



PCD04I Features

Turbo Decoder

- 16 state constituent decoder
- Rate 1/2, 1/3, 1/4 or 1/5
- Inmarsat compatible
- Data lengths from 1 to 4092, 6140, 20476 or 22524 bits
- External interleaver address table
- Up to 89 MHz internal clock
- Up to 8.4 Mbit/s with 5 decoder iterations
- 6-bit signed magnitude input data
- Log-MAP or max-log-MAP constituent decoder algorithms
- Up to 128 iterations in 1/2 iteration steps
- Power efficient early stopping
- Extrinsic information output with optional scaling and limiting
- Estimated channel error output
- Free simulation software

Viterbi Decoder (Available Separately)

- 64 or 256 state (constraint length 7 or 9)
- Rate 1/2, 1/3 or 1/4
- Block length from 1 to 32760 (256 state) or 32762 (64 state) bits
- Up to 2.0 Mbit/s (256 state) or 7.0 Mbit/s (64 state)
- 4-bit signed magnitude input data
- Estimated channel error output
- Available as EDIF core and VHDL simulation core for Xilinx Virtex-II Pro, Spartan-3 family and Virtex-4 FPGAs under SignOnce IP License
- Available as VHDL core for ASICs
- Low cost university license also available

Introduction

The PCD04I is a 16 state parallel concatenated error control turbo decoder. The interleaver address table is external to the core. Data lengths from 1 to 4092, 6140, 20476 or 22524 bits can be implemented. Interleaver sizes are 4 bits greater, i.e., either 4096, 6144, 20480 or 22528 bits. Turbo code rates from 1/2 to 1/5 can be selected. The un-interleaved data is terminated with a tail. The data and this tail are interleaved. No tail is added to the interleaved data, implying that the

end state of the interleaved data is unknown. The input block size is K . The interleaver size is $K+4$. The number of coded bits is $n(K+4)$ where the nominal code rate is $1/n$.

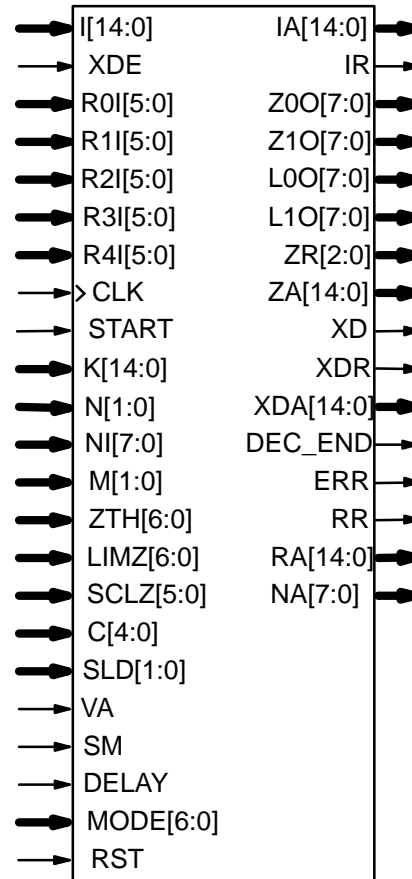


Figure 1: PCD04I schematic symbol.

The MAP04V MAP decoder core is used with the PCD04I core to iteratively decode the turbo code. The Log-MAP algorithm for maximum performance or the max-log-MAP algorithm for minimum complexity can be selected. The sliding block algorithm is used with sliding block lengths of 32, 64, or 128. Six-bit quantisation is used for maximum performance. The extrinsic information can be scaled and limited with each half iteration, improving performance with max-log-MAP decoding. The extrinsic information of both the data and parity bits of the constituent code are also output. The decoder is Inmarsat compatible.

The VA08V Viterbi decoder core can be used with the PCD04I core to decode 64 or 256 state rate 1/2 to 1/4 convolutional codes. The decoder shares its traceback memory with the internal interleaver memory of the turbo decoder, minimising complexity. Maximum traceback lengths of 48 or 96 bits for 64 states or 60 or 120 bits for 256 states can be selected. 4-bit quantisation is used.

The turbo decoder can achieve up to 8.4 Mbit/s with 5 iterations using an 89 MHz internal clock ($K = 5120$). Optional early stopping allows the decoder to greatly reduce power consumption with little degradation in performance. The Viterbi decoder can achieve 2.0 Mbit/s with 256 states and 7.0 Mbit/s with 64 states with $K = 504$ and 506, respectively.

Figure 1 shows the schematic symbol for the PCD04I decoder. This symbol is used to compile various BIT files for download into Xilinx FPGA's. Table 1 shows the performance achieved with various Xilinx parts. T_{cp} is the minimum clock period over recommended operating conditions. These performance figures may change due to device utilisation and configuration.

Table 1: Performance of Xilinx parts.

