

# Models Documentation - Requirements Extractor

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## Overview

The Requirements Extractor uses a two-stage AI pipeline:

- 1. Stage 1: Transcription** - Converts audio/video to text
  - Local Option:** OpenAI Whisper (openai-whisper library)
  - Cloud Option:** OpenAI Whisper API (whisper-1 model)
- 2. Stage 2: Requirements Extraction** - Extracts structured requirements from text
  - Local Option:** Ollama (llama3.2, mistral, or other local LLMs)
  - Cloud Option:** OpenAI GPT models (gpt-4o-mini, gpt-4o, gpt-3.5-turbo)

## Transcription Models (Audio to Text)

### 1. Local Whisper (openai-whisper)

**Library:** openai-whisper  
**Model Type:** Automatic Speech Recognition (ASR)  
**Default Model Size:** base

#### Model Variants Available

Model Size	Parameters	VRAM Required	Speed	Accuracy	Disk Space
tiny	39M	~1 GB	Fastest	Good	~75 MB
base	74M	~1 GB	Fast	Better	~142 MB
small	244M	~2 GB	Medium	Very Good	~466 MB
medium	769M	~5 GB	Slow	Excellent	~1.5 GB
large	1550M	~10 GB	Slowest	Best	~3 GB
large-v2	1550M	~10 GB	Slowest	Best	~3 GB

large-v3    1550M    ~10 GB    Slowest Best    ~3 GB

### Technical Details

- **Architecture:** Transformer-based encoder-decoder
- **Training Data:** 680,000 hours of multilingual and multitask supervised data
- **Languages Supported:** 99+ languages
- **Audio Formats:** MP3, WAV, M4A, FLAC, and more (via FFmpeg)
- **Processing:** Runs entirely on local machine
- **API Key Required:** No

### Current Implementation

```
# Current default: base model
model = whisper.load_model("base")
result = model.transcribe(audio_path)
text = result.get("text", "")
```

**Code Location:** app.py, function transcribe\_audio\_local\_whisper()

### Performance Characteristics

- **Processing Speed:** ~1x real-time (base model on CPU)
- **GPU Acceleration:** Significantly faster with CUDA/ROCm
- **Memory Usage:** Model size + audio buffer
- **Accuracy:** ~95%+ word accuracy for clear audio

### Advantages

No API key required  
No data sent to external servers (privacy)  
No usage costs  
Works offline  
Supports 99+ languages

### Disadvantages

Slower than cloud API  
Requires significant disk space for larger models  
Requires FFmpeg for audio processing  
Higher memory usage for large models

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## 2. OpenAI Whisper API (whisper-1)

**Service:** OpenAI API

**Model Name:** whisper-1

**Model Type:** Automatic Speech Recognition (ASR)

**Endpoint:** <https://api.openai.com/v1/audio/transcriptions>

### Technical Details

- **Architecture:** Same as local Whisper (large-v2 equivalent)

- **Processing:** Cloud-based, runs on OpenAI servers
- **File Size Limit:** 25 MB per request
- **Audio Formats:** MP3, MP4, MPEG, MPGA, M4A, WAV, WEBM
- **API Key Required:** Yes

### Current Implementation

```
client = OpenAI(api_key=api_key)
transcript = client.audio.transcriptions.create(
    model="whisper-1",
    file=audio_file,
    response_format="text"
)
text = transcript.text
```

**Code Location:** app.py, function transcribe\_audio\_with\_whisper()

### Chunking for Large Files

For files > 25 MB, the system automatically:

1. Splits audio into ~20-minute chunks using pydub
2. Transcribes each chunk separately
3. Combines transcripts in order

**Code Location:** app.py, lines 556-671

### Performance Characteristics

- **Processing Speed:** ~0.1x real-time (much faster than local)
- **Latency:** Network-dependent (typically 2-10 seconds)
- **Accuracy:** ~98%+ word accuracy
- **Cost:** ~\$0.006 per minute of audio

### Advantages

Very fast processing  
 High accuracy  
 No local resources required  
 Automatic chunking for large files  
 No FFmpeg setup needed

### Disadvantages

Requires API key  
 Data sent to OpenAI servers  
 Usage costs  
 Requires internet connection  
 25 MB file size limit per request

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## Requirements Extraction Models (Text to Structured Requirements)

