

Transcribe RO - Project Summary

Overview

Transcribe RO is a portable command-line tool designed to transcribe audio recordings in any language and translate them to Romanian. Built with OpenAI Whisper and designed for portability, it can run directly from a USB flash drive with minimal setup.

Key Features

Core Functionality

- **Automatic Language Detection:** Detects source language from 99+ supported languages
- **High-Quality Transcription:** Uses OpenAI Whisper AI models
- **Romanian Translation:** Automatically translates to Romanian using Google Translate
- **Multi-Format Audio Support:** MP3, WAV, M4A, FLAC, OGG, AAC, OPUS, and more
- **Multiple Output Formats:** TXT, JSON, SRT (subtitles), VTT (web subtitles)
- **Timestamp Support:** Detailed timestamps for each transcription segment
- **Portable Design:** Runs from flash drive without installation
- **User-Friendly CLI:** Simple command-line interface with helpful options

Technical Features

- **Multiple Model Sizes:** Choose from tiny, base, small, medium, or large models
- **GPU Support:** Optional CUDA acceleration for faster processing
- **Offline Capable:** Works offline after initial model download
- **Error Handling:** Graceful error handling with clear messages
- **Cross-Platform:** Works on Windows, macOS, and Linux

Project Structure

```

transcribe_ro/
├── transcribe_ro.py          # Main application script
├── requirements.txt           # Python dependencies
├── README.md                 # Comprehensive documentation
├── QUICKSTART.md             # Quick start guide for beginners
├── TESTING.md                # Testing guide and test plans
├── CONTRIBUTING.md           # Contribution guidelines
├── LICENSE                   # MIT License
└── .gitignore                # Git ignore rules

├── setup_portable.bat        # Windows portable setup script
└── setup_portable.sh         # Linux/macOS portable setup script

├── run_transcribe.bat        # Windows helper script
└── run_transcribe.sh         # Linux/macOS helper script

examples.sh                  # Example commands

```

Technical Stack

Core Dependencies

- **OpenAI Whisper**: State-of-the-art speech recognition
- **PyTorch**: Machine learning framework
- **googletrans**: Translation service
- **FFmpeg**: Audio processing

Python Version

- Python 3.8 or higher

Use Cases

1. **Conference Recordings**: Transcribe multilingual meetings
2. **Interviews**: Convert interviews to searchable text
3. **Lectures & Education**: Create lecture transcripts
4. **Podcasts**: Generate accessible transcripts
5. **Voice Notes**: Convert voice memos to text
6. **Video Subtitles**: Generate subtitle files
7. **Legal/Medical**: Professional recording transcription

Getting Started

Quick Installation

```
# Clone the repository
git clone https://github.com/yourusername/transcribe_ro.git
cd transcribe_ro

# Run setup
# Windows: setup_portable.bat
# Linux/macOS: ./setup_portable.sh

# Transcribe an audio file
python transcribe_ro.py audio.mp3
```

Basic Usage Examples

```
# Basic transcription with translation
python transcribe_ro.py audio.mp3

# Fast transcription (tiny model)
python transcribe_ro.py audio.mp3 --model tiny

# High accuracy (medium model)
python transcribe_ro.py audio.mp3 --model medium

# No translation (original language only)
python transcribe_ro.py audio.mp3 --no-translate

# Generate subtitles
python transcribe_ro.py audio.mp3 --format srt

# Export as JSON
python transcribe_ro.py audio.mp3 --format json
```

Model Selection

Model	Size	Speed	Accuracy	Use Case
tiny	~75MB	Fastest	Good	Quick tests
base	~150MB	Fast	Better	Default
small	~500MB	Moderate	Great	Quality work
medium	~1.5GB	Slow	Excellent	Professional
large	~3GB	Slowest	Best	Maximum accuracy

Output Formats

TXT (Default)

- Metadata section
- Original transcription
- Romanian translation
- Timestamps for each segment

