

Voice Notepad

User Manual v3

Version 1.9.11 · December 2025

DUAL-PIPELINE ARCHITECTURE

Local Preprocessing + Cloud Transcription

PLATFORMS

Linux Debian/Ubuntu Wayland

Author: Daniel Rosehill
Repository: github.com/danielrosehill/Voice-Notepad
License: MIT

Table of Contents

1	Introduction	5
1.1	Design Philosophy	5
1.2	What Makes This Different	5
1.3	Key Features	5
2	The Dual-Pipeline Architecture	6
2.1	Pipeline Overview	6
2.2	Stage 1: Local Preprocessing	6
2.3	Stage 2: Cloud Transcription	7
3	Installation	9
3.1	Stage 1: System Dependencies	9
3.2	Stage 2: Application Installation	9
3.2.1	Option A: Debian Package (Recommended)	9
3.2.2	Option B: AppImage	9
3.2.3	Option C: From Source	9
4	Hardware Recommendations	10
4.1	Design Rationale: F13-F24 Keys	10
4.2	Recommended Hardware	10
4.2.1	Simple: USB HID Button (\$5)	10
4.2.2	Full Setup: Macro Pad	10
4.2.3	Alternative: Foot Pedal	11
4.3	Setting Up Key Mapping	11
5	Configurable Hotkeys	12
5.1	Default Mappings	12
5.2	Configuration	12
5.3	Technical Notes	12
6	Audio Feedback	13
6.1	Modes	13
6.2	TTS Announcements	13
7	Transcript History	14
7.1	Database	14
7.2	History Tab Features	14
8	Analytics & Cost Tracking	15
8.1	Analytics Tab	15
8.2	Export Statistics	15
8.3	Cost Tab	15
8.4	Cost Effectiveness	15
9	Text Injection (Wayland)	16
9.1	Requirements	16
9.2	Quick Setup	16
9.3	Verify	16
9.4	Persistent Setup	16
10	Output Modes	17

11 Storage Locations	18
12 Troubleshooting	19
12.1 Audio Issues	19
12.2 API Issues	19
12.3 Hotkeys Not Working	19
12.4 Text Injection Not Working	19
12.5 Why Input Remapper?	20
12.6 Installation	20
12.7 Step 1: Select Your Device	20
12.8 Step 2: Select the Specific Interface	21
12.9 Step 3: Open the Editor	22
12.10 Step 4: Record the Input	22
12.11 Step 5: Choose the Output Key	23
12.12 Step 6: Apply and Enable Autoload	24
12.13 Recommended Multi-Button Setup	24
12.14 Core Technologies	25
12.15 Audio Processing	25
12.16 Text-to-Speech	26
12.17 Data Storage	26
12.18 Input Handling	26
12.19 Visualization	26
12.20 System Dependencies	27
12.21 Component Links	27

AI-Human Co-Authorship

This software was developed through **AI-human collaboration**.

The code was generated by **Claude Opus 4.5** and other Anthropic models under my direction and supervision. I designed the architecture, specified requirements, reviewed outputs, and guided the implementation. Claude wrote the code.

This represents an experimental approach to software development. I believe it produces high-quality results when the human provides clear direction, domain expertise, and ongoing oversight.

The core design philosophy—**audio multimodal first**—came from my experience with traditional ASR-then-LLM workflows and recognizing that multimodal models could do both in a single pass. The iterative development process has been guided by my own daily use of the application.

— Daniel Rosehill, December 2025

Introduction

Voice Notepad is an **audio multimodal first** desktop application for voice transcription. The core design philosophy is to send audio directly to multimodal AI models that can “hear” and process audio alongside text instructions—rather than traditional ASR-then-LLM approaches.

1.1 Design Philosophy

This is an **experimental, iterative software design process** based on my own need for a better transcription tool. In the first few weeks of use, the approach has been validated with:

- **Over 2,000 transcriptions** processed
- **More than 1 million characters** of output
- Excellent accuracy and formatting results
- API costs of just a few dollars total

The key insight is that multimodal models can do both transcription and cleanup in a single pass, while also “hearing” tone, emphasis, and verbal commands that get lost in text-only processing.