# 1. OpenAI GPT Models

**Service:** OpenAI API  
**Model Type:** Large Language Model (LLM)  
**Endpoint:** <https://api.openai.com/v1/chat/completions>

## Available Models

Model	Parameters	Context Window	Speed	Cost (per 1K tokens)	Best For
gpt-4o-mini	~7B	128K	Fast	\$0.15/\$0.60	Default, cost-effective
gpt-4o	~1.7T	128K	Medium	\$2.50/\$10.00	Higher accuracy needed
gpt-3.5-turbo	~175B	16K	Fastest	\$0.50/\$1.50	Legacy, lower cost

**Default Model:** gpt-4o-mini

## Technical Details

- **Architecture:** Transformer-based decoder
- **Training:** Pre-trained on large text corpus, fine-tuned for chat
- **Response Format:** JSON object (enforced via response\_format)
- **Temperature:** 0.3 (for consistent, deterministic output)
- **System Prompt:** Expert business analyst persona

## Current Implementation

```
response = client.chat.completions.create(  
    model=model, # e.g., "gpt-4o-mini"  
    messages=[  
        {  
            "role": "system",  
            "content": "You are an expert business analyst..."  
        },  
        {  
            "role": "user",  
            "content": prompt  
        }  
    ],  
    response_format={"type": "json_object"},  
    temperature=0.3  
)  
result = json.loads(response.choices[0].message.content)
```

**Code Location:** requirements\_extractor.py, lines 266-283

## Prompt Structure

The system uses a structured prompt that includes:

1. **System Message:** Defines the AI's role as a business analyst
2. **User Prompt:** Contains:
  - Full conversation transcript

- Extraction instructions
- JSON schema specification
- Examples of expected output

**Code Location:** requirements\_extractor.py, function \_create\_extraction\_prompt()

## Chunking Strategy

For large transcripts (>50 messages):

- Splits into chunks of 50 messages
- Processes each chunk independently
- Merges results and deduplicates
- Removes duplicate requirements based on description similarity

**Code Location:** app.py, function extract\_requirements(), lines 814-870

## Performance Characteristics

- **Processing Speed:**
  - gpt-4o-mini: ~2-5 seconds per chunk
  - gpt-4o: ~5-15 seconds per chunk
  - gpt-3.5-turbo: ~1-3 seconds per chunk
- **Token Usage:** ~500-2000 tokens per chunk (input + output)
- **Accuracy:** High for structured extraction tasks

## Advantages

High accuracy for structured extraction  
Fast processing  
Reliable JSON output  
Good at following instructions  
Handles complex requirements well

## Disadvantages

Requires API key  
Usage costs  
Data sent to OpenAI servers  
Requires internet connection

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## 2. Ollama (Local LLMs)

**Service:** Local Ollama server

**Model Type:** Large Language Model (LLM)

**Endpoint:** http://localhost:11434/api/generate

### Available Models

Model	Parameters	Context Window	VRAM Required	Best For
llama3.2	3B	128K	~4 GB	Default, balanced

llama3.2:3b 3B	128K	~4 GB	Smaller variant
mistral 7B	8K	~6 GB	Alternative option
codellama 7B-34B	16K-100K	6-20 GB	Code-focused
llama3 8B-70B	8K-128K	8-40 GB	Larger options

**Default Model:** llama3.2

### Technical Details

- **Architecture:** Various (depends on model)
- **Processing:** Runs entirely on local machine
- **API:** OpenAI-compatible API via Ollama server
- **Temperature:** 0.3 (for consistency)
- **Timeout:** 300 seconds (5 minutes) per request

### Current Implementation

```
response = requests.post(
    "http://localhost:11434/api/generate",
    json={
        "model": ollama_model, # e.g., "llama3.2"
        "prompt": full_prompt,
        "stream": False,
        "options": {
            "temperature": 0.3
        }
    },
    timeout=300
)
result_text = response.json().get("response", "")
# Extract JSON from response (may have extra text)
json_match = re.search(r'\{.*\}', result_text, re.DOTALL)
result = json.loads(json_match.group())
```

**Code Location:** requirements\_extractor.py, lines 263-283

### Health Check

The system checks if Ollama is running before use:

```
response = requests.get("http://localhost:11434/api/tags", timeout=3)
if response.status_code == 200:
    # Ollama is available
    models = response.json().get('models', [])
```