JSON

- Structured data format
- Includes all metadata
- Segment-by-segment information
- Easy to parse programmatically

SRT/VTT

- Standard subtitle formats
- Timestamp-synced text
- Video subtitle compatible
- Optional translation

Portability Features

Flash Drive Setup

- Run from USB without installation
- Standalone Python environment
- Self-contained dependencies
- Cross-computer compatibility

Offline Capability

- Works offline after model download
- Transcription fully offline
- Translation requires internet
- Use `--no-translate` for complete offline use

Performance Considerations

Speed Factors

- **Model Size:** Smaller = faster
- **Audio Length:** Longer = more time
- **Audio Quality:** Better quality = better accuracy
- **Hardware:** GPU acceleration available

Typical Processing Times

(Base model on CPU)

- 1 minute audio: ~30 seconds
- 5 minute audio: ~2-3 minutes
- 30 minute audio: ~10-15 minutes

Error Handling

The tool includes comprehensive error handling for:

- Missing audio files
- Unsupported formats
- Network issues (translation)
- Insufficient memory
- Corrupted audio files
- Missing dependencies

Future Enhancements

Planned Features

- Batch processing multiple files
- Progress bar for long transcriptions
- Configuration file support
- Better memory management
- Resume interrupted transcriptions

Potential Improvements

- Additional translation services (DeepL)
- GUI version
- Audio preprocessing (noise reduction)
- Speaker diarization
- Real-time transcription

Development

Contributing

Contributions are welcome! See [CONTRIBUTING.md](#) (CONTRIBUTING.md) for guidelines.

Testing

Comprehensive testing guide available in [TESTING.md](#) (TESTING.md).

Version Control

- Git repository initialized
- Commit history maintained
- Branch-based development encouraged

Documentation

Available Guides

1. **README.md:** Complete documentation
2. **QUICKSTART.md:** Quick start for beginners
3. **TESTING.md:** Testing procedures
4. **CONTRIBUTING.md:** Contribution guidelines
5. **examples.sh:** Command examples

Documentation Quality

- Clear and comprehensive
- Examples for all features
- Troubleshooting guides
- Multi-platform instructions

License

MIT License - Free to use, modify, and distribute.

Acknowledgments

- OpenAI for Whisper model
- FFmpeg project
- Google Translate service
- Open source community

Support

For issues or questions:

- Check documentation
- Review existing issues
- Open new issue with details
- Provide reproduction steps

Statistics

- **Lines of Code:** ~500+ (main script)
- **Supported Languages:** 99+ (Whisper)
- **Audio Formats:** 8+ major formats
- **Output Formats:** 4 (TXT, JSON, SRT, VTT)
- **Model Choices:** 5 size options
- **Platforms:** Windows, macOS, Linux

Target Audience

- **Primary:** Romanian speakers needing transcription/translation
- **Secondary:** Anyone needing multilingual transcription
- **Technical Level:** Beginner to advanced
- **Professional Use:** Suitable for professional work

Competitive Advantages

1. **Portability:** Runs from flash drive
2. **Offline Capable:** Works without internet (after setup)
3. **Free & Open Source:** No subscription fees
4. **Multi-language:** 99+ languages supported
5. **Professional Quality:** State-of-the-art AI models
6. **Flexible Output:** Multiple format options
7. **Easy to Use:** Simple CLI interface
8. **Well Documented:** Comprehensive guides

Project Status

 **COMPLETE** - Ready for production use

All core features implemented:

-  Audio transcription
-  Language detection
-  Romanian translation
-  Multiple formats
-  Portable setup
-  Documentation
-  Error handling
-  Testing guides
-  Version control

Contact & Links

- **Repository:** [GitHub - transcribe_ro](https://github.com/yourusername/transcribe_ro) (https://github.com/yourusername/transcribe_ro)
- **Issues:** GitHub Issues
- **Discussions:** GitHub Discussions
- **License:** MIT License

Project Created: January 2026

Version: 1.0.0

Status: Production Ready

Made with  for the Romanian community