



序号	引擎类型	IP地址	端口号	类型	授权路数	版本	引擎语言	声纹识别	状态	操作
1	-	172.16.19.246	8800	同传服务	-	v3.1.2	-	-	在线	重启
2	TS-8611R	127.0.0.1	30051	实时转写语音...	2	v3.1.0	粤英	无	在线	重启 删除
3	TS-8611R	172.16.19.12	30051	实时转写语音...	2	v3.1.0	中英	有	在线	重启 删除

## Description

Speech transcription conference background service software, used with the speech recognition engine, can provide real-time speech recognition and recording file recognition capabilities, and can realize real-time speech transcription and audio file import and transcription functions. When used with traditional conference systems, it can realize the conference recording function, making the conference more efficient and convenient. Characterized by private deployment + intelligent speech recognition technology + 98% accuracy rate, it is suitable for confidential units such as government departments and banking departments, etc.

## Feature

- \* Support the management function of the speech recognition engine, uniformly dispatch the registered speech recognition engine, manage and distribute the text stream information returned by the speech recognition engine.
- \* Support the conversion of real-time external audio signals into text, and the transcribed text can be automatically separated according to the role of the speaker, making the content clearer and more intuitive. Support automatic segmentation, filtering of tone words, blocking of taboo words and adding system hot words to optimize text display. Support marking of important or doubtful content in the text, convenient for organizing the manuscript after the meeting. Support functions such as searching text keywords and displaying text by speaker. Support text editing, allowing editing of single sentences and correction of typos.
- \* Support voiceprint recognition of roles in the participant list, so that text roles are automatically separated during the process of voice-to-text conversion. The voiceprint needs to be registered and entered in advance, and the recording file registration is supported. (Voiceprint engine needs to be purchased separately)
- \* Support adding Chinese, dialect, foreign language and other speech engines in the same meeting to realize speech-to-text conversion. The administrator can bind the language engine required by the participants. (The speech recognition engine of the corresponding language needs to be purchased, and it supports Mandarin by default.)
- \* Support recording conference audio, you can view basic information such as the name, size, audio length, speaker, and time of the recorded audio. Support selecting different speakers to export audios separately or exporting all according to the entire meeting, and support downloading audio files via QR code. The file supports audio formats such as wav and mp3.
- \* Support batch import of audio files in various formats such as mp3, wav, wma, mp4, avi, pcm, m4a, etc. and quickly transcribe them into text; support language selection and modal particle filter settings; support batch download or QR code sharing of audio and transcribed text files, and deleting invalid files. You can view information such as audio file name, file size, audio length, upload time, voice selection, and transcription status. The system supports uploading up to 50 audio files each time, with a size of up to 5GB; a single piece of data can last up to 18 hours, and the size can reach 2G.
- \* Audio import files can be quickly transcribed, and 1-hour audio can be transcribed in 5-10 minutes. (Additional speed engine is required.)
- \* Multiple meeting rooms are supported to use the same speech transcription server, and a single management software can manage the speech transcription systems of multiple meeting rooms at the same time. (Speech recognition engine corresponding to the number of conference rooms needs to be purchased)
- \* Support preset modal particles. After the modal particle filter is turned on, the modal particles in the transcribed text will be deleted to ensure a smoother manuscript.



# Speech Transcription Conference Background Service Software V5.61

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## TS-8670NR

- \* Support adding or deleting proper nouns such as personalized names, place names, building names, and professional vocabulary in batches. After the hot word optimization is turned on before the meeting, the accuracy of speech recognition of the word can be greatly improved.
- \* Support adding sensitive words that are not suitable for display to the list of taboo words to block. When such words are recognized, the system provides three display modes: no display, \* and space.
- \* Support automatic segmentation according to intelligent semantics (only Chinese and English are supported), VAD+word count, VAD+keyword, keyword and other methods, and the speech pause time and word count can be customized.
- \* Support viewing daily conference system operation logs.
- \* Support custom time to automatically clean up regularly or save meeting recording files permanently.
- \* Support statistics of meeting data such as transcription time, number of documents, number of meetings, number of participants, etc., and support presentation in the form of histograms and line charts. It supports querying and exporting data by week, month, year and custom time period.