response_format string

Optional Defaults to mp3
The format to audio in.
Supported formats are mp3,
opus, aac, flac, wav,
and pcm.

speed number Optional
Defaults to 1
The speed of the generated
audio. Select a value from
0.25 to 4.0 . 1.0 is the
default.

stream_format string
Optional Defaults to audio
The format to stream the audio
in. Supported formats are sse
and audio sse is not
supported for tts-1 or
tts-1-hd.

Returns

The audio file content or a stream of audio events.

Create transcription

POST https://api.openai. com/v1/audio/transc Default Streaming Logprobs Word timestamps

riptions

Transcribes audio into the input language.

Request body

file file Required

The audio file object (not file name) to transcribe, in one of these formats: flac, mp3, mp4, mpeg, mpga, m4a, ogg, wav, or webm.

model string Required
ID of the model to use. The options are

gpt-4o-transcribe,

gpt-4o-mini-transcribe,

and whisper-1 (which is powered by our open source

Whisper V2 model).

chunking_strategy

"auto" or object Optional
Controls how the audio is cut into chunks. When set to "auto", the server first normalizes loudness and then uses voice activity detection (VAD) to choose boundaries. server_vad object can be provided to tweak VAD detection parameters manually. If unset, the audio is transcribed as a single block.

✓ Show possible types

```
import OpenAI from "openai";

const openai = new OpenAI();

async function main() {
  const transcription = await openai.audio
  file: fs.createReadStream("audio.mp3")
  model: "gpt-4o-transcribe",
  response_format: "json",
  include: ["logprobs"]
});

console.log(transcription);

main();
```

```
ත
Response
   {
     "text": "Hey, my knee is hurting and I w
     "logprobs": [
       { "token": "Hey", "logprob": -1.041529
       { "token": ",", "logprob": -9.805982e-
       { "token": " my", "logprob": -0.002297
       {
         "token": " knee",
         "logprob": -4.7159858e-5,
         "bytes": [32, 107, 110, 101, 101]
       },
       { "token": " is", "logprob": -0.043909
         "token": " hurting",
         III aanzahii.
```

Additional information to include in the transcription response.

logprobs will return the log probabilities of the tokens in the response to understand the model's confidence in the transcription. logprobs only works with response_format set to json and only with the

include[] array Optional

gpt-4o-mini-transcribe.

models gpt-4o-transcribe

and

language string Optional
The language of the input audio.
Supplying the input language in
ISO-639-1 (e.g. en) format will
improve accuracy and latency.

prompt string Optional
An optional text to guide the model's style or continue a previous audio segment. The prompt should match the audio language.

response_format string
Optional Defaults to json
The format of the output, in one
of these options: json , text ,
srt , verbose_json ,or
vtt .For
gpt-4o-transcribe and
gpt-4o-mini-transcribe ,
the only supported format is
json .

Stream boolean or null Optional
Defaults to false
If set to true, the model response
data will be streamed to the client
as it is generated using
server-sent events. See the
Streaming section of the Speechto-Text guide
for more information.

Note: Streaming is not supported for the whisper-1 model and will be ignored.

temperature number Optional Defaults to 0

The sampling temperature, between 0 and 1. Higher values like 0.8 will make the output more random, while lower values like 0.2 will make it more focused and deterministic. If set to 0, the model will use log probability to automatically increase the temperature until certain thresholds are hit.

timestamp_granularities[]

array Optional
Defaults to segment
The timestamp granularities to
populate for this transcription.
response_format must be set
verbose_json to use
timestamp granularities. Either or
both of these options are
supported: word , or segment
. Note: There is no additional

latency for segment timestamps, but generating word timestamps incurs additional latency.

Returns

The <u>transcription object</u>, a <u>verbose transcription object</u> or a <u>stream of transcript events</u>.

Create translation

POST https://api.openai. com/v1/audio/transl ations

Translates audio into English.

Request body

file file Required

The audio file object (not file name) translate, in one of these formats: flac, mp3, mp4, mpeg, mpga, m4a, ogg, wav, or webm.

model string or "whisper-1"

Required

ID of the model to use. Only whisper-1 (which is powered by our open source Whisper V2

```
import fs from "fs";
import OpenAI from "openai";

const openai = new OpenAI();

async function main() {
   const translation = await openai.audio
     file: fs.createReadStream("speech.
     model: "whisper-1",
});

console.log(translation.text);

main();
```

```
Response 

1 {
2 "text": "Hello, my name is Wolfgang and I
```

3 }

model) is currently available.

prompt string Optional
An optional text to guide the model's style or continue a previous audio segment. The <u>prompt</u> should be in English.

response_format string
Optional Defaults to json
The format of the output, in one
of these options: json , text
, srt , verbose_json , or
vtt .

temperature number Optional Defaults to 0

The sampling temperature, between 0 and 1. Higher values like 0.8 will make the output more random, while lower values like 0.2 will make it more focused and deterministic. If set to 0, the model will use <u>log probability</u> to automatically increase the temperature until certain thresholds are hit.

