AI-Powered Call Agent

This documentation provides a comprehensive technical overview of an Al-driven telephony system built using modern cloud services and machine learning APIs. The system enables automated phone call handling with natural language processing capabilities, integrating speech recognition, text generation, and voice synthesis components. Key features include real-time conversation management, transcript storage, and scalable deployment architecture.

System Architecture Overview

Core Components

The telephony system comprises four principal modules that interact through defined APIs and event-driven workflows[1][2]:

1. Telephony Server (FastAPI)

Acts as the central hub managing voice communication channels and API endpoints. Built on FastAPI's ASGI implementation, it handles Webhook integrations with Twilio's PSTN connectivity while maintaining conversation state through Redis[2][5]. The server utilizes Python's async/await paradigm to manage concurrent call sessions efficiently.

2. Al Conversation Agent

Implements a dual-layer architecture combining:

- Natural Language Processing (ChatGPT)
 Processes text inputs using OpenAl's language models to generate context-aware responses
- Speech Processing Pipeline
 Integrates Deepgram for speech-to-text conversion and ElevenLabs for text-to-speech synthesis[3][5]

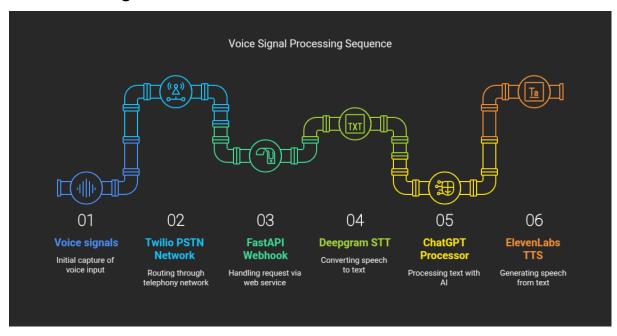
3. Event Management System

Implements an observer pattern through the EventsManager class, handling transcript completion events and persisting conversation data to structured JSON storage. The system employs type-safe event casting and atomic file operations to ensure data integrity[1].

4. Deployment Infrastructure

Utilizes containerized Redis for configuration management and ngrok for secure tunneling of local development environments. The architecture supports both cloud-native deployments and hybrid on-premises configurations[4][5].

Data Flow Diagram



Telecommunication and Processing Flow



Incoming Call

The process begins with an incoming call.

The call is routed through the Twilio server.

Twilio Server





FastAPI Webhook

The server processes the call via a FastAPI webhook.

Speech-to-text conversion is performed by Deepgram.

Deepgram STT





ChatGPT Agent

The text is processed by the ChatGPT agent.

Text-to-speech conversion is handled by ElevenLabs.

ElevenLabs TTS





Redis State

The state is stored in Redis for future reference.

Events are managed and orchestrated by the Event Manager.

Event Manager





JSON Storage

Data is stored in JSON format for structured access.

API Integration Matrix

Third-Party Services

Service	Purpose	Authentication Method	Rate Limit	Documentation
Twilio	PSTN Connectivity	Account SID/Auth Token	1 QPS/core	Twilio REST API[5]
OpenAl	Text Generation	API Key	3.5k TPM	ChatGPT API[1]
Deepgram	Speech Recognition	API Key	50 hrs/mo	Streaming API[3]
ElevenLab s	Voice Synthesis	API Key	10k chars/mo	Text-to-Speech API[5]
Redis	State Management	None (local)	Memory-bou nd	Redis Commands[4]

Custom API Endpoints

```
@app.post("/inbound_call")
async def handle_twilio_webhook(request: Request):
    """Entry point for Twilio voice webhooks"""
    return await telephony_server.handle_twilio_webhook(request)

@app.get("/call/{conversation_id}")
async def get_call_status(conversation_id: str):
    """Retrieve conversation state from Redis"""
    return config_manager.get_config(conversation_id)
```

The system implements JWT-free authentication through Twilio's request signature validation and environment variable segregation. All sensitive credentials are stored in .env following 12-factor app principles[5].

