

Audion Media Tools - Wrappers of Scripts

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Made from the data of the program by `media_cli.cmd map --docs`: the manifest of the operations (`config\tool_manifest.yaml`), the layout of the window (`config>window_layout.yaml`) and the table of the wrappers (`config>wrappers.yaml`). It is not edited by hand: a text is changed there and the command is run again.

Every wrapper is a door of `media_cli.cmd wrapper NAME`: a file of its own name that runs one or more operations of the program with the values written for it. The environment variables of the table at the end are read by the wrappers as they always were.

Wrapper	What it is	Takes	The same through the console command
<code>audio-downmix-51-to-20</code>	5.1 down to stereo, WAV 24-bit	-	<code>media_cli.cmd run --operation audio_run --set audio_workflow=extract --set audio_format=wav --set audio_bit_depth=24 --set audio_sample_rate=source --set audio_channel_mode=stereo --set downmix_formula=surround_6db</code> <code>media_cli.cmd run --operation audio_run --set audio_workflow=audio_files --set audio_format=wav --set audio_bit_depth=24 --set audio_sample_rate=source --set audio_channel_mode=stereo --set downmix_formula=surround_6db</code>

Wrapper	What it is	Takes	The same through the console command
<code>audio-extract-audio-copy</code>	The first sound stream as it is, in a container of its codec	-	<code>media_cli.cmd run --operation audio_run --set audio_workflow=extract --set audio_format=copy --set audio_copy_container=auto</code>
<code>audio-extract-audio-wav</code>	The sound to WAV 24-bit at its own rate	-	<code>media_cli.cmd run --operation audio_run --set audio_workflow=extract --set audio_format=wav --set audio_bit_depth=24 --set audio_sample_rate=source</code> <code>media_cli.cmd run --operation audio_run --set audio_workflow=audio_files --set audio_format=wav --set audio_bit_depth=24 --set audio_sample_rate=source</code>
<code>audio-extract-audio-wav-32</code>	The sound to WAV 32-bit float at its own rate	-	<code>media_cli.cmd run --operation audio_run --set audio_workflow=extract --set audio_format=wav --set audio_bit_depth=32f --set audio_sample_rate=source</code> <code>media_cli.cmd run --operation audio_run --set audio_workflow=audio_files --set audio_format=wav --set audio_bit_depth=32f --set audio_sample_rate=source</code>
<code>audio-lufs-14</code>	Loudness -14 LUFS (streaming), WAV 24-bit	-	<code>media_cli.cmd run --operation audio_run --set audio_workflow=extract --set audio_format=wav --set audio_bit_depth=24 --set audio_sample_rate=source --set audio_lufs=-14 --set audio_lufs_mode=one_pass</code> <code>media_cli.cmd run --operation audio_run --set audio_workflow=audio_files --set audio_format=wav --set audio_bit_depth=24 --set audio_sample_rate=source --set audio_lufs=-14 --set audio_lufs_mode=one_pass</code>