Xilinx Part	T_{cp} (ns)	Turbo* Mbit/s	K=9 Mbit/s	K=7 Mbit/s
XC3S1000-4	23.339	4.06	0.99	3.38
XC3S1000-5	20.369	4.66	1.13	3.88
XC2VP7-5	17.137	5.54	1.35	4.61
XC2VP7-6	15.255	6.22	1.51	5.18
XC2VP7-7	13.196	7.19	1.75	5.98
XC4VLX15-10	15.211	6.24	1.52	5.19
XC4VLX15-11	12.975	7.32	1.78	6.09
XC4VLX15-12	11.199	8.48	2.06	7.05

*small log-MAP, 5 iterations, $K = 5120$

Table 2 shows the resources used for Virtex-II Pro and Spartan-3/Virtex-4 devices (L00 and L10 are not connected). The MODE[6:0] inputs can be used to select various decoder implementations. The input/output memory is not included. Only one global clock is used. No other resources are used. The RAMs refer to BlockRAMs. Turbo* indicates small log-MAP.

Signal Descriptions

C	MAP Decoder Constant 0-9 (MODE1 = 0) 0-17 (MODE1 = 1)
CLK	System Clock
DEC_END	Decode End Signal

DELAY	Viterbi Decoder Delay 0 = delay 70 (SM = 0) 0 = delay 72 (SM = 1) 1 = delay 134 (SM = 0) 1 = delay 136 (SM = 1)
ERR	Estimated Error
I	Interleaver Address Input
IA	Interleaver Address Output
IR	Interleaver Address Ready
K	Data Length (1 – 4092, 6140, 20476 or 22524)
M	Early Stopping Mode 0 = no early stopping 1 = early stop at odd half iteration 2 = early stop at even half iteration 3 = early stop at any half iteration
MODE	Implementation Mode (see Table 3)
L00	Data Log-Likelihood Information
L10	Parity Log-Likelihood Information
LIMZ	Extrinsic Information Limit (1-127)
N	Code Rate 2 = rate 1/2 3 = rate 1/3 0 = rate 1/4 1 = rate 1/5 (turbo only)
NA	Half Iteration Number (0-255)
NI	Number of Half Iterations (0-255) NI = 2I-1 where I is number of iterations
R0I-R4I	Received Data
RA	Received Data Address
RR	Received Data Ready
RST	Synchronous Reset
SCLZ	Extrinsic Information Scale (1-32)
SLD	MAP Decoder Delay 0 = delay 137 1 = delay 265 2 = delay 521
SM	Viterbi Decoder Memory 0 = 64 state (constraint length 7) 1 = 256 state (constraint length 9)
START	Decoder Start
VA	Viterbi Decoder Select 0 = turbo decoder 1 = Viterbi decoder
XD	Decoded Data
XDA	Decoded Data and Z0A Address
XDE	Decoded Data Enable
XDR	Decoded Data Ready
Z00	Data Extrinsic Information
Z10	Parity Extrinsic Information
ZA	Z10 Address
ZR	Extrinsic Information Ready
ZTH	Early Stopping Threshold (1-127)

Table 2: Resources used

Configuration	K	SLD	Turbo Rates	Viterbi Rates	Virtex-IIIP Slices	Spartan-3/Virtex-4 Slices	Block-RAMs
Turbo (max-log-MAP)	6140	1	1/3	–	2971	2973	3
Turbo (small log-MAP)	6140	1	1/3	–	4122	4119	3
Turbo (large log-MAP)	6140	1	1/3	–	4796	4796	3
Turbo (small log-MAP)	6140	2	1/3	–	4926	5008	3
Turbo* and Viterbi	6140	1	1/2–1/3	1/2–1/3	4672	4709	3
Turbo* and Viterbi	22524	1	1/2–1/5	1/2–1/4	4926	5031	11

Table 3 describes each of the MODE[6:0] inputs that are used to select various decoder implementations. Note that MODE[6:0] are “soft” inputs and should not be connected to input pins or logic. These inputs are designed to minimise decoder complexity for the configuration selected

Table 3: MODE selection

Input	Description
MODE0	0 = max-log-MAP 1 = log-MAP
MODE1	0 = small log-MAP (C4 = 0) 1 = large log-MAP
MODE2	0 = rate 1/2–1/3 turbo 1 = rate 1/2–1/5 turbo
MODE3	0 = turbo decoder 1 = turbo and Viterbi decoder
MODE4	0 = rate 1/2–1/3 Viterbi 1 = rate 1/2–1/4 Viterbi
MODE[6:5]	0 = 4K Interleaver 1 = 6K Interleaver 2 = 20K Interleaver 3 = 22K Interleaver

For MODE[6:5] = 0 to 3, Virtex-II Pro, Spartan-3 and Virtex-4 uses 2, 3, 10 and 11 Block-RAMs, respectively.

Turbo Decoder Parameters

For optimal performance, the maximum a posteriori (MAP) [1] constituent decoder is used which is dependent on the signal to noise ratio (SNR). Unlike other turbo decoders with suboptimum soft-in-soft-in (SISO) decoders, using the MAP (or specifically the log-MAP [2]) algorithm can provide up to 0.5 dB coding gain at low SNRs. Log-MAP operation is enabled when MODE0 is high.