1.2 What Makes This Different

Traditional Approach

1. Record audio
2. Send to ASR (Whisper, etc.)
3. Get raw transcript
4. Send text to LLM for cleanup
5. Get formatted output

Two API calls, higher cost

Voice Notepad Approach

1. Record audio
2. Local preprocessing (VAD + AGC)
3. Send audio + prompt to Gemini
4. Get formatted output

Single API call, lower cost
AI “hears” your voice

1.3 Key Features

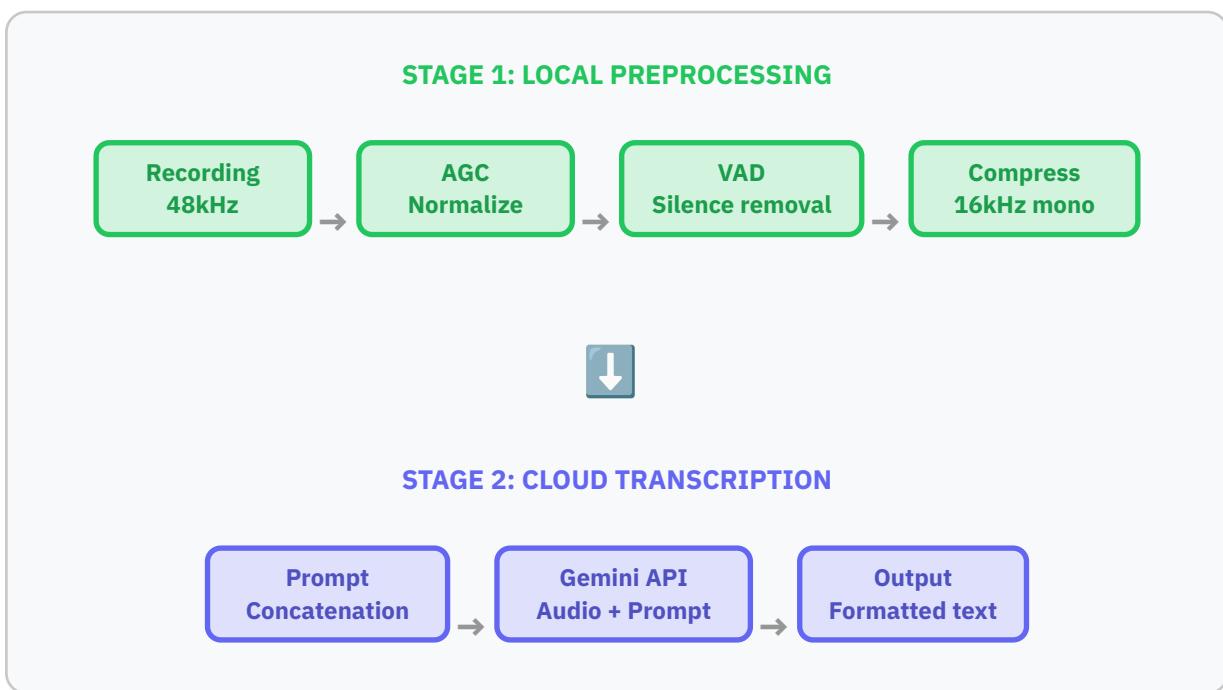
- Dual-pipeline processing
- Voice Activity Detection (VAD)
- Automatic Gain Control (AGC)
- Single-pass multimodal transcription
- Layered prompt concatenation
- Configurable hotkeys (F13-F24)

- Text injection (auto-paste)
- Audio feedback (beeps or TTS)
- Transcript history with search
- Usage analytics and charts
- Cost tracking
- Statistics export

The Dual-Pipeline Architecture

Voice Notepad's effectiveness comes from combining local preprocessing with cloud-based multimodal transcription.

2.1 Pipeline Overview



The combination achieves:

- **Cost reduction:** VAD removes 30-80% of audio (silence/pauses)
- **Quality improvement:** AGC ensures consistent input levels
- **Precise formatting:** Layered prompts produce exactly the output you need
- **Single API call:** Multimodal models transcribe and format in one pass

2.2 Stage 1: Local Preprocessing

The local pipeline processes audio on your machine before any cloud upload.

1.1 Recording (PyAudio)

- Captures at device's native sample rate (typically 48kHz)
- 16-bit PCM, mono audio
- Automatic sample rate negotiation

- Handles microphone disconnection gracefully

1.2 Automatic Gain Control (AGC)

Normalizes audio levels for consistent transcription accuracy.