Wrapper	What it is	Takes	The same through the console command
<code>audio-lufs-23</code>	Loudness -23 LUFS (EBU R128), WAV 24-bit	-	<pre>media_cli.cmd run --operation audio_run --set audio_workflow=extract --set audio_format=wav -- set audio_bit_depth=24 --set audio_sample_rate=source --set audio_lufs=-23 -- set audio_lufs_mode=one_pass</pre> <pre>media_cli.cmd run --operation audio_run --set audio_workflow=audio_files --set audio_format=wav --set audio_bit_depth=24 --set audio_sample_rate=source --set audio_lufs=-23 -- set audio_lufs_mode=one_pass</pre>
<code>audio-resample-to-44100</code>	The sound to 44.1 kHz, WAV 24-bit	-	<pre>media_cli.cmd run --operation audio_run --set audio_workflow=extract --set audio_format=wav -- set audio_bit_depth=24 --set audio_sample_rate=44100</pre> <pre>media_cli.cmd run --operation audio_run --set audio_workflow=audio_files --set audio_format=wav --set audio_bit_depth=24 --set audio_sample_rate=44100</pre>
<code>audio-resample-to-44100-32f</code>	The sound to 44.1 kHz, WAV 32-bit float	-	<pre>media_cli.cmd run --operation audio_run --set audio_workflow=extract --set audio_format=wav -- set audio_bit_depth=32f --set audio_sample_rate=44100</pre> <pre>media_cli.cmd run --operation audio_run --set audio_workflow=audio_files --set audio_format=wav --set audio_bit_depth=32f --set audio_sample_rate=44100</pre>
<code>audio-resample-to-44100-32f_sync</code>	The sound to 44.1 kHz, WAV 32-bit float, timestamps evened	-	<pre>media_cli.cmd run --operation audio_run --set audio_workflow=extract --set audio_format=wav -- set audio_bit_depth=32f --set audio_sample_rate=44100 --set audio_resync=true</pre> <pre>media_cli.cmd run --operation audio_run --set audio_workflow=audio_files --set audio_format=wav --set audio_bit_depth=32f --set audio_sample_rate=44100 --set audio_resync=true</pre>

Wrapper	What it is	Takes	The same through the console command
<code>audio-resample-to-44100_sync</code>	The sound to 44.1 kHz, WAV 24-bit, timestamps evened	-	<pre>media_cli.cmd run --operation audio_run --set audio_workflow=extract --set audio_format=wav -- set audio_bit_depth=24 --set audio_sample_rate=44100 --set audio_resync=true</pre> <pre>media_cli.cmd run --operation audio_run --set audio_workflow=audio_files --set audio_format=wav --set audio_bit_depth=24 --set audio_sample_rate=44100 --set audio_resync=true</pre>
<code>audio-resample-to-48000</code>	The sound to 48 kHz, WAV 24-bit	-	<pre>media_cli.cmd run --operation audio_run --set audio_workflow=extract --set audio_format=wav -- set audio_bit_depth=24 --set audio_sample_rate=48000</pre> <pre>media_cli.cmd run --operation audio_run --set audio_workflow=audio_files --set audio_format=wav --set audio_bit_depth=24 --set audio_sample_rate=48000</pre>
<code>audio-resample-to-48000-32f</code>	The sound to 48 kHz, WAV 32-bit float	-	<pre>media_cli.cmd run --operation audio_run --set audio_workflow=extract --set audio_format=wav -- set audio_bit_depth=32f --set audio_sample_rate=48000</pre> <pre>media_cli.cmd run --operation audio_run --set audio_workflow=audio_files --set audio_format=wav --set audio_bit_depth=32f --set audio_sample_rate=48000</pre>
<code>audio-resample-to-48000-32f_sync</code>	The sound to 48 kHz, WAV 32-bit float, timestamps evened	-	<pre>media_cli.cmd run --operation audio_run --set audio_workflow=extract --set audio_format=wav -- set audio_bit_depth=32f --set audio_sample_rate=48000 --set audio_resync=true</pre> <pre>media_cli.cmd run --operation audio_run --set audio_workflow=audio_files --set audio_format=wav --set audio_bit_depth=32f --set audio_sample_rate=48000 --set audio_resync=true</pre>