BPSK and QPSK Parameters

With binary phase shift keying (BPSK, $m = 1$) or quadrature phase shift keying (QPSK, $m = 2$)

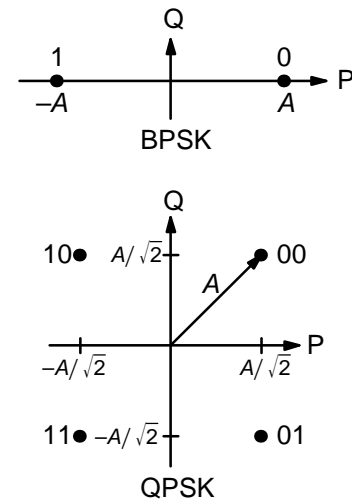


Figure 2: BPSK and QPSK signal sets.

modulation (see Figure 2) the decoder constant C should be adjusted such that

$$C = A\sigma^2 \sqrt{m}/2. \quad (1)$$

where A is the signal amplitude and σ^2 is the normalised noise variance given by

$$\sigma^2 = 1/(2mRE_b/N_0). \quad (2)$$

E_b/N_0 is the energy per bit to single sided noise density ratio and $R = 1/(n+6(\lfloor n/2 \rfloor + 1)/K)$, $n = 2-5$, $K = 40-5114$ is the code rate. C should be rounded to the nearest integer and limited to be no higher than 17 with MODE1 high and 9 with MODE1 low. Max-log-MAP [2] operation occurs when $C = 0$. Due to quantisation effects, $C = 1$ is equivalent to $C = 0$. Max-Log-MAP operation is also enabled when MODE0 is low.

For fading channels the value of A and σ^2 should be averaged across the block to determine the average value of C . Each received value r_k should then be scaled by $(A\sigma^2)/(A_k\sigma_k^2)$ where A_k and σ_k^2 are the amplitude and normalised variance of r_k . Note that this scaling should be performed for both the log-MAP and max-log-MAP algorithms for optimal performance.

The value of A directly corresponds to the 6-bit signed magnitude inputs (shown in Table 4). The 6-bit inputs have 63 quantisation regions with a central dead zone. The quantisation regions are labelled from -31 to $+31$. For example, one could have $A = 15.7$. This value of A lies in quantisation region 15 (which has a range between 15 and 16).

Since most analogue to digital (A/D) converters do not have a central dead zone, a 7-bit A/D should be used and then converted to 6-bit as shown in the table. This allows maximum performance to be achieved.

Due to quantisation and limiting effects the value of A should also be adjusted according to the received signal to noise ratio.

Example 1: Rate 1/3 BPSK code operating at $E_b/N_0 = 0.3$ dB. From (2) we have $\sigma^2 = 1.39988$. Assuming $A = 16$ we have from (1) that $C = 11$ to the nearest integer.

Table 4: Quantisation for R0I, R1I and R2I.

Decimal	Binary	Range
31	011111	30.5 ↔ ∞
30	011110	29.5 ↔ 30.5
⋮	⋮	⋮
2	000010	1.5 ↔ 2.5
1	000001	0.5 ↔ 1.5
0	000000	-0.5 ↔ 0.5
32	100000	-0.5 ↔ 0.5
33	100001	-1.5 ↔ -0.5
34	100010	-2.5 ↔ -1.5
⋮	⋮	⋮
62	111110	-30.5 ↔ -29.5
63	111111	-∞ ↔ -30.5

16QAM Parameters

The Inmarsat standard has the following mapping in the inphase (I) and quadrature (Q) components:

I1	I0	I	Q1	Q0	Q
0	1	-3	0	1	-3
0	0	-1	0	0	-1
1	0	1	1	0	1
1	1	3	1	1	3

For a 2-D signal set consisting of (I,Q), I,Q = $\{-3,-1,1,3\}$ the average energy is 10. This implies that the average signal amplitude is $\sqrt{10} = 3.16$.

Let the received 2-D signal be (x_k, y_k) where

$$x_k = s_k^I + \sqrt{10} n_k^I \quad (3)$$

$$y_k = s_k^Q + \sqrt{10} n_k^Q \quad (4)$$

where n_k^I and n_k^Q are the additive inphase and quadrature white Gaussian noise (AWGN), respectively, with zero mean and normalised variance σ^2 (2). We have $m = 4$ is the number of bits per symbol and $R = 1/2$ is the code rate (the Inmarsat code is punctured to give a rate of exactly a 1/2). We have multiplied n_k^I and n_k^Q by $\sqrt{10}$ in (3) and (4), respectively, to take into consideration that (2) is valid only if the signal set has been normalised to an energy of one.

In the Inmarsat code, I1 and Q1 are mapped with data bits and I0 and Q0 are mapped with parity bits (due to the extra puncturing four data bits are also mapped to I0 and Q0). We shall call the data bits d_k^I and d_k^Q and the parity bits p_k^I and p_k^Q . Since the in-phase and quadrature mapping are the same, we shall ignore the superscript I and Q notation in the following.