Parameter	Value	Purpose
Target peak	-3 dBFS	Optimal level with headroom
Min threshold	-40 dBFS	Skip if quieter than noise floor
Max gain	+20 dB	Prevent over-amplification

Behavior: Only boosts quiet audio—never attenuates loud audio.

1.3 Voice Activity Detection (VAD)

Removes silence using TEN VAD before API upload.

Parameter	Value	Purpose
Sample rate	16kHz	Required by TEN VAD
Hop size	256 samples	16ms analysis windows
Threshold	0.5	Speech probability cutoff
Min speech	250ms	Ignore very short sounds
Padding	30ms	Buffer around speech

Typical reduction: 30-80% depending on speaking pattern.

1.4 Compression

- Downsampled to 16kHz mono (matches Gemini's internal format)
- Converted to WAV for API upload
- 66% smaller than 48kHz stereo input

2.3 Stage 2: Cloud Transcription

2.1 Prompt Concatenation

Instructions are built from multiple layers:

Foundation Layer (always applied):

- Remove filler words (um, uh, like, you know)
- Add punctuation and paragraph breaks
- Follow verbal commands (“scratch that”, “new paragraph”)
- Fix grammar and spelling

Format Layer (based on preset):

- Email, meeting notes, bullet points, documentation, etc.

Style Layer:

- Formality: casual, neutral, professional
- Verbosity: none to maximum reduction

Personalization:

- Email signatures injected for email format

2.2 Multimodal API Submission

- Audio is base64-encoded with concatenated prompt
- Gemini processes audio + text instructions together
- The AI “hears” tone, emphasis, and verbal commands
- Returns formatted text in a single API call

Installation

Voice Notepad requires a two-stage installation: system dependencies first, then the application.

3.1 Stage 1: System Dependencies

Install required system packages on Ubuntu/Debian:

```
sudo apt install python3 python3-venv ffmpeg portaudio19-dev libc++1
```

Package	Purpose
python3, python3-venv	Python runtime and virtual environments
ffmpeg	Audio format conversion and compression
portaudio19-dev	PyAudio recording library headers
libc++1	Required by TEN VAD for voice activity detection

3.2 Stage 2: Application Installation

3.2.1 Option A: Debian Package (Recommended)

```
sudo apt install ./voice-notepad_1.9.11_amd64.deb
```

3.2.2 Option B: AppImage

```
chmod +x Voice_NotePad-1.9.11-x86_64.AppImage
./Voice_NotePad-1.9.11-x86_64.AppImage
```

3.2.3 Option C: From Source

```
git clone https://github.com/danielrosehill/Voice-Notepad.git
cd Voice-Notepad
./run.sh
```

Hardware Recommendations

Voice Notepad is designed to work with dedicated hardware for hands-free operation.

4.1 Design Rationale: F13-F24 Keys

The application uses **F13 through F24** as the default hotkey range for an important reason: **these keys are defined in Linux but virtually never used by applications.**

This design choice ensures:

- **No conflicts** with user-level programs (browsers, editors, etc.)
- **No conflicts** with desktop environment shortcuts
- **System-level interception** without interfering with normal keyboard use
- **Clean separation** between transcription controls and regular typing

Most standard keyboards only have F1-F12. To use Voice Notepad's hotkeys, you need either a keyboard with extended function keys or a separate input device mapped to F13-F24.

4.2 Recommended Hardware

4.2.1 Simple: USB HID Button (\$5)

A single USB HID programmable button (available on AliExpress for \$5) provides excellent **push-to-talk (PTT)** functionality.

- Map the button to **F15** (Toggle Recording)
- Press to start recording, press again to transcribe
- Simple, reliable, inexpensive
- No software required after initial mapping

4.2.2 Full Setup: Macro Pad

For power users, a USB macro pad with 6+ keys enables the full hotkey workflow:

Button	Maps To	Function
1	F15	Toggle (start/stop and transcribe)
2	F16	Tap Toggle (start/stop and cache)
3	F17	Transcribe cached audio
4	F18	Clear recording

Button	Maps To	Function
5	F19	Append to cache
6	F20	Pause/resume

The number of configurable hotkeys (6 functions) is specifically designed to match typical macro pad layouts.