Wrapper	What it is	Takes	The same through the console command
<code>audio-resample-to-48000_sync</code>	The sound to 48 kHz, WAV 24-bit, timestamps evened	-	<pre>media_cli.cmd run --operation audio_run --set audio_workflow=extract --set audio_format=wav -- set audio_bit_depth=24 --set audio_sample_rate=48000 --set audio_resync=true</pre> <pre>media_cli.cmd run --operation audio_run --set audio_workflow=audio_files --set audio_format=wav --set audio_bit_depth=24 --set audio_sample_rate=48000 --set audio_resync=true</pre>
<code>audio-to-flac</code>	The sound to FLAC at its own rate	-	<pre>media_cli.cmd run --operation audio_run --set audio_workflow=extract --set audio_format=flac -- set audio_bit_depth=source --set audio_sample_rate=source</pre> <pre>media_cli.cmd run --operation audio_run --set audio_workflow=audio_files --set audio_format=flac --set audio_bit_depth=source -- set audio_sample_rate=source</pre>
<code>audio-to-flac-48k</code>	The sound to FLAC 48 kHz	-	<pre>media_cli.cmd run --operation audio_run --set audio_workflow=extract --set audio_format=flac -- set audio_bit_depth=source --set audio_sample_rate=48000</pre> <pre>media_cli.cmd run --operation audio_run --set audio_workflow=audio_files --set audio_format=flac --set audio_bit_depth=source -- set audio_sample_rate=48000</pre>
<code>audio-to-flac-96k</code>	The sound to FLAC 96 kHz	-	<pre>media_cli.cmd run --operation audio_run --set audio_workflow=extract --set audio_format=flac -- set audio_bit_depth=source --set audio_sample_rate=96000</pre> <pre>media_cli.cmd run --operation audio_run --set audio_workflow=audio_files --set audio_format=flac --set audio_bit_depth=source -- set audio_sample_rate=96000</pre>

Wrapper	What it is	Takes	The same through the console command
<code>audio-to-m4a-256</code>	The sound to AAC 256 kbps in M4A	-	<pre>media_cli.cmd run --operation audio_run --set audio_workflow=extract --set audio_profile=m4a256</pre> <pre>media_cli.cmd run --operation audio_run --set audio_workflow=audio_files --set audio_profile=m4a256</pre>
<code>audio-to-m4a-384</code>	The sound to AAC 384 kbps in M4A	-	<pre>media_cli.cmd run --operation audio_run --set audio_workflow=extract --set audio_profile=m4a384</pre> <pre>media_cli.cmd run --operation audio_run --set audio_workflow=audio_files --set audio_profile=m4a384</pre>
<code>ff-100to23976-conform</code>	100 fps to 23.976, conform	-	<pre>media_cli.cmd run --operation fps_run --set fps_mode=conform --set fps_target=24000/1001</pre>
<code>ff-100to23976-varispeed</code>	100 fps to 23.976, varispeed	-	<pre>media_cli.cmd run --operation fps_run --set fps_mode=varispeed --set fps_target=24000/1001</pre>
<code>ff-11988to23976-conform</code>	119.88 fps to 23.976, conform	-	<pre>media_cli.cmd run --operation fps_run --set fps_mode=conform --set fps_target=24000/1001</pre>
<code>ff-11988to23976-varispeed</code>	119.88 fps to 23.976, varispeed	-	<pre>media_cli.cmd run --operation fps_run --set fps_mode=varispeed --set fps_target=24000/1001</pre>
<code>ff-17982to23976-conform</code>	179.82 fps to 23.976, conform	-	<pre>media_cli.cmd run --operation fps_run --set fps_mode=conform --set fps_target=24000/1001</pre>
<code>ff-17982to23976-varispeed</code>	179.82 fps to 23.976, varispeed	-	<pre>media_cli.cmd run --operation fps_run --set fps_mode=varispeed --set fps_target=24000/1001</pre>
<code>ff-24to23976-varispeed</code>	24 fps to 23.976, varispeed	-	<pre>media_cli.cmd run --operation fps_run --set fps_mode=varispeed --set fps_target=24000/1001</pre>

