

[Reference] Algorithm of ADPCM

A. Procedure of ADPCM voice analysis

- ① AD conversion The voice is converted into the PCM data of 8bit in each sampling rate.
- ② 8→16 conversion 256 obtained PCM data is multiplied, and the data of 16bit; It converts it into X_n .
- ③ Calculation of d_n This X_n is compared with forecast value \hat{x}_n , and the difference; d_n is obtained.
- ④ Decision of ADPCM data
 It is "0" that d_n is positive as for MSB(L4) of ADPCM data. It makes it to "1" negatively.
 Absolute value of difference; $|d_n|$ Width of quantization; Remainder 3bit(L3,L2,L1) of ADPCM data is decided from the relation of Δ_n .
 Encoding ADPCM data is as shown in Table 5.1.

Table 5-1: ADPCM data and quantization width change rate(f)

L4		L3	L2	L1	f	Condition ($l_n = d_n / \Delta_n$)
$D_n \geq 0$	$D_n < 0$					
0	1	0	0	0	57/64	$l_n < 1/4$
		0	0	1	57/64	$1/4 \leq l_n < 1/2$
		0	1	0	57/64	$1/2 \leq l_n < 3/4$
		0	1	1	57/64	$3/4 \leq l_n < 1$
		1	0	0	77/64	$1 \leq l_n < 5/4$
		1	0	1	102/64	$5/4 \leq l_n < 3/2$
		1	1	0	128/64	$3/2 \leq l_n < 7/4$
		1	1	1	153/64	$7/4 \leq l_n$

Conversion from the voice data to ADPCM data ends because of the above-mentioned operation.

- ⑤ Update of forecast value and width of quantization
 When ADPCM data is obtained, the forecast value of the next step; \hat{x}_{n+1}
 Width of quantization; Δ_{n+1} is renewed.

$$\hat{x}_{n+1} = (1 - 2 * L4) * (L3 + L2/2 + L1/4 + 1/8) * \Delta_n + \hat{x}_n$$

$$\Delta_{n+1} = f(L3, L2, L1) * \Delta_n \quad : \Delta_{nmin}=127, \Delta_{nmax}=24576$$

*Initialization: Forecast value $\hat{x}_1=0$
 Width of quantization $\Delta_1=127$

Hereafter, the voice analysis is done repeating the operation of ①-⑤ every each sampling time.