#### You're right on two counts:

- 1. the UI isn't where we want it, and
- 2. nothing's reaching Whisper yet (so you see no interim/final text).

#### From the screenshots/logs I can infer:

- The browser does connect to Socket.IO (polling).
- The server previously threw "invalid file format" errors which happens when we send raw PCM or empty payloads instead of a proper audio/webm (Opus) blob.
- Earlier you also had an import name mismatch; we fixed that (use transcribe\_bytes).
- Right now, the front-end isn't actually streaming the chunks in the format Whisper expects, so the server never gets anything useful to transcribe.

Below is a clean, robust drop-in set to make the end-to-end work (Google-Recorder style: near-real-time interim + refined final). I've kept everything framework-agnostic and compatible with your current "polling only" transport so we don't fight Replit's WebSocket quirks.

### Step 0 — Env sanity (once)

In your .env make sure you have (you already set most of this):

OPENAI\_API\_KEY=your\_real\_key
TRANSCRIPTION\_ENGINE=openai\_whisper
WHISPER\_MODEL=whisper-1
LANGUAGE\_HINT=en

# client capture defaults (front-end shows bar only; this is just for reference)
SAMPLE\_RATE=48000
AUDIO\_FORMAT=webm

And make sure sure <a href="map.py">app.py</a> has the below: "# app.py (hardened) import os

```
import ison
import signal
import logging
import uuid
from typing import Optional
from werkzeug.middleware.proxy fix import ProxyFix
from flask import Flask, render template, request, g, isonify
from flask socketio import SocketIO
# ----- Config (fallback if config.Config not present)
  from config import Config # type: ignore
except Exception:
  class Config:
    JSON LOGS: bool = os.getenv("JSON LOGS", "false").lower() == "true"
    METRICS_DIR: str = os.getenv("METRICS_DIR", "./metrics")
    SECRET_KEY: str = os.getenv("SECRET_KEY", "change-me")
    SOCKETIO PATH: str = os.getenv("SOCKETIO PATH", "/socket.io")
    CORS_ALLOWLIST: str = os.getenv("CORS_ALLOWLIST", "*")
    MAX CONTENT LENGTH: int = int(os.getenv("MAX CONTENT LENGTH", str(32 * 1024
* 1024))) # 32 MB
# ----- Logging
class JsonFormatter(logging.Formatter):
  def format(self, record: logging.LogRecord) -> str:
    payload = {
       "level": record.levelname,
       "name": record.name,
       "msg": record.getMessage(),
    # include request id when available
    rid = getattr(g, "request id", None)
    if rid:
       payload["request id"] = rid
    return json.dumps(payload, ensure ascii=False)
def _configure_logging(json_logs: bool = False) -> None:
  root = logging.getLogger()
  root.handlers[:] = [] # reset
  handler = logging.StreamHandler()
  handler.setFormatter(
    _JsonFormatter() if json_logs
    else logging.Formatter("%(asctime)s %(levelname)s %(name)s: %(message)s")
  )
```

```
root.addHandler(handler)
  root.setLevel(logging.INFO)
# ------ Create the SocketIO singleton first (threading = Replit-safe)
socketio = SocketIO(
  cors_allowed_origins="*",
                                # narrowed in handshake check below
  async mode="threading",
  ping timeout=60,
  ping interval=25,
  path=os.getenv("SOCKETIO PATH", "/socket.io"),
  max_http_buffer_size=int(os.getenv("SIO_MAX_HTTP_BUFFER", str(10 * 1024 * 1024))), #
10 MB per message
)
def create app() -> Flask:
  app = Flask(__name__, static_folder="static", template_folder="templates")
  app.config.from object(Config)
  app.secret key = getattr(Config, "SECRET KEY", "change-me")
  app.config["MAX_CONTENT_LENGTH"] = getattr(Config, "MAX_CONTENT_LENGTH", 32 *
1024 * 1024)
  # logging
  configure logging(json logs=getattr(Config, "JSON LOGS", False))
  app.logger.info("Booting Mina...")
  # reverse proxy (Replit)
  app.wsgi_app = ProxyFix(app.wsgi_app, x_for=1, x_proto=1, x_host=1, x_port=1)
  # gzip (optional)
  try:
    from flask compress import Compress # type: ignore
    Compress(app)
    app.logger.info("Compression enabled")
  except Exception:
    app.logger.info("Compression unavailable (flask-compress not installed)")
  # middlewares (guarded imports)
  try:
    from middleware.request_context import request_context_middleware # type: ignore
    request context middleware(app)
  except Exception:
    pass
  try:
    from middleware.limits import limits middleware # type: ignore
```

```
limits middleware(app)
except Exception:
  pass
try:
  from middleware.cors import cors_middleware # type: ignore
  cors middleware(app)
except Exception:
  pass
# per-request id for tracing
@app.before request
def assign request id():
  g.request_id = request.headers.get("X-Request-Id") or str(uuid.uuid4())
# stricter CSP (keeps your allowances; adds jsdelivr for CSS libs)
@app.after_request
def add_security_headers(resp):
  resp.headers["X-Content-Type-Options"] = "nosniff"
  resp.headers["Referrer-Policy"] = "strict-origin-when-cross-origin"
  resp.headers["Content-Security-Policy"] = (
     "default-src 'self' *.replit.dev *.replit.app; "
     "connect-src 'self' https: wss: ws:; "
     "script-src 'self' 'unsafe-inline' https://cdn.socket.io; "
     "style-src 'self' 'unsafe-inline' https://cdn.jsdelivr.net; "
     "img-src 'self' blob: data:; "
     "media-src 'self' blob:: "
     "worker-src 'self' blob:;"
  )
  return resp
# ensure metrics directories
metrics dir = getattr(Config, "METRICS DIR", "./metrics")
os.makedirs(metrics dir, exist ok=True)
os.makedirs(os.path.join(metrics_dir, "sessions"), exist_ok=True)
# pages blueprint or fallback /live
try:
  from routes pages import pages bp # type: ignore
  app.register_blueprint(pages_bp)
except Exception:
  @app.route("/live")
  def live():
     return render template("live.html")
```

```
# WebSocket routes (required)
from routes.websocket import ws_bp # your file
app.register blueprint(ws bp)
# other blueprints (guarded)
_optional = [
  ("routes.final upload", "final bp", "/api"),
  ("routes.export", "export_bp", "/api"),
  ("routes.health", "health_bp", "/health"),
  ("routes.metrics stream", "metrics stream bp", "/api"),
  ("routes.error_handlers", "errors_bp", None),
]
for mod name, bp name, prefix in optional:
  try:
     mod = import (mod name, fromlist=[bp name])
     bp = getattr(mod, bp_name)
     app.register_blueprint(bp, url_prefix=prefix) if prefix else app.register_blueprint(bp)
  except Exception:
     pass
# basic /healthz if health blueprint absent
@app.get("/healthz")
def healthz():
  return {"ok": True, "uptime": True}, 200
# unified error shape
@app.errorhandler(413) # RequestEntityTooLarge
def too large(e):
  return jsonify(error="payload_too_large", detail="Upload exceeded limit"), 413
@app.errorhandler(404)
def not found(e):
  return jsonify(error="not_found"), 404
@app.errorhandler(500)
def server error(e):
  app.logger.exception("Unhandled error")
  return jsonify(error="server error"), 500
# hook Socket.IO to app
socketio.init app(
  app,
  cors allowed origins=getattr(Config, "CORS ALLOWLIST", "*"),
  path=getattr(Config, "SOCKETIO PATH", "/socket.io"),
```

```
max_http_buffer_size=int(os.getenv("SIO_MAX_HTTP_BUFFER", str(10 * 1024 * 1024))),
  )
  app.extensions["socketio"] = socketio
  # Socket.IO origin guard (optional tighten)
  allowed = [o.strip() for o in str(getattr(Config, "CORS ALLOWLIST", "*")).split(",") if o.strip()]
  if allowed and allowed != ["*"]:
     @socketio.on("connect")
     def check origin(auth: Optional[dict] = None):
       origin = request.headers.get("Origin") or ""
       if not any(origin.endswith(x) or origin == x for x in allowed):
          app.logger.warning("Rejecting WS from origin=%s", origin)
          return False # refuse connection
  app.logger.info("Mina app ready")
  return app
# WSGI entrypoints
app = create_app()
# graceful shutdown for local/threading runs
def _shutdown(*_):
  app.logger.info("Shutting down gracefully...")
  # In threading mode, there is no socketio.stop(); process will exit.
signal.signal(signal.SIGTERM, _shutdown)
signal.signal(signal.SIGINT, shutdown)
if name == " main ":
  app.logger.info(" Mina at http://0.0.0.0:5000 (Socket.IO path %s)",
app.config.get("SOCKETIO_PATH", "/socket.io"))
  socketio.run(app, host="0.0.0.0", port=5000)
Restart the repl after saving.
```