Wrapper	What it is	Takes	The same through the console command
<code>ff-25to23976-conform</code>	25 fps to 23.976, conform	-	<code>media_cli.cmd run --operation fps_run --set fps_mode=conform --set fps_target=24000/1001</code>
<code>ff-25to23976-varispeed</code>	25 fps to 23.976, varispeed	-	<code>media_cli.cmd run --operation fps_run --set fps_mode=varispeed --set fps_target=24000/1001</code>
<code>ff-2997to23976-conform</code>	29.97 fps to 23.976, conform	-	<code>media_cli.cmd run --operation fps_run --set fps_mode=conform --set fps_target=24000/1001</code>
<code>ff-2997to23976-varispeed</code>	29.97 fps to 23.976, varispeed	-	<code>media_cli.cmd run --operation fps_run --set fps_mode=varispeed --set fps_target=24000/1001</code>
<code>ff-50to23976-conform</code>	50 fps to 23.976, conform	-	<code>media_cli.cmd run --operation fps_run --set fps_mode=conform --set fps_target=24000/1001</code>
<code>ff-50to23976-varispeed</code>	50 fps to 23.976, varispeed	-	<code>media_cli.cmd run --operation fps_run --set fps_mode=varispeed --set fps_target=24000/1001</code>
<code>ff-5994to23976-conform</code>	59.94 fps to 23.976, conform	-	<code>media_cli.cmd run --operation fps_run --set fps_mode=conform --set fps_target=24000/1001</code>
<code>ff-5994to23976-varispeed</code>	59.94 fps to 23.976, varispeed	-	<code>media_cli.cmd run --operation fps_run --set fps_mode=varispeed --set fps_target=24000/1001</code>
<code>ff-av1-crf14</code>	AV1 (SVT), CRF 14, MKV	-	<code>media_cli.cmd run --operation storage_run --set storage_profile=svt_av1_crf14 --set audio_mode=aac_320</code>
<code>ff-av1-fast-crf14</code>	AV1 (SVT), fast preset, CRF 14, MP4	-	<code>media_cli.cmd run --operation storage_run --set storage_profile=svt_av1_crf14 --set av1_preset=8 --set output_container=mp4 --set audio_mode=aac_320</code>

Wrapper	What it is	Takes	The same through the console command
<code>ff-dnxhr-hq-mov</code>	DNxHR HQ in MOV, PCM 24-bit	-	<code>media_cli.cmd run --operation editing_run --set editing_profile=dnxhr_hq_mov</code>
<code>ff-dnxhr-hq-mxf</code>	DNxHR HQ in MXF, PCM 24-bit	-	<code>media_cli.cmd run --operation editing_run --set editing_profile=dnxhr_hq_mxf</code>
<code>ff-dnxhr-hqx-mov</code>	DNxHR HQX 10-bit in MOV, PCM 24-bit	-	<code>media_cli.cmd run --operation editing_run --set editing_profile=dnxhr_hqx_mov</code>
<code>ff-dnxhr-hqx-mxf</code>	DNxHR HQX 10-bit in MXF, PCM 24-bit	-	<code>media_cli.cmd run --operation editing_run --set editing_profile=dnxhr_hqx_mxf</code>
<code>ff-extract-audio</code>	The first sound stream as it is, in a container of its codec	-	<code>media_cli.cmd run --operation audio_run --set audio_workflow=extract --set audio_format=copy --set audio_copy_container=auto</code>
<code>ff-extract-subs</code>	The subtitle streams as they are, to MKS	-	<code>media_cli.cmd run --operation subtitles_extract</code>
<code>ff-grade-lut-only-hevc10</code>	The LUT, HEVC Main10	-	<code>media_cli.cmd run --operation grading_run --set grading_profile=lut_hevc10 --set audio_mode=aac_320 --set cpu_encoder_preset=medium</code>
<code>ff-grade-lut-only-prores422</code>	The LUT, ProRes 422	-	<code>media_cli.cmd run --operation grading_run --set grading_profile=lut_prores422 --set audio_mode=aac_320</code>
<code>ff-grade-lut-only-proresLT</code>	The LUT, ProRes LT	-	<code>media_cli.cmd run --operation grading_run --set grading_profile=lut_prores422 --set audio_mode=aac_320 --set target_codecs=prores_lt</code>