Returns

The translated text.

The transcription object (JSON)

Represents a transcription response returned by model, based on the provided input.

logprobs array

The log probabilities of the tokens in the transcription. Only returned with the models gpt-4o-transcribe and gpt-4o-mini-transcribe if logprobs is added to the include array.

∨ Show properties

text string

The transcribed text.

usage object

Token usage statistics for the request.

✓ Show possible types

```
OBJECT The transcription object (JSON)

1 {
2  "text": "Imagine the wildest idea that y
3  "usage": {
4   "type": "tokens",
5   "input_tokens": 14,
6   "input_token_details": {
7    "text_tokens": 10,
8    "audio_tokens": 4
9   },
10   "output_tokens": 101,
11   "total_tokens": 115
12  }
13 }
```

The transcription object (Verbose JSON)

Represents a verbose json transcription response returned by model, based on the provided input.

duration number

The duration of the input audio.

language string

The language of the input audio.

segments array

Segments of the transcribed text and their corresponding details.

✓ Show properties

text string

The transcribed text.

usage object

Usage statistics for models billed by audio input duration.

∨ Show properties

words array

Extracted words and their corresponding timestamps.

✓ Show properties

```
OBJECT The transcription object (Verbose JS...
                                            ഹ
   {
     "task": "transcribe",
     "language": "english",
      "duration": 8.470000267028809,
     "text": "The beach was a popular spot on
     "segments": [
       {
          "id": 0,
          "seek": 0,
          "start": 0.0,
          "end": 3.319999933242798,
          "text": " The beach was a popular sp
          "tokens": [
            50364, 440, 7534, 390, 257, 3743,
          ],
          "temperature": 0.0,
          "avg_logprob": -0.2860786020755768,
          "compression_ratio": 1.2363636493682
          "no_speech_prob": 0.0098597947508096
       },
     ],
     "usage": {
       "type": "duration",
       "seconds": 9
     }
27 }
```

Stream Event (speech.audio.delta)

Emitted for each chunk of audio data generated during speech synthesis.

audio string

A chunk of Base64-encoded audio data.

type string

The type of the event. Always speech.audio.delta.

```
OBJECT Stream Event (speech.audio.delta)

1 {
2  "type": "speech.audio.delta",
3  "audio": "base64-encoded-audio-data"
4 }
```

Stream Event (speech.audio.done)

Emitted when the speech synthesis is complete and all audio has been streamed.

type string

The type of the event. Always speech.audio.done.

usage object

Token usage statistics for the request.

✓ Show properties

```
OBJECT Stream Event (speech.audio.done)

1 {
2  "type": "speech.audio.done",
3  "usage": {
4    "input_tokens": 14,
5    "output_tokens": 101,
6    "total_tokens": 115
7  }
8 }
```

Stream Event (transcript.text.delta)

Emitted when there is an additional text delta. This is also the first event emitted when the transcription starts. Only emitted when you create a transcription with the Stream parameter set to true.

delta string

The text delta that was additionally transcribed.

logprobs array

The log probabilities of the delta.

Only included if you

create a transcription with the

include[] parameter set to

logprobs.

✓ Show properties

type string

The type of the event. Always transcript.text.delta.

```
OBJECT Stream Event (transcript.text.delta)

1 {
2  "type": "transcript.text.delta",
3  "delta": " wonderful"
4 }
```

Stream Event (transcript.text.done)

Emitted when the transcription is complete. Contains the complete transcription text. Only emitted when you create a transcription with the Stream parameter set to true.

logprobs array

The log probabilities of the individual tokens in the transcription. Only included if you <u>create a transcription</u> with the <u>include[]</u> parameter set to <u>logprobs</u>.

✓ Show properties

text string

The text that was transcribed.

type string

The type of the event. Always transcript.text.done.

usage object

Usage statistics for models billed by token usage.

✓ Show properties

```
OBJECT Stream Event (transcript.text.done)
                                            എ
   {
     "type": "transcript.text.done",
     "text": "I see skies of blue and clouds
     "usage": {
       "type": "tokens",
       "input_tokens": 14,
       "input_token_details": {
          "text_tokens": 10,
          "audio_tokens": 4
       },
       "output_tokens": 31,
       "total_tokens": 45
     }
14 }
```

Images