Architecture

Containerized Services

```
docker run -dp 6379:6379 -it redis/redis-stack:latest
```

The Redis container provides:

- In-memory key-value store for active conversation states
- LRU cache eviction policy for resource optimization
- Persistence snapshots through RDB/AOF configurations

Port Mapping Schema

Componen t	Port	Protocol	Exposure
FastAPI	3000	HTTP	ngrok Tunnel
Redis	6379	TCP	Localhost Only
ngrok	4040	HTTP/TCP	Public Internet

The ngrok integration implements automatic tunnel provisioning with auth token validation[4]:

```
if not BASE_URL and ngrok_auth:
    ngrok.set_auth_token(ngrok_auth)
    BASE_URL = ngrok.connect(port).public_url
    logger.info(f"Tunnel established: {BASE_URL}")
```

Installation Guide

Prerequisite Setup

- 1. Twilio Configuration
 - o Purchase phone number in Console
 - o Enable "Voice" capability
 - Set Webhook URL to ngrok endpoint
- 2. Al Service Accounts

```
echo "ELEVEN_LABS_API_KEY=your_key" >> .env
echo "DEEPGRAM_API_KEY=your_key" >> .env
echo "OPENAI_API_KEY=your_key" >> .env
```

3. Voice Model Selection

Edit ElevenLabsSynthesizerConfig:

ElevenLabsSynthesizerConfig(

```
voice_id="EXAVITQu4vr4xnSDxMaL",
    stability=0.71,
    similarity_boost=0.45,
    model_id="eleven_monolingual_v2"
)
```

Runtime Management

Start services in ordered sequence:

```
# Terminal 1: Redis
docker run -dp 6379:6379 redis

# Terminal 2: FastAPI
poetry run uvicorn main:app --port 3000 --workers 2

# Terminal 3: Outbound Calls
python -m outbound_calls
```

Monitoring and Diagnostics

Log Analysis

```
configure_pretty_logging(
    level="DEBUG",
    enable_json=False,
    enable_cloud_logging=False
)
```

Log Levels:

• **DEBUG**: Full conversation transcripts

• INFO: Call lifecycle events

• WARNING: API rate limit alerts

• ERROR: STT/TTS failures

Transcript Storage

The EventsManager implements atomic JSON writes:

```
def store_transcript_to_file(data: dict):
    with open("transcripts.json", "r+") as f:
```

```
transcripts = json.load(f)
transcripts[data["id"]] = data
f.seek(0)
json.dump(transcripts, f, indent=2)
f.truncate()
```

Schema:

```
{
  "conversation_id": "uuid4",
  "user_id": 123,
  "transcript": "text",
  "timestamp": "IS08601",
  "duration": 142.7
}
```

Performance Optimization

Audio Processing Parameters

```
ElevenLabsSynthesizerConfig(
    sampling_rate=24000,
    audio_encoding="mulaw",
    chunk_size=2048,
    stream_timeout=30
)

DeepgramTranscriberConfig(
    model="nova-2",
    language="en-US",
    endpointing="punctuation",
    interim_results=True
)
```

Conversation Timeouts

```
ChatGPTAgentConfig(
    allowed_idle_time_seconds=30,
    num_check_human_present_times=3,
    interrupt_sensitivity="high",
    end_conversation_on_goodbye=True
)
```

Security Considerations

Twilio Request Validation

```
from fastapi.security import HTTPBearer
security = HTTPBearer(auto_error=False)

async def verify_twilio_signature(request: Request):
    validator = RequestValidator(os.environ['TWILIO_AUTH_TOKEN'])
    return validator.validate(
        str(request.url),
        await request.body(),
        request.headers.get('X-TWILIO-SIGNATURE')
    )
```

Secret Management

- Environment variables stored in .env
- Git-ignored from version control
- Encrypted in production using Vault/SSM

TLS Configuration

```
poetry add uvicorn[standard]
uvicorn main:app --ssl-keyfile key.pem --ssl-certfile cert.pem
```

Scaling Strategies

Horizontal Scaling

```
telephony_server = TelephonyServer(
    base_url=os.getenv("LOAD_BALANCER_URL"),
    config_manager=RedisClusterConfigManager(
        nodes=[{"host": "redis-node1", "port": 6379}]
    ),
    inbound_call_configs=[...],
    worker_count=os.cpu_count() * 2
)
```

Performance Metrics

Metric	Monitoring Tool	Alert Threshold
API Latency	Prometheus	>2000ms
Error Rate	Grafana	>5%
Active Calls	CloudWatch	>80% Capacity
CPU Usage	Datadog	>75%

Conclusion

This architecture demonstrates an enterprise-grade telephony solution combining modern Al services with robust infrastructure components. The system's modular design allows for:

1. Component Swapping

- o Replace Deepgram with Azure Speech Services
- Substitute ElevenLabs with Amazon Polly
- o Migrate Redis to managed cloud service

2. Feature Extensions

- Add SMS handling via Twilio Programmable Messaging
- Implement sentiment analysis on conversation transcripts
- Integrate CRM systems through REST API adapters

3. Compliance Enhancements

- GDPR-compliant data anonymization
- HIPAA-ready encryption protocols
- PCI DSS payment processing integration

Future development should focus on implementing adaptive voice models and multi-lingual support while maintaining the system's low-latency characteristics[1][3]. Regular security audits and performance benchmarking are recommended for production deployments[5].