Wrapper	What it is	Takes	The same through the console command
<code>ff-grade-lut-only-x264</code>	The LUT, H.264	-	<code>media_cli.cmd run --operation grading_run --set grading_profile=lut_x264 --set audio_mode=aac_320 --set cpu_encoder_preset=medium</code>
<code>ff-grade-preexpose-lut-x264</code>	A gamma of 1.15 before the LUT, H.264	-	<code>media_cli.cmd run --operation grading_run --set grading_profile=preexpose_x264 --set audio_mode=aac_320 --set cpu_encoder_preset=medium</code>
<code>ff-grade-pregamma-lut-hevc10</code>	A gamma of 0.85 before the LUT, HEVC Main10	-	<code>media_cli.cmd run --operation grading_run --set grading_profile=pregamma_hevc10 --set audio_mode=aac_320 --set cpu_encoder_preset=medium</code>
<code>ff-grade-pregamma-lut-prores422</code>	A gamma of 0.85 before the LUT, ProRes 422	-	<code>media_cli.cmd run --operation grading_run --set script_preset=ff-grade-pregamma-lut-prores422.cmd --set output_container=mov --set pix_fmt=yuv422p10le --set audio_mode=aac_320</code>
<code>ff-grade-pregamma-lut-x264</code>	A gamma of 0.85 before the LUT, H.264	-	<code>media_cli.cmd run --operation grading_run --set script_preset=ff-grade-pregamma-lut-x264.cmd --set output_container=mp4 --set pix_fmt=yuv420p --set audio_mode=aac_320 --set cpu_encoder_preset=medium</code>
<code>ff-hdr2sdr-hable-hevc10</code>	HDR to SDR (Hable), HEVC Main10	-	<code>media_cli.cmd run --operation grading_run --set grading_profile=hdr_hable_x264 --set target_codecs=hevc_x265 --set pix_fmt=yuv420p10le --set audio_mode=aac_320 --set cpu_encoder_preset=medium</code>
<code>ff-hdr2sdr-hable-x264</code>	HDR to SDR (Hable), H.264	-	<code>media_cli.cmd run --operation grading_run --set grading_profile=hdr_hable_x264 --set audio_mode=aac_320 --set cpu_encoder_preset=medium</code>
<code>ff-hevc10-crf14</code>	HEVC Main10, CRF 14	-	<code>media_cli.cmd run --operation storage_run --set storage_profile=hevc10_crf14 --set audio_mode=aac_320 --set cpu_encoder_preset=medium</code>

Wrapper	What it is	Takes	The same through the console command
<code>ff-prores-422-mov</code>	ProRes 422 in MOV, PCM 24-bit	-	<code>media_cli.cmd run --operation editing_run --set editing_profile=prores_422_mov</code>
<code>ff-prores-422-mxf</code>	ProRes 422 in MXF, PCM 24-bit	-	<code>media_cli.cmd run --operation editing_run --set editing_profile=prores_422_mxf</code>
<code>ff-prores-lt-mov</code>	ProRes LT in MOV, PCM 24-bit	-	<code>media_cli.cmd run --operation editing_run --set editing_profile=prores_lt_mov</code>
<code>ff-prores-lt-mxf</code>	ProRes LT in MXF, PCM 24-bit	-	<code>media_cli.cmd run --operation editing_run --set editing_profile=prores_lt_mxf</code>
<code>ff-prores-proxy-mov</code>	ProRes Proxy in MOV, PCM 24-bit	-	<code>media_cli.cmd run --operation editing_run --set editing_profile=prores_proxy_mov --set audio_mode=pcm_s24</code>
<code>ff-prores-proxy-mxf</code>	ProRes Proxy in MXF, PCM 24-bit	-	<code>media_cli.cmd run --operation editing_run --set editing_profile=prores_proxy_mxf --set audio_mode=pcm_s24</code>
<code>ff-remux-mkv-to-mp4</code>	MKV to MP4 without re-encoding	-	<code>media_cli.cmd run --operation remux_run --set remux_stream_mode=video_audio --set remux_target_format=mp4 --only mkv</code>
<code>ff-remux-mkv-to-mp4-faststart</code>	MKV to MP4 without re-encoding, for the web	-	<code>media_cli.cmd run --operation remux_run --set remux_stream_mode=video_audio --set remux_target_format=mp4 --only mkv</code>
<code>ff-remux-mov-to-mxf-copy</code>	MOV to MXF without re-encoding	-	<code>media_cli.cmd run --operation remux_run --set remux_stream_mode=video_audio --set remux_target_format=mxf --only mov</code>
<code>ff-remux-mp4-to-mkv</code>	MP4 to MKV without re-encoding	-	<code>media_cli.cmd run --operation remux_run --set remux_stream_mode=video_audio --set remux_target_format=mkv --only mp4</code>