4.2.3 Alternative: Foot Pedal

USB foot pedals are popular for hands-free transcription:

- Keep hands on keyboard while controlling recording
- Map pedal buttons to F15/F17/F18
- Available from \$15-30

4.3 Setting Up Key Mapping

Use **Input Remapper** on Linux to map your device:

```
sudo apt install input-remapper
```

1. Open Input Remapper
2. Select your USB device
3. Click “Record” and press the button
4. Set output to KEY_F15 (or desired F-key)
5. Click Apply and enable Autoload

Configurable Hotkeys

5.1 Default Mappings

Key	Function	Description
F15	Toggle	Start recording, or stop and transcribe immediately
F16	Tap Toggle	Start recording, or stop and cache
F17	Transcribe	Transcribe cached audio
F18	Clear	Delete recording and clear cache
F19	Append	Start recording that appends to cache
F20	Pause	Pause/resume current recording

5.2 Configuration

Go to **Settings → Hotkeys** to customize mappings. Each function can be assigned any key from F13-F24, or disabled entirely.

5.3 Technical Notes

- On Linux/Wayland: Hotkeys work via evdev (reads directly from input devices)
- Requires user to be in the `input` group: `sudo usermod -aG input $USER`
- Falls back to pynput/X11 on non-Linux systems

Audio Feedback

Voice Notepad provides audio notifications for recording events.

6.1 Modes

Configure in **Settings → Behavior → Audio feedback:**

Mode	Description
Beeps	Short beep tones for events (default)
Voice (TTS)	Spoken announcements via Edge TTS
Silent	No audio feedback

6.2 TTS Announcements

When Voice mode is enabled, you'll hear spoken feedback:

Event	Announcement
Recording started	“Recording”
Recording stopped	“Recording stopped”
Audio cached	“Cached”
Transcription started	“Transcribing”
Transcription complete	“Complete”
Text copied	“Text on clipboard”
Recording cleared	“Recording discarded”
Error occurred	“Error”

TTS uses pre-generated audio files with a British English male voice (en-GB-RyanNeural via Microsoft Edge TTS).

Transcript History

All transcriptions are automatically saved to a local MongoDB-compatible database.

7.1 Database

Voice Notepad uses **Mongita**, a pure Python MongoDB implementation. Data is stored locally at `~/.config/voice-notepad-v3/mongita/`.

Each transcript record includes:

- Full transcript text
- Timestamp
- Provider and model used
- Audio duration (original and after VAD)
- Inference time
- Token usage and cost
- Word and character counts

7.2 History Tab Features

- **Full-text search** across all transcriptions
- **Click to preview** any transcript
- **Double-click to load** into editor for reuse
- **Delete individual** transcriptions
- **Delete all** with confirmation

Analytics & Cost Tracking

8.1 Analytics Tab

The Analytics tab provides performance insights:

- **Summary statistics:** Total transcriptions, words, characters
- **Daily activity chart:** Visual bar chart with metric toggles
- **Model performance table:** Compare inference times across models
- **Time period selector:** Today, 7 days, 30 days, or all time

8.2 Export Statistics

Click **Export Stats** to save anonymized statistics as JSON. Exported data includes:

- Transcription counts and volume
- Model performance metrics (no transcript content)
- Daily activity breakdown

Useful for benchmarking or sharing performance data.

8.3 Cost Tab

For OpenRouter users, the Cost tab shows:

- **Account balance:** Live credit balance
- **Key-specific usage:** Daily, weekly, monthly spend
- **Model breakdown:** Usage by model

8.4 Cost Effectiveness

Real usage data with Gemini 2.5 Flash:

- 848 transcriptions for \$1.17 total
- 84,000 words transcribed and cleaned
- About \$0.014 per 1,000 words (1.4 cents)

Text Injection (Wayland)

Text injection automatically pastes transcribed text at your cursor after transcription.

9.1 Requirements

- `ydotool` package installed
- `ydotoold` daemon running as your user (not root)

9.2 Quick Setup

```
sudo apt install ydotool
sudo pkill ydotoold
sudo rm -f /tmp/.ydotool_socket
ydotoold &
```

9.3 Verify

```
ls -la /tmp/.ydotool_socket
# Should show YOUR username as owner, not root
```

9.4 Persistent Setup

Create `~/.config/systemd/user/ydotoold.service`:

```
[Unit]
Description=ydotool daemon

[Service]
ExecStart=/usr/bin/ydotoold

[Install]
WantedBy=default.target
```

Enable: `systemctl --user enable --now ydotoold`

Output Modes

Three independent output modes can be combined:

Mode	Behavior
App	Text appears in the application window
Clipboard	Text copied to system clipboard
Inject	Text typed at cursor via ydotool

Enable any combination: all three, any two, or just one.

