null for voice calls
codec_audio	TEXT	"opus"
codec_video	TEXT	"h264" "vp8" null

shop_conversation_access

shop_conversation_access

shop_id	UUID	
user_id	UUID	
role	ENUM	MANAGER / SUPPORT_AGENT / READ_ONLY
granted_by	UUID	
granted_at	TIMESTAMPTZ	
revoked_at	TIMESTAMPTZ	

notification_log

notification_log

id	UUID	
user_id	UUID	
message_id	UUID	
level	ENUM	NORMAL / IMPORTANT / CRITICAL
fcm_status	ENUM	SENT / DELIVERED / FAILED
sms_status	ENUM	SENT / DELIVERED / FAILED / SKIPPED

sms_provider	TEXT
sent_at	TIMESTAMPZ
delivered_at	TIMESTAMPZ
opened_at	TIMESTAMPZ

Summary

Private chat and calls in NexGate Phase 2 are built on four pillars:

Ejabberd Cluster runs as two Docker containers on the same Hetzner VPS at launch. Erlang Distribution connects them directly — messages between nodes route in microseconds without Redis pub/sub. Traefik sticky sessions keep each user's WebSocket on one node. If one node crashes the other keeps serving. At growth stage two separate Hetzner VPS give true hardware redundancy.

Ejabberd handles everything real-time — WebSocket connections, XMPP stanza routing, presence, chat states, message receipts, MUC group chats, and Jingle call signaling. All message interactions (edit XEP-0308, delete XEP-0424, reactions XEP-0444, forwarding XEP-0297, replies XEP-0461) are routed automatically — Spring Boot only handles persistence and rule validation.

Group chats use a consent-based join model. Nobody enters a group without actively choosing. Two mechanisms: consent DM invitation (admin handpicks from their network, each person accepts or declines) and invite link (private groups require admin approval, public groups allow instant join). Both private and public group types supported. No forced adding — better than WhatsApp.

WebRTC handles all calls. 1:1 calls use P2P or Coturn relay via Jingle signaling. Group calls use LiveKit SFU (already deployed for Audio Spaces) — zero new infrastructure. Audio↔video switching uses Jingle content-add/remove without ending the session. Screen sharing uses MediaProjection (Android) and ReplayKit (iOS) as a special video track. Opus adapts from 64kbps to 6kbps. H.264 hardware acceleration keeps battery impact low on EA phones.

Spring Boot Chat Service handles all business logic — message persistence, commerce context, offer sessions, shop inbox access control, notification routing, and call records. Auth with Ejabberd is synchronous HTTP (needs immediate allow/deny). Everything else is async via RabbitMQ.

The shop inbox is isolated from personal DMs at the JID level — the shop has its own Ejabberd identity, multiple staff share it, and customers always see the shop brand, never individual staff names.

NexGate Private Chat & Calls — Phase 2 Deep Dive v1.0 QBIT SPARK | XMPP · Ejabberd · WebRTC · Jingle · Coturn · Opus · H.264 · Group Calls · Screen Share