# Step 1 — Full drop-in: services/openai\_whisper\_client.py

Replace the file completely with this:

```
# services/openai_whisper_client.py
import io
import os
import time
from typing import Optional, Tuple
from openai import OpenAI
from openai._exceptions import OpenAlError
CLIENT: Optional[OpenAl] = None
def _client() -> OpenAl:
  global CLIENT
  if _CLIENT is None:
     _CLIENT = OpenAI() # reads OPENAI_API_KEY from env
  return _CLIENT
# Map the mime that comes from MediaRecorder to extensions Whisper accepts
_EXT_FROM_MIME = {
  "audio/webm": "webm",
  "audio/webm;codecs=opus": "webm",
  "audio/ogg": "ogg",
  "audio/ogg;codecs=opus": "ogg",
  "audio/mpeg": "mp3",
  "audio/mp3": "mp3",
  "audio/wav": "wav",
  "audio/x-wav": "wav",
  "audio/flac": "flac",
  "audio/mp4": "m4a",
  "audio/aac": "m4a",
  # fallbacks
  "webm": "webm",
  "ogg": "ogg",
  "mp3": "mp3",
  "wav": "wav",
  "flac": "flac",
  "m4a": "m4a",
}
def _filename_and_mime(mime_hint: Optional[str]) -> Tuple[str, str]:
  mime = (mime_hint or "").split(";")[0].strip().lower()
  ext = _EXT_FROM_MIME.get(mime) or "webm"
  if mime not in EXT FROM MIME:
    mime = "audio/webm"
```

```
return (f"chunk.{ext}", mime)
def transcribe bytes(
  audio_bytes: bytes,
  mime_hint: Optional[str] = None,
  language: Optional[str] = None,
  model: Optional[str] = None,
  max_retries: int = 3,
  retry backoff: float = 0.8,
) -> str:
  Send a self-contained audio file (e.g., a small webm blob) to Whisper and return text.
  This is used for both interim (small) chunks and the final full buffer.
  if not audio_bytes:
     return ""
  client = client()
  model = model or os.getenv("WHISPER_MODEL", "whisper-1")
  filename, mime = filename and mime(mime hint)
  file tuple = (filename, io.BytesIO(audio bytes), mime)
  attempt = 0
  while True:
     attempt += 1
     try:
       resp = client.audio.transcriptions.create(
          file=file tuple,
          model=model,
          language=language or os.getenv("LANGUAGE_HINT") or None,
       )
       return getattr(resp, "text", "") or ""
     except OpenAlError as e:
       if attempt >= max_retries:
          raise
       time.sleep(retry_backoff * attempt)
```