Storage Locations

All data is stored in `~/.config/voice-notepad-v3/`:

Path	Contents
<code>config.json</code>	API keys and preferences
<code>mongita/</code>	MongoDB-compatible transcript database
<code>usage/</code>	Daily cost tracking JSON files
<code>audio-archive/</code>	Opus audio recordings (if enabled)

Troubleshooting

12.1 Audio Issues

No microphone detected: Check `pactl list sources short`. Verify PipeWire is running.

Poor quality: Enable AGC in Settings → Behavior. Position microphone closer.

12.2 API Issues

Transcription fails: Verify API key. Check internet. Try different model.

High costs: Enable VAD. Use Flash Lite models.

12.3 Hotkeys Not Working

Add yourself to input group: `sudo usermod -aG input $USER` then log out/in.

12.4 Text Injection Not Working

Verify `ydotoold` is running as your user, not root. Check socket ownership.

Appendix A

Hotkey Setup Guide

This appendix provides a visual walkthrough for setting up global hotkeys using Input Remapper on Linux.

12.5 Why Input Remapper?

[Input Remapper](#) is an open-source tool for remapping keys and buttons from any input device. It works with USB foot pedals, macro keypads, extra mouse buttons, and any HID device.

12.6 Installation

Ubuntu/Debian

```
sudo apt install \  
    input-remapper
```

Fedora

```
sudo dnf install \  
    input-remapper
```

Arch

```
sudo pacman -S \  
    input-remapper-git
```

12.7 Step 1: Select Your Device

Open Input Remapper and select the device you want to remap. This could be a USB foot pedal, macro keypad, or any HID device.

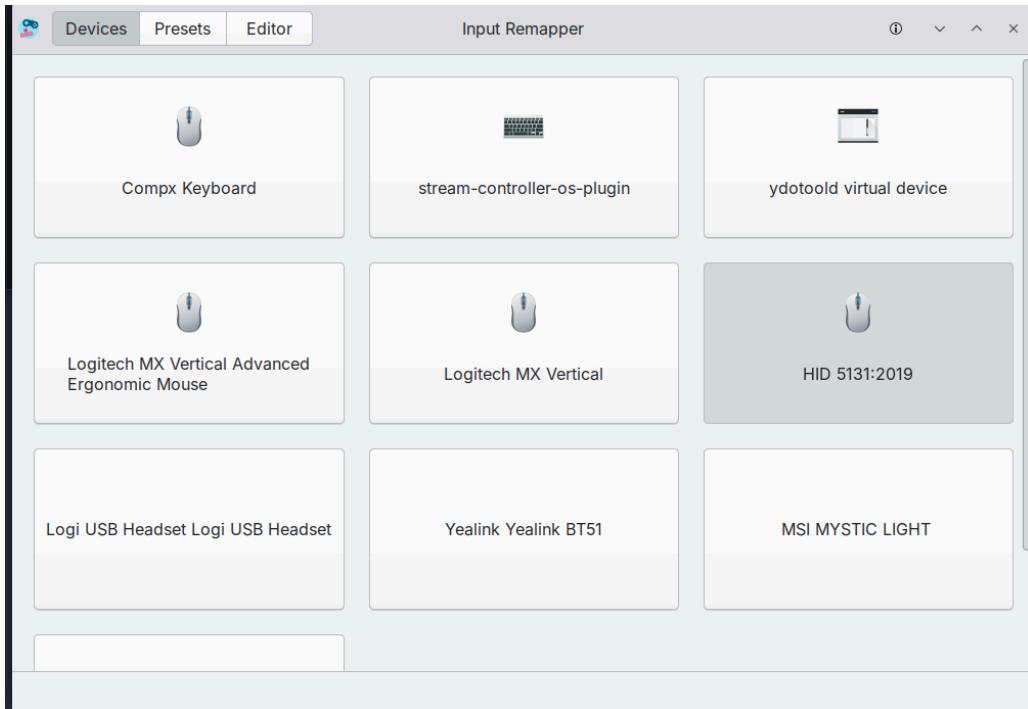


Figure 1: Input Remapper device selection

12.8 Step 2: Select the Specific Interface

Some devices appear multiple times (keyboard interface, mouse interface, etc.). Select the one that matches your input type.