The likelihood ratio of the data bit d_k and the parity bit p_k are given by

$$r_k^d = \frac{P(d_k = 1|x_k)}{P(d_k = 0|x_k)} = \frac{\sum_{s=\{1,3\}} \exp\left(\frac{-1}{2\sigma^2 10} (x_k - s)^2\right)}{\sum_{s=\{-1,-3\}} \exp\left(\frac{-1}{2\sigma^2 10} (x_k - s)^2\right)} \quad (5)$$

$$r_k^p = \frac{P(p_k = 1|x_k)}{P(p_k = 0|x_k)} = \frac{\sum_{s=\{-3,3\}} \exp\left(\frac{-1}{2\sigma^2 10} (x_k - s)^2\right)}{\sum_{s=\{-1,1\}} \exp\left(\frac{-1}{2\sigma^2 10} (x_k - s)^2\right)} \quad (6)$$

Notice that we have a division by 10 in the exponential so as to normalise the signal set energy to one. Taking the log-likelihood ratios, we have

$$R_k^d = -\log_\epsilon r_k^d = [B(x_k - 1)^2 E B(x_k - 3)^2] - [B(x_k + 1)^2 E B(x_k + 3)^2] \quad (7)$$

$$R_k^p = -\log_\epsilon r_k^p = [B(x_k - 3)^2 E B(x_k + 3)^2] - [B(x_k - 1)^2 E B(x_k + 1)^2] \quad (8)$$

where

$$a E b = \min(a, b) - C \ln(1 + e^{-|a-b|/C}) \quad (9)$$

and $C = 1/\ln \epsilon = \log_\epsilon e$ with

$$B = -\log_\epsilon e^{-1/(2\sigma^2 10)} = 0.05C/\sigma^2. \quad (10)$$

Using the fact that $(a + b) E (a + c) = a + [b E c]$ we can simplify (7) and (8) to

$$R_k^d = [Ax_k/2 E A] - [(Ax_k/2 + A) E 0] - Ax_k \quad (11)$$

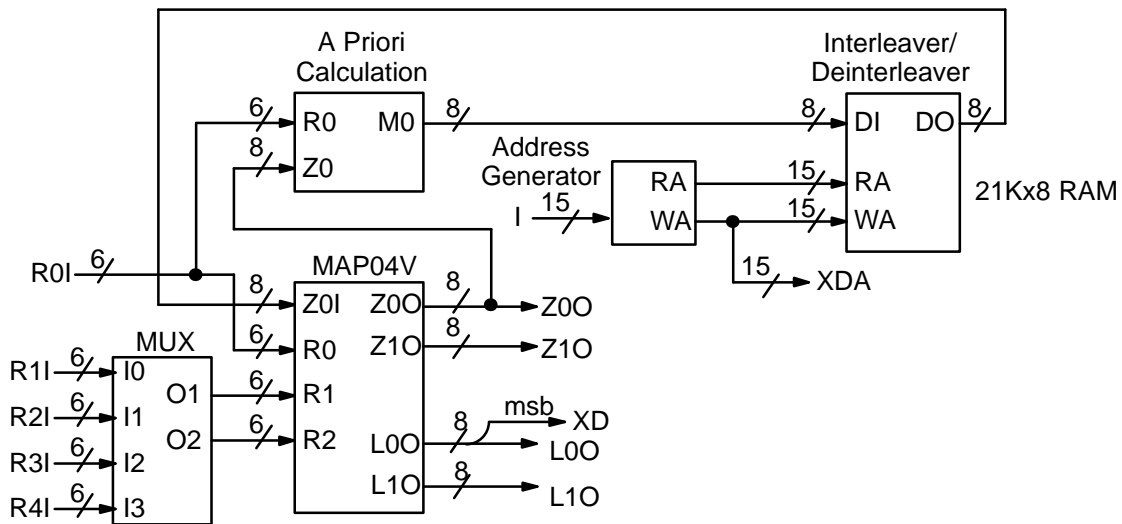


Figure 3: Simplified block diagram of PCD04I 16 state turbo decoder.

$$R_k^p = ([3Ax_k/2 \ E \ 0] - [Ax_k/2 \ E \ 0] - Ax_k/2 + A) \quad (12)$$

where

$$A = 8B = 0.4C/\sigma^2. \quad (13)$$

Note that Ax_k represents the received value in quantised form. For example, the received quantised form could be from -31 to 31 , with the transmitted signal set points being $\{-24, -8, 8, 24\}$ implying that $A = 8$. Note that A does not have to be an integer. Knowing the normalised variance σ^2 and A we can calculate C to be

$$C = 2.5A\sigma^2. \quad (14)$$

For large values of Ax_k , R_k^d is approximately equal to $-Ax_k$. Since a transmitted signal can equal $3Ax_k$ with no noise, then R_k^d can vary from $3Ax_k$ to $-3Ax_k$.

The data log-likelihood ratios R_k^d should be input to R0I[5:0] and the parity log-likelihood ratios R_k^p should be input to either R1I[5:0] or R2I[5:0], depending on whether the parity information belongs to non-interleaved or interleaved data, respectively. If a parity input has been punctured then 0 should be input to either R1I[5:0] or R2I[5:0].

Example 2: Let $E_b/N_0 = 4.0$ dB and $A = 8$. This implies $\sigma^2 = 0.0995$ and $C = 2$ to the nearest integer.

For a fading channel where the amplitude A_k varies with time, we should substitute A with A_k in (11) and (12) and calculate C from

$$C = \frac{5\sigma^2}{K} \sum_{k=0}^{K/2-1} A_k \quad (15)$$

where K is the number of data bits.