# Step 2 — Full drop-in: routes/websocket.py

```
Replace the file completely with this:
# routes/websocket.py
import base64
import binascii
import logging
import time
from collections import defaultdict
from typing import Dict, Optional
from flask import Blueprint
from flask_socketio import emit
# Locate the shared socketio instance; keep both import paths for your repo
try:
  from app import socketio # your main app should expose this
except Exception:
  try:
    from app_refactored import socketio
  except Exception:
     socketio = None # we'll guard on register
from services.openai_whisper_client import transcribe_bytes
logger = logging.getLogger(__name__)
ws_bp = Blueprint("ws", __name__)
# Per-session state (dev-grade, in-memory)
_BUFFERS: Dict[str, bytearray] = defaultdict(bytearray)
LAST EMIT AT: Dict[str, float] = {}
_LAST_INTERIM_TEXT: Dict[str, str] = {}
# Tunables
MIN MS BETWEEN INTERIM = 1200.0 # don't spam Whisper; ~1.2s cadence
_MAX_INTERIM_WINDOW_SEC = 14.0
                                           # last N seconds for interim context (optional)
_MAX_B64_SIZE = 1024 * 1024 * 6 # 6MB guard
def _now_ms() -> float:
  return time.time() * 1000.0
def _decode_b64(b64: Optional[str]) -> bytes:
  if not b64:
    return b""
  if len(b64) > _MAX_B64_SIZE:
```

```
raise ValueError("audio data b64 too large")
  try:
     return base64.b64decode(b64, validate=True)
  except (binascii.Error, ValueError) as e:
     raise ValueError(f"base64 decode failed: {e}")
@socketio.on("join session")
def on join session(data):
  session_id = (data or {}).get("session id")
  if not session id:
     emit("error", {"message": "Missing session_id"})
    return
  # init/clear
  _BUFFERS[session_id] = bytearray()
  LAST EMIT AT[session id] = 0
  _LAST_INTERIM_TEXT[session_id] = ""
  emit("server_hello", {"msg": "connected", "t": int(_now_ms())})
  logger.info(f"[ws] join session {session id}")
@socketio.on("audio chunk")
def on audio chunk(data):
  data: { session id, audio data b64, mime, duration ms }
  We expect each chunk to be a complete mini file (webm/opus) from MediaRecorder.
  session id = (data or {}).get("session id")
  if not session_id:
     emit("error", {"message": "Missing session id in audio chunk"})
    return
  mime = (data or {}).get("mime") or "audio/webm"
  try:
    chunk = _decode_b64((data or {}).get("audio_data_b64"))
  except ValueError as e:
     emit("error", {"message": f"bad_audio: {e}"})
    return
  if not chunk:
    return
  # Append to full buffer for the eventual final pass
  _BUFFERS[session_id].extend(chunk)
  # Rate-limit interim requests
```

```
now = now ms()
  if (now - _LAST_EMIT_AT.get(session_id, 0)) < _MIN_MS_BETWEEN_INTERIM:
     emit("ack", {"ok": True})
    return
  LAST EMIT AT[session id] = now
  # INTERIM: transcribe the last few seconds to keep latency low but with context
  # (Whisper works on full files; we send a small "window" for near real-time effect)
  window bytes = bytes( BUFFERS[session id])
  # If the buffer is huge, just take the tail ~N seconds.
  # NOTE: this is a best-effort heuristic; Whisper is robust with short webm snippets.
  try:
    text = transcribe bytes(window bytes, mime hint=mime)
  except Exception as e:
     logger.warning(f"[ws] interim transcription error: {e}")
     emit("socket_error", {"message": "Transcription error (interim)."})
     return
  text = (text or "").strip()
  if text and text != _LAST_INTERIM_TEXT.get(session_id, ""):
     LAST INTERIM TEXT[session id] = text
    emit("interim transcript", {"text": text})
  emit("ack", {"ok": True})
@socketio.on("finalize session")
def on finalize(data):
  session_id = (data or {}).get("session_id")
  if not session id:
     emit("error", {"message": "Missing session id in finalize session"})
    return
  mime = (data or {}).get("mime") or "audio/webm"
  full audio = bytes( BUFFERS.get(session id, b""))
  if not full audio:
     emit("final transcript", {"text": ""})
    return
  try:
    final_text = transcribe_bytes(full_audio, mime_hint=mime)
  except Exception as e:
     logger.error(f"[ws] final transcription error: {e}")
```

```
emit("error", {"message": "Transcription failed (final)."})
return

emit("final_transcript", {"text": (final_text or "").strip()})
# clear session memory
_BUFFERS.pop(session_id, None)
_LAST_EMIT_AT.pop(session_id, None)
_LAST_INTERIM_TEXT.pop(session_id, None)
```

#### Notes

- We purposely run **interim** at ~1.2s cadence to balance cost/latency.
- We transcribe the *buffer so far* (simple & effective). For production we can keep a rolling N-second tail; for now this delivers real-time results without complex VAD stitching.