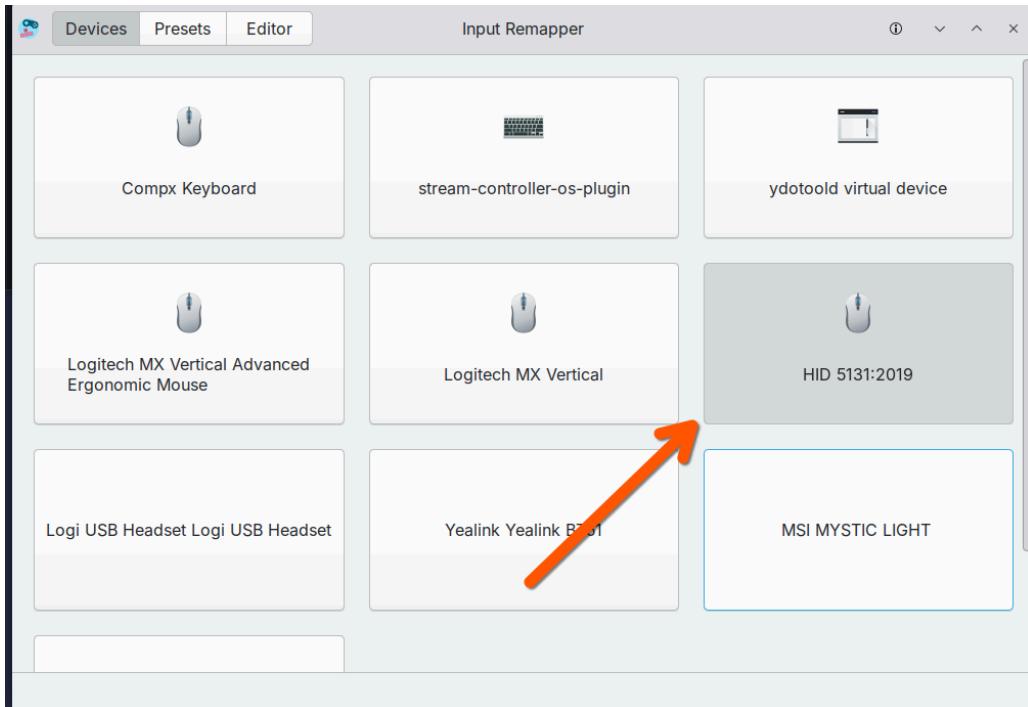


Figure 2: Selecting the HID device interface

12.9 Step 3: Open the Editor

Switch to the **Editor** tab to create key mappings.