Other Parameters

Figure 3 gives a block diagram of the PCD04I 16 state turbo decoder. The number of turbo decoder half-iterations is given by NI , ranging from 0 to 255. $NI = 2I - 1$ where I is the number of iterations. This is equivalent to 0.5 to 128 iterations. The decoder initially starts at half iteration $NA = 0$, increasing by one until NI is reached or an earlier time if early stopping is enabled. The NA output can be used to select LIMZ and SCLZ values, especially for max-log-MAP decoding.

The turbo decoder speed f_d is given by

$$f_d = \frac{F_d}{(NI + 1)(1 + (L + 5)/K) + 1/K} \quad (16)$$

where F_d is the CLK frequency and L is the MAP decoder delay in bits (equal to either 137, 265, or 521). The three delays indicate the sliding block length used in the MAP decoder, either 32, 64, or 128, respectively. For short block lengths $L = 137$ should be used to increase decoder speed, while $L = 265$ should be used for larger block sizes to increase performance. For high rate punctured turbo codes, $L = 521$ should be used. This parameter can be selected with the SLD input.

For example, if $F_d = 50$ MHz and $I = 5$ ($NI = 9$) the decoder speed ranges from 1.1 Mbit/s for $K = 40$ and $L = 137$ to 4.7 Mbit/s for $K = 5116$ and $L = 265$.

An important parameter is LIMZ, the limit factors for the extrinsic information. Extrinsic information is the "correction" term that the MAP decoder determines from the received data and a *priori* information. It is used as a *priori* information for the next MAP decoding or half iteration. By limiting the correction term, we can pre-

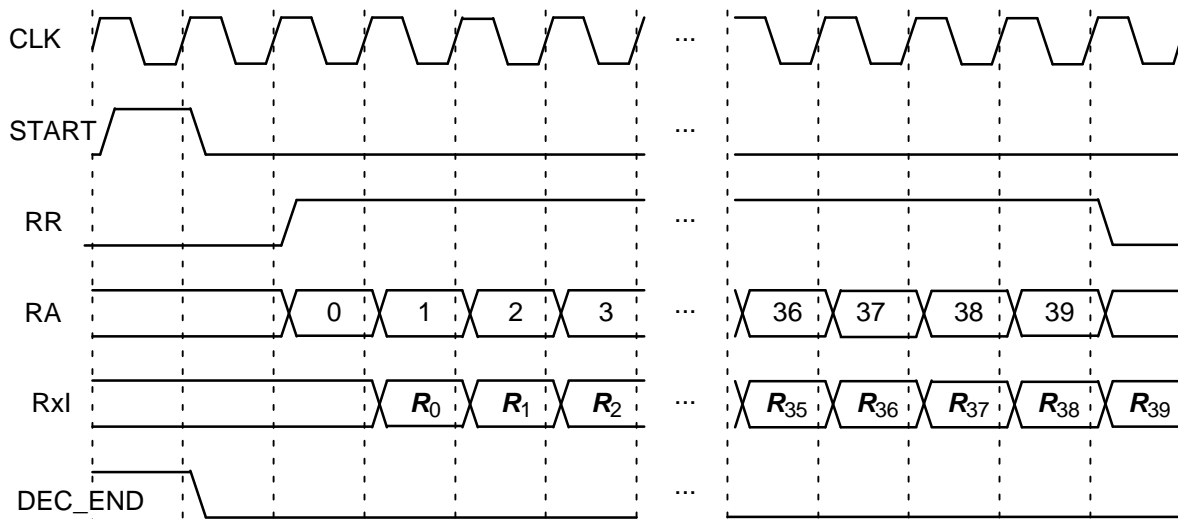


Figure 4: Turbo Decoder Input Timing ($K = 36$).

vent the decoder from making decisions too early, which improves decoder performance.

The limit factor LIMZ should vary between 1 and 127, although we recommend that 96 be used.

Another parameter that can be used to adjust decoder performance is SCLZ which ranges from 1 to 32. The extrinsic information is scaled by $SCLZ/32$. Thus, when $SCLZ = 32$, no scaling is performed. For log-MAP decoding we recommend $SCLZ = 29$. For max-log-MAP decoding we recommend $SCLZ = 23$. The NA output can be used to adjust LIMZ and SCLZ with the number of iterations for optimum performance.

There are four decoder operation modes given by M . Mode $M = 0$ decodes a received block with a fixed number of iterations (given by N). Modes 1 to 3 are various early stopping algorithms. Early stopping is used to stop the decoder from iterating further once it has estimated there are zero errors in the block. Mode 1 will stop decoding after an odd number of half-iterations. Mode 2 will stop decoding after an even number of half iterations. Mode 3 will stop after either an odd or even number of half iterations. Further details are given in the next section.

Turbo Decoder Operation

After the START signal is sent, the decoder will read the received data at the CLK speed. It is assumed that the received data is stored in a synchronous read RAM of size $(K+4) \times 6n$, $n = 2$ to 5. The received data ready signal RR goes high to indicate the data to be read from the address given by RA[14:0]. Table 5 illustrates which data is stored for address 0 to $K-1$ for the main data and K to $K+3$ for the tail. The entries for the table indi-

cate which encoded data output is selected, X, Y1 and Y2 for the first encoder and X', Y1' and Y2' for the second encoder. The code polynomials are $g^0(D) = 1+D^3+D^4$ (23 in octal), $g^1(D) = 1+D+D^2+D^4$ (35) and $g^2(D) = 1+D+D^2+D^3+D^4$ (37). For rate 1/2 and 1/4 the data and tail are punctured, which is why two entries are shown.