Wrapper	What it is	Takes	The same through the console command
<code>ff-remux-mp4-to-mkv-copy</code>	MP4 to MKV without re-encoding	-	<code>media_cli.cmd run --operation remux_run --set remux_stream_mode=video_audio --set remux_target_format=mkv --only mp4</code>
<code>ff-remux-mxf-to-mov-copy</code>	MXF to MOV without re-encoding	-	<code>media_cli.cmd run --operation remux_run --set remux_stream_mode=video_audio --set remux_target_format=mov --only mxf</code>
<code>ff-selftest</code>	Versions of FFmpeg, ffprobe, yt-dlp and deno	-	<code>media_cli.cmd run --operation selftest</code>
<code>ff-trim-both</code>	Keep what lies between IN and OUT	-	<code>media_cli.cmd run --operation trim_run --set trim_action=keep</code>
<code>ff-trim-end</code>	Cut the tail off: keep up to OUT	-	<code>media_cli.cmd run --operation trim_run --set trim_action=keep --set "trim_start="</code>
<code>ff-trim-middle</code>	Cut out what lies between IN and OUT	-	<code>media_cli.cmd run --operation trim_run --set trim_action=cut</code>
<code>ff-trim-start</code>	Cut the head off: keep from IN	-	<code>media_cli.cmd run --operation trim_run --set trim_action=keep --set "trim_end="</code>
<code>ff-x264-4k-to-1080-crf14</code>	H.264 1080p from 4K, CRF 14	-	<code>media_cli.cmd run --operation storage_run --set storage_profile=x264_1080_crf14 --set audio_mode=aac_320 --set cpu_encoder_preset=medium</code>
<code>ff-x264-4k-to-1080-crf17</code>	H.264 1080p from 4K, CRF 17	-	<code>media_cli.cmd run --operation storage_run --set storage_profile=x264_1080_crf14 --set crf=17 --set audio_mode=aac_320 --set cpu_encoder_preset=medium</code>