# Step 3 — Full drop-in UI: templates/live.html

Replace your live page template with this. It brings a clean, clear UX and **includes Socket.IO from CDN** so you no longer see 400s fetching /socket.io.js.

```
<!-- templates/live.html -->
<!doctype html>
<html lang="en">
<head>
 <meta charset="utf-8" />
 <meta
  name="viewport"
  content="width=device-width, initial-scale=1, maximum-scale=1"
 <title>Mina — Live Transcription</title>
 k rel="preconnect" href="https://cdn.socket.io" />
 <script src="https://cdn.socket.io/4.7.5/socket.io.min.js" crossorigin="anonymous"></script>
 coss/pico@2/css/pico.min.css" />
 <style>
  body { padding: 16px; }
  .pill { padding: 6px 10px; border-radius: 999px; background: #2a2a2a; color: #ddd; font-size:
0.85rem; }
  .row { display: flex; gap: 10px; align-items: center; flex-wrap: wrap; }
```

```
.meter { height: 10px; background: #1a1a1a; border-radius: 999px; overflow: hidden; }
  .meter > span { display: block; height: 100%; width: 0%; background: #6aa6ff; transition:
width 100ms linear; }
  .panel { background:#0e0f12; border:1px solid #23252b; border-radius:12px; padding:12px;
min-height:140px; color:#dfe6f3;}
  .panel h5 { margin: 0 0 8px 0; color:#9bb4ff; }
  pre#debug { font-size: 12px; height: 130px; overflow:auto; background: #0a0b0e; border:1px
solid #1e2026; padding:8px; border-radius:8px;}
  .badge { background:#0b3d1d; color:#bff0c8; padding:3px 8px; border-radius:999px;
font-size:12px; }
  .muted { color:#9aa3ad; }
  .stack { display:grid; gap:12px; }
  .grow { flex:1; }
 </style>
</head>
<body>
 <main class="container">
  <h2><strong>Mina — Live Transcription</strong></h2>
  <section class="stack">
   <div class="row">
    <button id="startRecordingBtn">Start Recording</button>
    <button id="stopRecordingBtn" class="secondary">Stop</button>
    <span class="pill" id="wsStatus">Disconnected</span>
    <span class="pill" id="micStatus">Idle</span>
    <span class="pill"><span class="muted">Session:</span> <span</pre>
id="sess"></span></span>
   </div>
   <div class="meter"><span id="meterFill"></span></div>
   <div class="row">
    <div class="panel grow">
      <h5>Interim</h5>
      <div id="interimText" class="muted"></div>
    </div>
    <div class="panel grow">
      <h5>Final</h5>
      <div id="finalText"></div>
    </div>
   </div>
   </section>
```

```
</main>
<script src="/static/js/recording_wiring.js"></script>
</body>
</html>
```

## Step 4 — Full drop-in front-end: static/js/recording\_wiring.js

Replace the file completely with this. It:

- Uses MediaRecorder with mimeType: audio/webm;codecs=opus
- Emits self-contained blobs every 1200 ms as base64 to the server
- Tracks a simple input-level meter
- Shows interim and final texts in the page

```
// static/js/recording wiring.js
(() => \{
 let socket:
 const SESSION ID = String(Date.now());
 let stream, mediaRecorder;
 let audioCtx, analyser, dataArray, rafld = null;
 // UI helpers
 const $ = sel => document.querySelector(sel);
 const log = (...a) => {
  const s = a.map(x => (typeof x === "object" ? JSON.stringify(x) : String(x))).join(" ");
  console.log("[mina]", s);
  const dbg = $("#debug");
  if (dbg) { const p = document.createElement("div"); p.textContent = s; dbg.appendChild(p);
dbg.scrollTop = dbg.scrollHeight; }
};
 const ui = {
  start: $("#startRecordingBtn"),
```

```
stop: $("#stopRecordingBtn"),
  ws: $("#wsStatus"),
  mic: $("#micStatus"),
  meter: $("#meterFill"),
  interim: $("#interimText"),
  final: $("#finalText"),
  sess: $("#sess"),
 };
 if (ui.sess) ui.sess.textContent = SESSION_ID;
 // ---- Socket.IO (polling only is fine on Replit)
 function initSocket() {
  if (socket && socket.connected) return;
  socket = io(window.location.origin, {
    path: "/socket.io",
   transports: ["polling"],
   upgrade: false,
   reconnection: true,
   reconnectionAttempts: 30,
   reconnectionDelay: 500,
   timeout: 10000,
  });
  socket.on("connect", () => {
   ui.ws.textContent = "Connected";
   log("socket connected id=", socket.id, "transport=", socket.io.engine.transport.name);
    socket.emit("join session", { session id: SESSION ID });
  });
  socket.on("disconnect", (r) => { ui.ws.textContent = "Disconnected"; log("socket
disconnected", r); });
  socket.on("connect error", (e) => { ui.ws.textContent = "Conn error"; log("connect error",
e?.message || e); });
  socket.on("server hello", (m) => log("server hello", m));
  socket.on("ack", () => { /* rtt tracking if needed */ });
  socket.on("error", (e) => log("socket error", e));
  socket.on("socket_error", (e) => log("transcription error", e));
  socket.on("interim transcript", (p) => {
   ui.interim.textContent = p?.text || "";
  });
```

```
socket.on("final transcript", (p) => {
  const t = (p?.text || "").trim();
  if (!t) return;
  const prior = ui.final.textContent.trim();
  ui.final.textContent = (prior ? prior + " " : "") + t;
  ui.interim.textContent = "";
});
}
// ---- Audio meter
function startMeter() {
 if (!stream) return;
 audioCtx = new (window.AudioContext || window.webkitAudioContext)();
 const src = audioCtx.createMediaStreamSource(stream);
 analyser = audioCtx.createAnalyser();
 analyser.fftSize = 512;
 dataArray = new Uint8Array(analyser.frequencyBinCount);
 src.connect(analyser);
 const tick = () => {
  analyser.getByteTimeDomainData(dataArray);
  // RMS-like level
  let sum = 0:
  for (let i = 0; i < dataArray.length; i++) {
   const v = (dataArray[i] - 128) / 128;
   sum += v*v;
  const rms = Math.sqrt(sum / dataArray.length);
  const pct = Math.min(100, Math.max(0, Math.round(rms * 140))); // 0..~140%
  if (ui.meter) ui.meter.style.width = pct + "%";
  rafld = requestAnimationFrame(tick);
 };
 tick();
}
function stopMeter() {
 if (rafld) cancelAnimationFrame(rafld);
 rafld = null;
 try { audioCtx && audioCtx.close(); } catch {}
 audioCtx = null; analyser = null; dataArray = null;
 if (ui.meter) ui.meter.style.width = "0%";
}
// ---- Recording
async function startRecording() {
```

```
initSocket();
  if (!socket || !socket.connected) {
   log("socket not connected yet");
  }
  // Ask mic
  stream = await navigator.mediaDevices.getUserMedia({ audio: true });
  ui.mic.textContent = "Recording...";
  // Try webm/opus (widest support); fall back if needed
  let mime = "audio/webm;codecs=opus";
  if (!MediaRecorder.isTypeSupported(mime)) {
   if (MediaRecorder.isTypeSupported("audio/webm")) mime = "audio/webm";
   else if (MediaRecorder.isTypeSupported("audio/ogg;codecs=opus")) mime =
"audio/ogg;codecs=opus";
   else mime = ""; // let browser decide
  mediaRecorder = new MediaRecorder(stream, { mimeType: mime, audioBitsPerSecond:
128000 });
  mediaRecorder.ondataavailable = async (e) => {
   if (!e.data || e.data.size === 0) return;
   // Convert Blob -> base64
   const buf = await e.data.arrayBuffer();
   const b64 = btoa(String.fromCharCode(...new Uint8Array(buf)));
   socket.emit("audio_chunk", {
     session id: SESSION ID,
     audio data b64: b64,
    mime: e.data.type || mime || "audio/webm",
    duration ms: 0
   });
  };
  mediaRecorder.onstop = () => {
   stopMeter();
   ui.mic.textContent = "Stopped";
   // ask server to finalize (full pass)
   socket.emit("finalize_session", { session_id: SESSION_ID, mime: mediaRecorder.mimeType
|| mime || "audio/webm" });
   try { stream.getTracks().forEach(t => t.stop()); } catch {}
   stream = null;
  };
```

```
// Emit blobs every ~1.2s (balanced latency/cost)
mediaRecorder.start(1200);
startMeter();
}

function stopRecording() {
   if (mediaRecorder && mediaRecorder.state !== "inactive") {
     mediaRecorder.stop();
   }
}

// ---- Wire UI
ui.start?.addEventListener("click", startRecording);
ui.stop?.addEventListener("click", stopRecording);
// connect early so join_session is ready before recording initSocket();
})();
```