The decoder then iteratively decodes the received data for $N+1$ half iterations, rereading the received data for each half iteration for $K+4$ CLK cycles. The signal RR goes high for $K+4$ clock cycles while data is being output. Figure 4 illustrates the decoder timing where the data is input on the first half iteration.

Table 5: Input data format

Rate	Data	Output
1/2	R0I	X X
	R1I	Y1 Y1'
1/3	R0I	X
	R1I	Y1
	R2I	Y1'
1/4	R0I	X X
	R1I	Y1 Y1
	R2I	Y2 Y1'
	R3I	Y2' Y2'
1/5	R0I	X
	R1I	Y1
	R2I	Y2
	R3I	Y1'
	R4I	Y2'

If the START signal goes high while decoding, the decoder is reset and decoding starts anew. A synchronous reset is also provided. All flip flops in

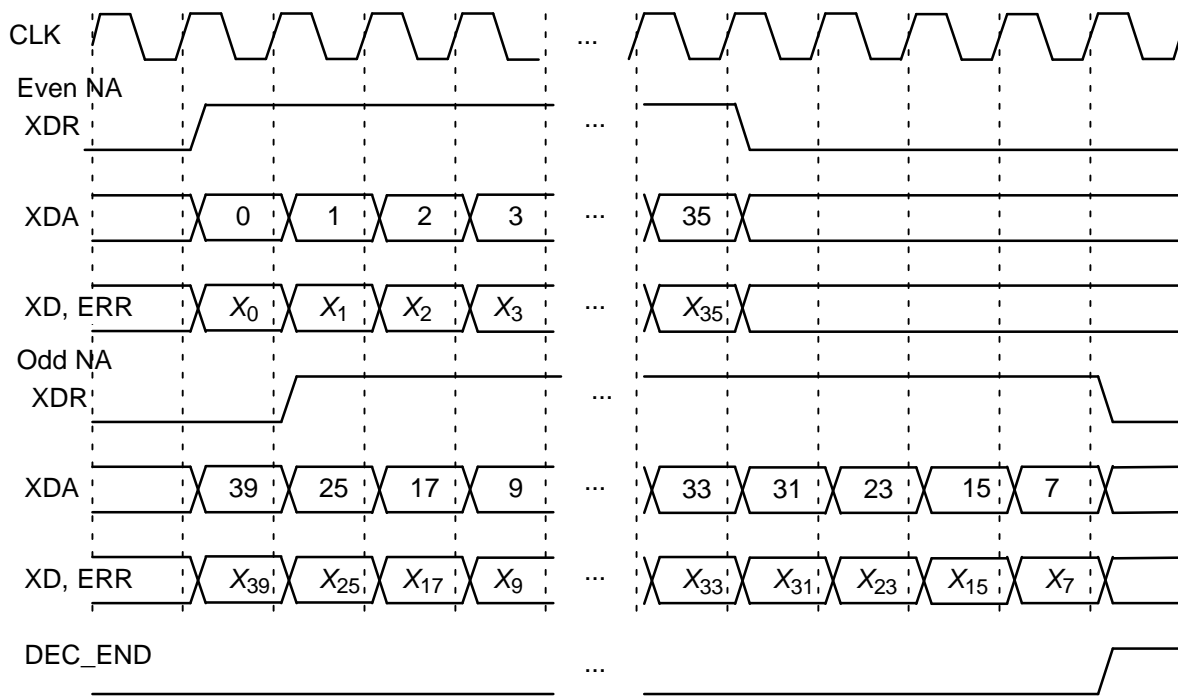


Figure 5: Turbo Decoder Output Timing ($K = 36$).

the turbo decoder are reset during a low to high transition of CLK while RST is high.

The decoded block is output during the last half-iteration. The signal XDR goes high for K CLK cycles while the block is output. If NI is even, the block is output in sequential order. For NI odd, the block is output in interleaved order. To deinterleave the block, the output XDA[14:0] can be used as the write address to a buffer RAM. After the block has been written to the buffer RAM, the decoded block can be sequentially read from the buffer RAM.

The signal ERR is a channel error estimator output. For even NI , it is the exclusive OR of XD and the sign bit of the input R0I. For odd NI , it is the exclusive OR of XD re-encoded to give the first parity bit and the sign bit of the input corresponding to $Y1'$ (this is because R0I is input in sequential order, not in the interleaved order of the output). If the output of the MAP decoder has zero errors, then this gives an approximation of the channel bit error rate (BER). Since $Y1'$ is punctured for rate 1/2 and 1/4, the number of bits counted is $\lfloor K/2 \rfloor$.

The DEC_END signal is low during decoding. At the end of decoding, DEC_END goes high. Figure 5 illustrates the decoder timing where data is output on the last half iteration. Note that for even half iterations (odd NA), XDR only goes high when $XDA < K$. After startup, the maximum number of clock cycles for decoding is $(NI+1)(K+L+5)+1$.