**Code Location:** requirements\_extractor.py, lines 184-199

### Performance Characteristics

- **Processing Speed:**
  - llama3.2 (3B): ~5-20 seconds per chunk (CPU)

- llama3.2 (3B): ~2-5 seconds per chunk (GPU)
- Larger models: Slower but more accurate
- **Memory Usage:** Model size + context buffer
- **Accuracy:** Good for structured tasks, may need prompt tuning

## Advantages

No API key required  
 No data sent externally (privacy)  
 No usage costs  
 Works offline  
 Full control over model selection

## Disadvantages

Requires local installation and setup  
 Slower than cloud API (especially on CPU)  
 Requires significant RAM/VRAM  
 May need prompt engineering for best results  
 JSON extraction may need regex parsing (less reliable)

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# Text Parsing (Post-Transcription)

After transcription, the raw text is parsed into structured message format before requirements extraction.

## Parsing Methods

### 1. Simple Text Parsing (for Whisper output)

Whisper outputs plain text without speaker identification. The system:

```
lines = transcript_text.split('\n')
messages = []
for line in lines:
    line = line.strip()
    if line:
        messages.append({
            'speaker': 'Speaker', # Generic speaker
            'text': line,
            'timestamp': None
        })
```

**Code Location:** app.py, lines 752-761

### 2. Structured Transcript Parsing

For pre-formatted transcripts (TXT, VTT, JSON), the TranscriptParser class extracts:

- **Speaker names:** From patterns like "Speaker: text" or "[Speaker] text"
- **Timestamps:** From VTT format or JSON structure
- **Message text:** Cleaned and formatted

**Code Location:** requirements\_extractor.py, class TranscriptParser

**Parsing Patterns**

**Text Format:**

John Doe: We need **user authentication**

Jane Smith: That's a good point

**VTT Format:**

WEBVTT

00:00:10.000 --> 00:00:15.000

John Doe: We need user authentication

00:00:15.000 --> 00:00:20.000

Jane Smith: That's a good point

**JSON Format:**

```
{
  "messages": [
    {
      "speaker": "John Doe",
      "text": "We need user authentication",
      "timestamp": "00:00:10"
    }
  ]
}
```

**Model Comparison**

**Transcription Models**

Feature	Local Whisper (base)	OpenAI Whisper API
Speed	~1x real-time	~0.1x real-time
Accuracy	~95%	~98%
Cost	Free	~\$0.006/min
Privacy	Local only	Cloud
Setup	Requires FFmpeg	Just API key
File Size	Unlimited	25 MB limit
Languages	99+	99+
Offline	Yes	No

**Requirements Extraction Models**

Feature	OpenAI GPT (gpt-4o-mini)	Ollama (llama3.2)
Speed	~2-5 sec/chunk	~5-20 sec/chunk (CPU)
Accuracy	High	Good



<b>Cost</b>	~\$0.15/1K tokens	Free
<b>Privacy</b>	Cloud	Local
<b>Setup</b>	Just API key	Install + setup
<b>JSON Reliability</b>	Excellent	⚠ May need parsing
<b>Context Window</b>	128K	128K
<b>Offline</b>	No	Yes

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## Model Selection Guide

### When to Use Local Whisper

You want complete privacy  
 You process many files (cost savings)  
 You have sufficient disk space  
 You don't mind slower processing  
 You work offline frequently

### When to Use OpenAI Whisper API

You need fast processing  
 You want highest accuracy  
 You have API budget  
 You process occasional files  
 You want minimal setup

### When to Use OpenAI GPT Models

You need highest extraction accuracy  
 You want reliable JSON output  
 You have API budget  
 You process many different requirement types  
 You want consistent results

### When to Use Ollama

You want complete privacy  
 You process many files (cost savings)  
 You have sufficient RAM/VRAM  
 You don't mind slower processing  
 You want full control

## Recommended Combinations

### Privacy-First Setup

- Local Whisper + Ollama
- Complete privacy
- No costs
- ⚠ Slower processing

## Balanced Setup

- Local Whisper + OpenAI GPT
- Privacy for transcription
- Fast, accurate extraction
- ⚠ Extraction costs