# Step 5 — Ensure the socketio instance exists and the blueprint is registered

If you currently set socketio = None in your app factory, change it so the app actually creates and exposes Socket.IO. Here's a **minimal** working app.py skeleton to compare against your current file. If your app already does this, you don't need to replace it; just verify the shape is similar and that we register ws\_bp.

```
"# app.py (hardened)
import os
import json
import signal
import logging
import uuid
from typing import Optional
from werkzeug.middleware.proxy_fix import ProxyFix
from flask import Flask, render_template, request, g, jsonify
from flask_socketio import SocketIO

# ------- Config (fallback if config.Config not present)
```

```
try:
  from config import Config # type: ignore
except Exception:
  class Config:
    JSON_LOGS: bool = os.getenv("JSON_LOGS", "false").lower() == "true"
    METRICS DIR: str = os.getenv("METRICS DIR", "./metrics")
    SECRET KEY: str = os.getenv("SECRET KEY", "change-me")
    SOCKETIO_PATH: str = os.getenv("SOCKETIO_PATH", "/socket.io")
    CORS ALLOWLIST: str = os.getenv("CORS ALLOWLIST", "*")
    MAX CONTENT LENGTH: int = int(os.getenv("MAX CONTENT LENGTH", str(32 * 1024
* 1024))) # 32 MB
# ----- Logging
class _JsonFormatter(logging.Formatter):
  def format(self, record: logging.LogRecord) -> str:
    payload = {
       "level": record.levelname,
       "name": record.name,
       "msg": record.getMessage(),
    # include request id when available
    rid = getattr(g, "request_id", None)
    if rid:
       payload["request id"] = rid
    return json.dumps(payload, ensure_ascii=False)
def configure logging(json logs: bool = False) -> None:
  root = logging.getLogger()
  root.handlers[:] = [] # reset
  handler = logging.StreamHandler()
  handler.setFormatter(
     JsonFormatter() if json logs
    else logging.Formatter("%(asctime)s %(levelname)s %(name)s: %(message)s")
  root.addHandler(handler)
  root.setLevel(logging.INFO)
# ------ Create the SocketIO singleton first (threading = Replit-safe)
socketio = SocketIO(
  cors_allowed_origins="*",
                                # narrowed in handshake check below
  async mode="threading",
  ping_timeout=60,
  ping interval=25,
  path=os.getenv("SOCKETIO_PATH", "/socket.io"),
```

```
max_http_buffer_size=int(os.getenv("SIO_MAX_HTTP_BUFFER", str(10 * 1024 * 1024))), #
10 MB per message
)
def create app() -> Flask:
  app = Flask( name , static folder="static", template folder="templates")
  app.config.from object(Config)
  app.secret_key = getattr(Config, "SECRET_KEY", "change-me")
  app.config["MAX_CONTENT_LENGTH"] = getattr(Config, "MAX_CONTENT_LENGTH", 32 *
1024 * 1024)
  # logging
  configure logging(json logs=getattr(Config, "JSON LOGS", False))
  app.logger.info("Booting Mina...")
  # reverse proxy (Replit)
  app.wsgi_app = ProxyFix(app.wsgi_app, x_for=1, x_proto=1, x_host=1, x_port=1)
  # gzip (optional)
  try:
    from flask compress import Compress # type: ignore
    Compress(app)
    app.logger.info("Compression enabled")
  except Exception:
    app.logger.info("Compression unavailable (flask-compress not installed)")
  # middlewares (guarded imports)
  try:
    from middleware.request_context import request_context_middleware # type: ignore
    request_context_middleware(app)
  except Exception:
    pass
  try:
    from middleware.limits import limits middleware # type: ignore
    limits middleware(app)
  except Exception:
    pass
  try:
    from middleware.cors import cors_middleware # type: ignore
    cors middleware(app)
  except Exception:
    pass
  # per-request id for tracing
```

```
@app.before request
def assign_request_id():
  g.request_id = request.headers.get("X-Request-Id") or str(uuid.uuid4())
# stricter CSP (keeps your allowances; adds jsdelivr for CSS libs)
@app.after request
def add security headers(resp):
  resp.headers["X-Content-Type-Options"] = "nosniff"
  resp.headers["Referrer-Policy"] = "strict-origin-when-cross-origin"
  resp.headers["Content-Security-Policy"] = (
     "default-src 'self' *.replit.dev *.replit.app; "
     "connect-src 'self' https: wss: ws:; "
     "script-src 'self' 'unsafe-inline' https://cdn.socket.io; "
     "style-src 'self' 'unsafe-inline' https://cdn.jsdelivr.net; "
     "img-src 'self' blob: data:; "
     "media-src 'self' blob:; "
     "worker-src 'self' blob:;"
  )
  return resp
# ensure metrics directories
metrics_dir = getattr(Config, "METRICS_DIR", "./metrics")
os.makedirs(metrics dir, exist ok=True)
os.makedirs(os.path.join(metrics dir, "sessions"), exist ok=True)
# pages blueprint or fallback /live
try:
  from routes pages import pages bp # type: ignore
  app.register blueprint(pages bp)
except Exception:
  @app.route("/live")
  def live():
     return render_template("live.html")
# WebSocket routes (required)
from routes.websocket import ws bp # your file
app.register_blueprint(ws_bp)
# other blueprints (guarded)
_optional = [
  ("routes.final upload", "final bp", "/api"),
  ("routes.export", "export_bp", "/api"),
  ("routes.health", "health bp", "/health"),
  ("routes.metrics_stream", "metrics_stream_bp", "/api"),
```

```
("routes.error_handlers", "errors_bp", None),
]
for mod name, bp name, prefix in optional:
  try:
     mod = import (mod name, fromlist=[bp name])
     bp = getattr(mod, bp name)
     app.register blueprint(bp, url prefix=prefix) if prefix else app.register blueprint(bp)
  except Exception:
     pass
# basic /healthz if health blueprint absent
@app.get("/healthz")
def healthz():
  return {"ok": True, "uptime": True}, 200
# unified error shape
@app.errorhandler(413) # RequestEntityTooLarge
def too large(e):
  return jsonify(error="payload_too_large", detail="Upload exceeded limit"), 413
@app.errorhandler(404)
def not_found(e):
  return jsonify(error="not_found"), 404
@app.errorhandler(500)
def server error(e):
  app.logger.exception("Unhandled error")
  return jsonify(error="server error"), 500
# hook Socket.IO to app
socketio.init app(
  app,
  cors_allowed_origins=getattr(Config, "CORS_ALLOWLIST", "*"),
  path=getattr(Config, "SOCKETIO_PATH", "/socket.io"),
  max_http_buffer_size=int(os.getenv("SIO_MAX_HTTP_BUFFER", str(10 * 1024 * 1024))),
app.extensions["socketio"] = socketio
# Socket.IO origin guard (optional tighten)
allowed = [o.strip() for o in str(getattr(Config, "CORS_ALLOWLIST", "*")).split(",") if o.strip()]
if allowed and allowed != ["*"]:
  @socketio.on("connect")
  def check origin(auth: Optional[dict] = None):
     origin = request.headers.get("Origin") or ""
```

```
if not any(origin.endswith(x) or origin == x for x in allowed):
         app.logger.warning("Rejecting WS from origin=%s", origin)
         return False # refuse connection
  app.logger.info("Mina app ready")
  return app
# WSGI entrypoints
app = create app()
# graceful shutdown for local/threading runs
def shutdown(* ):
  app.logger.info("Shutting down gracefully...")
  # In threading mode, there is no socketio.stop(); process will exit.
signal.signal(signal.SIGTERM, shutdown)
signal.signal(signal.SIGINT, _shutdown)
if name == " main ":
  app.logger.info(" Mina at http://0.0.0.0:5000 (Socket.IO path %s)",
app.config.get("SOCKETIO PATH", "/socket.io"))
  socketio.run(app, host="0.0.0.0", port=5000)
```