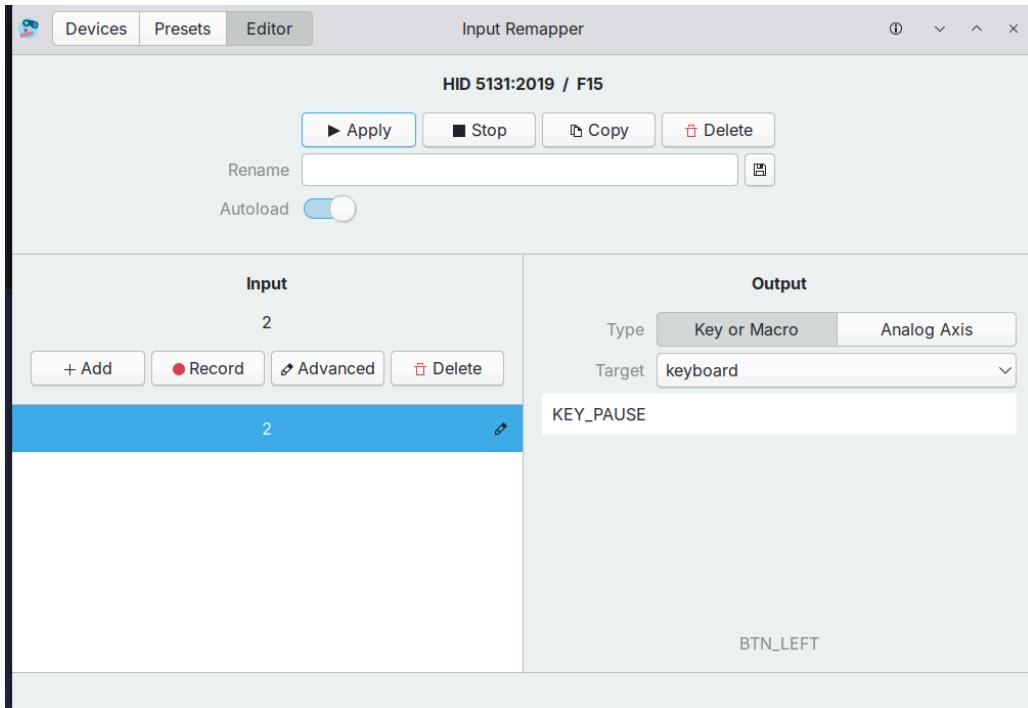


Figure 3: Editor tab for creating mappings

12.10 Step 4: Record the Input

Click **Record** and press the button you want to remap. Input Remapper will capture it.

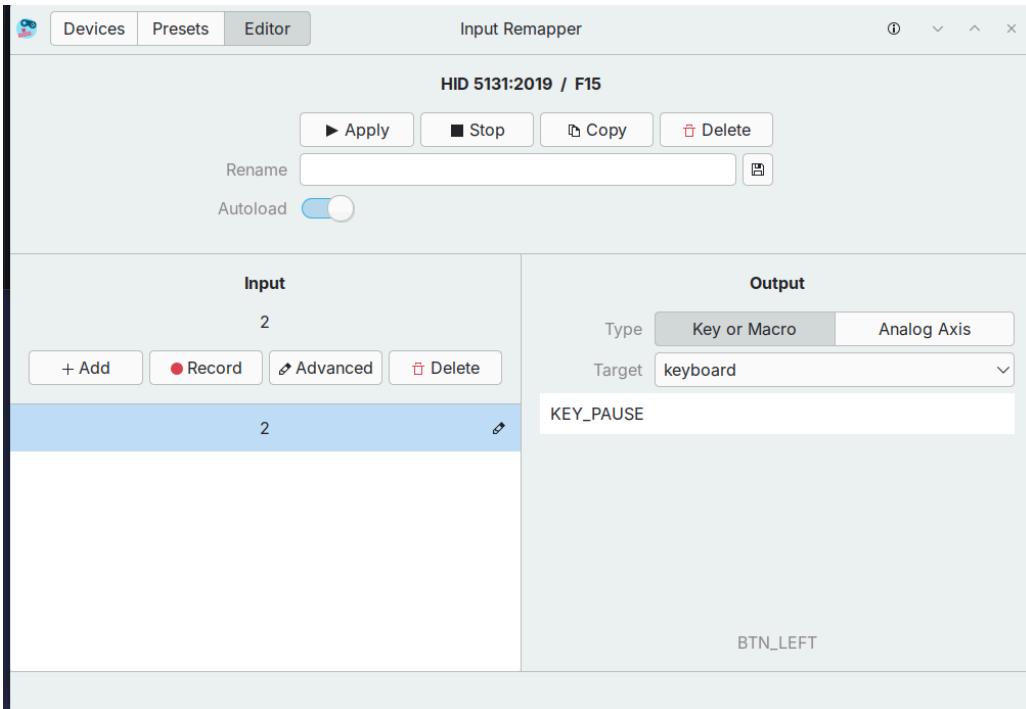


Figure 4: Recording the input button

12.11 Step 5: Choose the Output Key

Type the key name in the output field. For Voice Notepad, use KEY_F15 through KEY_F20.