Wrapper	What it is	Takes	The same through the console command
<code>ff-x264-4k-to-1080-crf21</code>	H.264 1080p from 4K, CRF 21	-	<code>media_cli.cmd run --operation storage_run --set storage_profile=x264_1080_crf14 --set crf=21 --set audio_mode=aac_320 --set cpu_encoder_preset=medium</code>
<code>ff-x264-crf14</code>	H.264, CRF 14	-	<code>media_cli.cmd run --operation storage_run --set storage_profile=x264_crf14 --set audio_mode=aac_320 --set cpu_encoder_preset=medium</code>
<code>ff-x264-crf17</code>	H.264, CRF 17	-	<code>media_cli.cmd run --operation storage_run --set storage_profile=x264_crf14 --set crf=17 --set audio_mode=aac_320 --set cpu_encoder_preset=medium</code>
<code>ff-x264-crf21</code>	H.264, CRF 21	-	<code>media_cli.cmd run --operation storage_run --set storage_profile=x264_crf14 --set crf=21 --set audio_mode=aac_320 --set cpu_encoder_preset=medium</code>
<code>yt-1080p-avc-mp4</code>	1080p H.264 with AAC in MP4	urls	<code>media_cli.cmd run --operation youtube_run --set youtube_profile=upload_ready --set youtube_resolution=1080 --set youtube_video_codec=avc --set youtube_container=mp4 --set "youtube_options="</code>
<code>yt-480p-aac</code>	480p with AAC in MP4	urls	<code>media_cli.cmd run --operation youtube_run --set youtube_profile=video_custom --set youtube_resolution=480 --set youtube_container=mp4 --set "youtube_options="</code>
<code>yt-batch-480p-aac</code>	480p from the links of input\urls.txt	-	<code>media_cli.cmd run --operation youtube_run --set youtube_source_mode=batch --set youtube_profile=video_custom --set youtube_resolution=480 --set youtube_container=mp4 --set "youtube_options="</code>
<code>yt-batch-best-mp4</code>	The best MP4 from the links of input\urls.txt	-	<code>media_cli.cmd run --operation youtube_run --set youtube_source_mode=batch --set youtube_profile=best_mp4 --set "youtube_options="</code>

Wrapper	What it is	Takes	The same through the console command
<code>yt-best-av1</code>	The best AV1 in MP4	urls	<code>media_cli.cmd run --operation youtube_run --set youtube_profile=video_custom --set youtube_video_codec=av1 --set youtube_container=mp4 --set "youtube_options="</code>
<code>yt-best-m4a</code>	The sound alone, M4A, with cover and tags	urls	<code>media_cli.cmd run --operation youtube_run --set youtube_profile=audio_custom --set youtube_audio_format=m4a --set youtube_options=embed_metadata</code>
<code>yt-best-mp4</code>	The best picture in MP4	urls	<code>media_cli.cmd run --operation youtube_run --set youtube_profile=best_mp4 --set "youtube_options="</code>
<code>yt-best-mp4-safari</code>	The best MP4 as Safari asks for it	urls	<code>media_cli.cmd run --operation youtube_run --set youtube_profile=best_mp4 --set youtube_client=safari --set "youtube_options="</code>
<code>yt-best-mp4-safe</code>	The best MP4, slowly and with retries	urls	<code>media_cli.cmd run --operation youtube_run --set youtube_profile=safe_mp4</code>
<code>yt-clip-sections</code>	A section of a video: the first argument is *00:01:00-00:02:00	sections	<code>media_cli.cmd run --operation youtube_run --set youtube_profile=best_mp4 --set "youtube_options="</code>
<code>yt-embed-sub</code>	The best MP4 with subtitles inside	urls	<code>media_cli.cmd run --operation youtube_run --set youtube_profile=best_mp4 --set "youtube_options=" --set subtitle_options=write_subs,embed_subs --set "subtitle_languages=ru,en.*"</code>
<code>yt-mp3</code>	The sound alone, MP3, with cover and tags	urls	<code>media_cli.cmd run --operation youtube_run --set youtube_profile=audio_custom --set youtube_audio_format=mp3 --set youtube_options=embed_metadata</code>
<code>yt-opus</code>	The sound alone, Opus, with cover and tags	urls	<code>media_cli.cmd run --operation youtube_run --set youtube_profile=audio_custom --set youtube_audio_format=opus --set youtube_options=embed_metadata</code>