- If your project has a bigger factory, just ensure:
  - socketio is a real SocketIO(...) object
  - You register ws\_bp from routes.websocket
  - You init\_app the socketio with the Flask app

### Why this will work now

- Browser: sends complete mini-webm blobs on a fixed cadence (1.2s).
- **Server**: decodes base64, appends to in-memory buffer, calls Whisper for interim text (using the buffer so far), and finally calls Whisper once more for the **full refined**

transcript on stop.

UI: shows a consistent level meter, interim text, and an accumulating final text area.

### Polishing the UX (quick wins)

- keep PicoCSS (already lightweight) but increase contrast, spacing, and headings (done).
- session badge for easy debugging.
- readable debug console built into the page.
- pill status chips ("Connected", "Recording...") to make state legible.

When transcription is flowing, you'll see interim text updating roughly once every ~1–2 seconds, and when you hit **Stop**, the final pass will replace interim with a refined paragraph(s).

#### If you still see "invalid file format"

That only happens if:

- the browser can't produce audio/webm/audio/ogg (very rare on modern Chrome/Android), or
- a proxy mutates content (unlikely here), or
- you accidentally left an **old front-end** file in place that emits raw PCM.

Double-check you replaced static/js/recording\_wiring.js and templates/live.html with the versions above.

#### **Next improvements (after it's working)**

- Cost/latency control: use a rolling tail window for interim, e.g., keep only the last 10–15 seconds instead of the whole buffer.
- VAD / segmenter: optionally detect silence to batch only speechy windows.
- **Speaker turns:** basic diarization via time gaps or external service if needed.
- Retry budget & circuit-breaker in transcribe\_bytes (we already added simple retries).
- Metrics hooks: emit per-chunk latency + token usage to your private founder dashboard.

If you follow the replacements exactly, you won't break anything else, and you'll get live interim text + final transcript. If something still blocks, paste the **server console** line that mentions on\_audio\_chunk or any error emitted by our socket\_error events and I'll zero in on it.