During the last half iteration the decoded data is stored into the interleaver memory. Once decoding has been completed, the input XDE can be used to sequentially clock the decoded data from from the interleaver memory (regardless of the number of iterations). XDE is disabled while the decoder is iterating. Figure 6 shows the decoder timing when XDE is used.

The output ERR is also output when XDE goes high. The outputs RA and RR are used to read the sign bit of R0I which is exclusive-ORed with XD to give ERR.

The early stopping algorithm uses the magnitude of the extrinsic information to determine when to stop. As the decoder iterates, the magnitudes generally increases in value as the decoder becomes more confident in its decision. By comparing the smallest magnitude of a block with threshold ZTH, we can decide when to stop. If the smallest magnitude is greater than ZTH, i.e., not equal or less than ZTH, the decoder will stop iterating if early stopping has been enabled.

Since the last half iteration is used to store the decoded data into the interleaver memory, the decoder performs an extra half iteration once the threshold has been exceeded.

Increasing ZTH will increase the average number of iterations and decrease the BER. Decreasing ZTH will decrease the average number of iterations and increase the BER. In general, higher values of SNR will decrease the number of

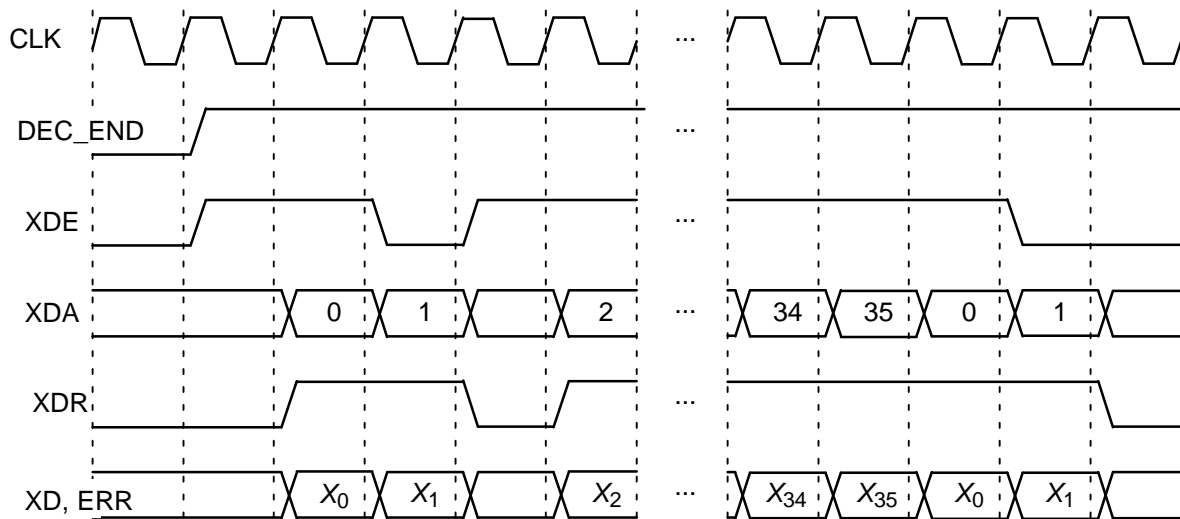


Figure 6: XDE Timing ($K = 36$).

iterations. A value of $ZTH = 23$ was found to give a good trade off between the average number of iterations and BER performance.

For high SNR operation early stopping can lead to significantly reduced power consumption, since most blocks will be decoded in one or two iterations.

The extrinsic (log-likelihood) information from the MAP decoder are output from $Z0O[7:0]$ and $Z1O[7:0]$ ($L0O[7:0]$ and $L1O[7:0]$). The outputs $Z0O$ and $Z1O$ ($L0O$ and $L1O$) corresponds to the data and parity, respectively, of the rate 1/2 MAP decoder. The information for both the data and tail bits are output and are in two's complement form.

$L0O$ contains is the sum of $R0I$, the unchanged (not scaled or limited) $Z0O$ for the current half iteration, and the scaled and limited $Z0O$ from the previous half iteration. $L1O$ is the sum of $R1I$ or $R2I$ and the unchanged $Z1O$ for odd or even half iterations, respectively.

$Z0O$ ($L0O$) is output every half iteration, using XDA as the write address and $ZR0$ is the ready address. For even half iterations (NA odd) $Z0O$ ($L0O$) is interleaved. For odd half iterations (NA even) $Z0O$ ($L0O$) is not interleaved. $Z0R$ is high for $K+4$ clock cycles every half iteration.

$Z1O$ ($L1O$) is also output every half iteration, using ZA as the write address. $Z1O$ ($L1O$) corresponds to the information for $R1I$ and $R2I$ for odd and even half iterations, respectively. The outputs $ZR1$ and $ZR2$ are the corresponding ready addresses. $ZR1$ and $ZR2$ goes high for $K+4$ clock cycles every odd and even half iteration, respectively.