Wrapper	What it is	Takes	The same through the console command
yt-playlist-1080p-avc	A whole playlist, 1080p H.264, numbered	urls	media_cli.cmd run --operation youtube_run --set youtube_source_mode=playlist --set youtube_profile=upload_ready --set youtube_resolution=1080 --set youtube_video_codec=avc --set youtube_container=mp4 --set "youtube_options="
yt-playlist-best	A whole playlist, the best MP4, numbered	urls	media_cli.cmd run --operation youtube_run --set youtube_source_mode=playlist --set youtube_profile=best_mp4 --set "youtube_options="
yt-selftest	Versions of FFmpeg, ffprobe, yt-dlp and deno	-	media_cli.cmd run --operation selftest
yt-subtitles-only	The subtitles alone	urls	media_cli.cmd run --operation youtube_run --set youtube_profile=subs_only --set "youtube_options=" --set subtitle_options=write_subs --set "subtitle_languages=ru,en.*"
yt-update	Update yt-dlp	-	media_cli.cmd run --operation install_ytdlp
yt-webm-vp9	VP9 in WebM	urls	media_cli.cmd run --operation youtube_run --set youtube_profile=video_custom --set youtube_video_codec=vp9 --set youtube_container=webm --set "youtube_options="

Environment variables the wrappers read

Field	Variables
amf_quality	AUDION_AMF_QUALITY
audio_bitrate	AUDION_AUDIO_BITRATE , AUDIO_BITRATE
audio_copy_container	AUDION_AUDIO_COPY_CONTAINER , AUDIO_COPY_CONTAINER

Field	Variables
audio_mode	AUDION_AUDIO_MODE , AUDIO_MODE
audio_sample_rate	AUDION_AUDIO_SAMPLE_RATE , AUDIO_SAMPLE_RATE
av1_preset	AUDION_AV1_PRESET , AV1_PRESET
batch_file	AUDION_URLS_FILE , URLS_FILE , LIST
cpu_encoder_preset	AUDION_CPU_ENCODER_PRESET , AUDION_ENCODER_PRESET , ENCODER_PRESET , PRESET
cq	AUDION_CQ , AUDION_QP , CQ , QP
crf	AUDION_CRF , CRF
decode_backend	AUDION_DECODE_BACKEND , DECODE_BACKEND
encode_backend	AUDION_ENCODE_BACKEND , AUDION_BACKEND , ENCODE_BACKEND
encode_tuning	AUDION_ENCODE_TUNING , ENCODE_TUNING
fps_audio_oversample	AUDION_FPS_AUDIO_OVERSAMPLE
fps_audio_pcm_depth	AUDION_FPS_AUDIO_PCM_DEPTH
fps_output_profile	AUDION_FPS_OUTPUT_PROFILE , OUT_MODE
input_formats	AUDION_INPUT_FORMATS , INPUT_FORMATS
limit_first_file	AUDION_LIMIT_FIRST_FILE , AUDION_FIRST_FILE , FIRST_FILE
lut_file	AUDION_LUT_FILE , LUT_FILE , ACTIVE_LUT
mp3_lame_preset	AUDION_MP3_LAME_PRESET , MP3_LAME_PRESET
nvenc_preset	AUDION_NVENC_PRESET
overwrite	AUDION_OVERWRITE , OVERWRITE
pix_fmt	AUDION_PIX_FMT , PIX_FMT
pre_gamma	AUDION_PRE_GAMMA , PRE_GAMMA
qsv_encoder_preset	AUDION_QSV_ENCODER_PRESET

Field	Variables
target_codecs	AUDION_TARGET_CODECS , TARGET_CODECS
trim_audio	AUDION_TRIM_AUDIO
trim_audio_channel	AUDION_TRIM_AUDIO_CHANNEL
trim_container	AUDION_TRIM_CONTAINER
trim_data	AUDION_TRIM_DATA
trim_end	AUDION_TRIM_END , TRIM_END
trim_faststart	AUDION_TRIM_FASTSTART
trim_file	AUDION_TRIM_FILE , TRIM_FILE
trim_start	AUDION_TRIM_START , TRIM_START
trim_subtitles	AUDION_TRIM_SUBTITLES
trim_timecode	AUDION_TRIM_TIMECODE
youtube_url	AUDION_YOUTUBE_URL , YOUTUBE_URL , URL