## Speed-First Setup

- OpenAI Whisper API + OpenAI GPT
- Fastest overall
- Highest accuracy
- All data to cloud
- Highest costs

## Cost-Effective Setup

- Local Whisper + Ollama
  - No costs
  - Privacy
  - ⚠ Requires local resources
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# Technical Specifications

## Model Input/Output Formats

### Whisper Input

- **Format:** Audio file (MP3, WAV, M4A, etc.)
- **Size:** Unlimited (local), 25 MB (API)
- **Sample Rate:** Auto-detected
- **Channels:** Mono or stereo

### Whisper Output

- **Format:** Plain text string
- **Language:** Detected automatically
- **Punctuation:** Included
- **Speaker IDs:** Not included (single speaker assumed)

### GPT/Ollama Input

- **Format:** Text string (conversation transcript)
- **Structure:** Formatted as conversation
- **Length:** Up to context window limit
- **Encoding:** UTF-8

### GPT/Ollama Output

- **Format:** JSON object
- **Structure:**

```
{
  "functional_requirements": [...],
  "non_functional_requirements": [...],
  "business_rules": [...],
  "action_items": [...],
  "decisions": [...],
  "stakeholders": [...]
}
```

- **Validation:** JSON schema validation

## Token Usage Estimates

### Transcription

- **Local Whisper:** No tokens (local processing)
- **OpenAI Whisper API:** No tokens (separate pricing)

### Requirements Extraction

- **Input:** ~500-2000 tokens per chunk
- **Output:** ~200-800 tokens per chunk
- **Total:** ~700-2800 tokens per chunk

**Example:** 100-message transcript (2 chunks)

- Input: ~2000 tokens
- Output: ~800 tokens
- Total: ~2800 tokens
- Cost (gpt-4o-mini): ~\$0.42

## Memory Requirements

### Local Whisper

- **Base Model:** ~1 GB RAM
- **Small Model:** ~2 GB RAM
- **Medium Model:** ~5 GB RAM
- **Large Model:** ~10 GB RAM

### Ollama

- **llama3.2 (3B):** ~4 GB RAM/VRAM
- **mistral (7B):** ~6 GB RAM/VRAM
- **llama3 (8B):** ~8 GB RAM/VRAM

## Processing Time Estimates

**Small File (5 minutes audio, 50 messages)**

- **Transcription (Local):** ~5 minutes
- **Transcription (API):** ~30 seconds
- **Extraction (GPT):** ~5 seconds

- **Extraction (Ollama):** ~10-20 seconds
- **Total (Local):** ~5-6 minutes
- **Total (API):** ~35 seconds

#### Large File (60 minutes audio, 500 messages)

- **Transcription (Local):** ~60 minutes
  - **Transcription (API):** ~6 minutes
  - **Extraction (GPT):** ~50 seconds (10 chunks)
  - **Extraction (Ollama):** ~2-5 minutes (10 chunks)
  - **Total (Local):** ~62-65 minutes
  - **Total (API):** ~7 minutes
- 

## Appendix

### A. Model Version History

#### Whisper

- **v1:** Initial release (2022)
- **v2:** Improved accuracy (2023)
- **v3:** Latest version (2024)

#### GPT Models

- **gpt-3.5-turbo:** Legacy model
- **gpt-4:** Previous generation
- **gpt-4o:** Current generation (optimized)
- **gpt-4o-mini:** Smaller, faster variant

#### Ollama Models

- **llama3.2:** Latest (2024)
- **llama3:** Previous generation
- **mistral:** Alternative option

### B. API Rate Limits

#### OpenAI Whisper API

- **Free Tier:** Limited requests
- **Paid Tier:** Based on usage
- **Rate Limits:** Vary by account tier

#### OpenAI GPT API

- **Rate Limits:** Based on account tier
- **Free Tier:** 3 requests/minute
- **Paid Tier:** Higher limits

#### Ollama

- **No Rate Limits:** Local processing
- **Concurrent Requests:** Limited by hardware

### C. Error Handling

The system includes comprehensive error handling for:

- Model loading failures
- API connection errors
- Invalid responses
- JSON parsing errors
- Timeout handling
- Rate limit handling

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## Document History

Version	Date	Changes
1.0	Nov 2024	Initial documentation

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End of Models Documentation