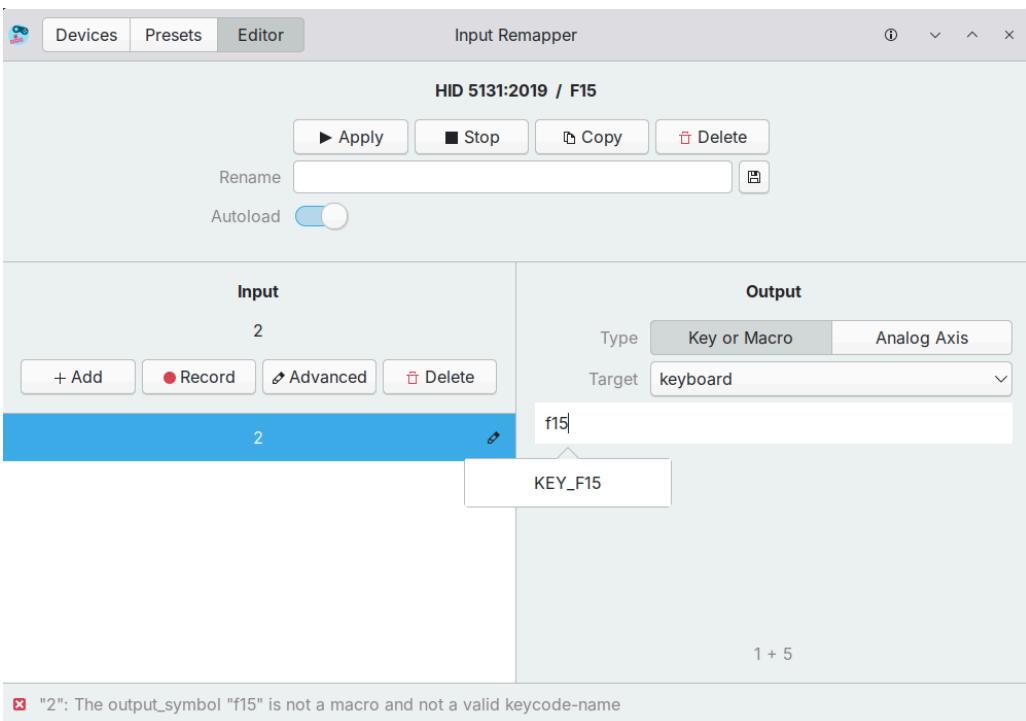


Figure 5: Selecting KEY_F15 as the output

12.12 Step 6: Apply and Enable Autoload

-
1. Click **Apply** to activate the mapping
 2. Enable **Autoload** to persist across reboots
 3. Enable the service: `systemctl --user enable input-remapper`

12.13 Recommended Multi-Button Setup

Button	Maps To	Function
1	KEY_F15	Toggle (start/stop and transcribe)
2	KEY_F17	Transcribe cached audio
3	KEY_F18	Clear recording

Appendix B

Bill of Materials

This appendix lists all major components and dependencies integrated into Voice Notepad.

12.14 Core Technologies

Component	License	Purpose
PyQt6	GPL v3	Desktop GUI framework with tabbed interface, system tray, and keyboard shortcuts
Google Gemini API	Proprietary	Multimodal AI for single-pass audio transcription and cleanup
OpenRouter API	Proprietary	Alternative API gateway with per-key cost tracking

12.15 Audio Processing

Component	License	Purpose
PyAudio	MIT	Audio recording with microphone device selection
TEN VAD	Apache 2.0	Voice Activity Detection for silence removal (306KB native library)
pydub	MIT	Audio format conversion and manipulation
FFmpeg	LGPL/GPL	Audio compression, format conversion, and resampling

12.16 Text-to-Speech

Component	License	Purpose
edge-tts	GPL v3	Microsoft Edge TTS for voice announcements
en-GB-RyanNeural	Microsoft	British English male voice for audio feedback

12.17 Data Storage

Component	License	Purpose
Mongita	BSD 3-Clause	MongoDB-compatible pure Python database for transcript storage
Opus codec	BSD 3-Clause	Audio archival format (24kbps for speech)

12.18 Input Handling

Component	License	Purpose
pynput	LGPL v3	Keyboard input handling (fallback for non-Linux)
evdev	BSD 3-Clause	Linux input device access for global hotkeys on Wayland
ydotool	MIT	Text injection on Wayland via virtual keyboard

12.19 Visualization

Component	License	Purpose
pyqtgraph	MIT	Charts and visualizations for analytics tab
Markdown	BSD 3-Clause	Markdown rendering in transcript display

12.20 System Dependencies

These must be installed separately on the host system:

Package	Purpose
python3	Python 3.10+ runtime
ffmpeg	Audio format conversion and compression
portaudio19-dev	PyAudio recording library headers
libc++1	C++ standard library required by TEN VAD
ydotool	Text injection daemon (optional, for Wayland)

12.21 Component Links

- **TEN VAD:** <https://github.com/TEN-framework/ten-vad>
- **Mongita:** <https://github.com/scottrogowski/mongita>
- **edge-tts:** <https://github.com/rany2/edge-tts>
- **Input Remapper:** <https://github.com/sezanzeb/input-remapper>
- **ydotool:** <https://github.com/ReimuNotMoe/ydotool>