Figure 7 illustrates how to connect $Z0O$ ($L0O$) and $Z1O$ ($L1O$) to three $5K \times 8$ memories. At the

end of every decoding the memories will have stored the information for $R0I$, $R1I$ and $R2I$.

The interleaver address table needs to be stored in an external synchronous ROM. $IA[14:0]$ is the input address to the ROM with $I[14:0]$ being the output of the ROM. The signal IR indicates when IA is valid. Figure 8 shows the timing for the ROM.

Simulation Software

Free software for simulating the PCD04I turbo decoder in additive white Gaussian noise (AWGN) or with external data is available by sending an email to info@sworld.com.au with "pcd04sim request" in the subject header. The software uses an exact functional simulation of the PCD04I turbo decoder, including all quantisation and limiting effects.

After unzipping `pcd04sim.zip`, there should be `pcd04sim.exe` and `code.txt`. The file `code.txt` con-

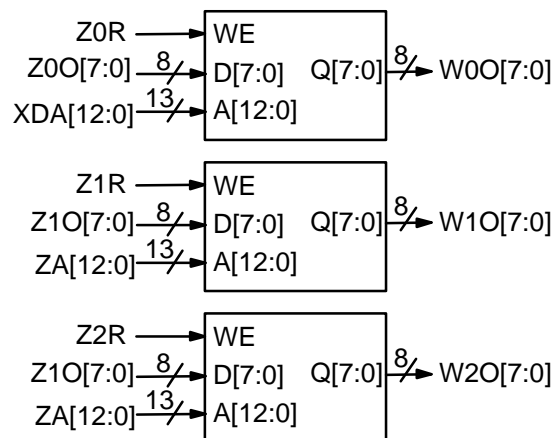


Figure 7: Output RAM for extrinsic information.

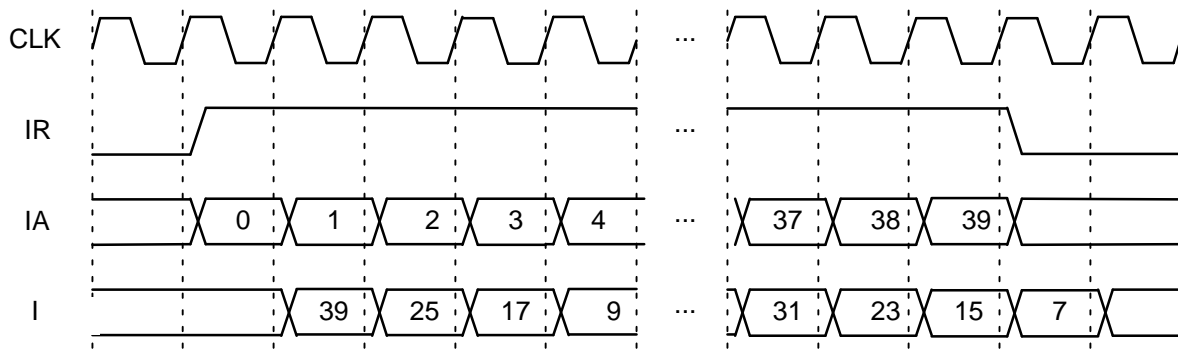


Figure 8: Interleaver address timing ($K = 36$).

tains the parameters for running pcd04sim. These parameters are

- m Constituent code (CC) memory (2 to 4)
- nt Number of turbo code outputs (2 to 5)
- g0 Divisor polynomial of CC in octal notation
- g1 1st numerator polynomial of CC
- g2 2nd numerator polynomial of CC
- EbNomin Minimum E_b/N_0 (in dB)
- EbNomax Maximum E_b/N_0 (in dB)
- EbNoinc E_b/N_0 increment (in dB)
- optC Input scaling parameter (normally 0.65)
- ferrmax Number of frame errors to count
- NI Number of half iterations-1 (0 to 255)
- SLD MAP decoder delay select (0 to 2)
- LIMZ Extrinsic information limit (1 to 127)
- SCLZ Extrinsic information scale (1 to 32)
- M Stopping mode (0 to 3)
- ZTH Extrinsic info. threshold (0 to 127)
- SI Select interleaver (0 or 1)
- K Block length
- q Number of quantisation bits (3 to 6)
- LOGMAP Log-MAP decoding (MODE0, 0 or 1)
- C4PIN Use five-bit C (MODE1, 0 or 1)
- state State file (0 to 2)
- s1 Seed 1 (1 to 2147483562)
- s2 Seed 2 (1 to 2147483398)
- read_x Use external information data (y or n)
- read_r Use external received data (y or n)
- out_dir Output directory
- in_dir Input directory
- C C (0 to 17)

Note that g_0 , g_1 and g_2 are given in octal notation, e.g., $g_0 = 23 \equiv 10011_2 \equiv 1+D^3+D^4$. For the Inmarsat standard, $m = 4$, $nt = 2$, $g_0 = 23$ and $g_1 = 35$ (g_2 is not used). The nominal turbo code rate is $1/nt$.

The parameter $optC$ is used to determine the “optimum” values of A and C . The “optimum” value of A for BPSK or QPSK is

$$A = \frac{optC(2^{q-1} - 1)}{mag(\sigma)} \quad (17)$$

where σ^2 is the normalised noise variance given by (2) and $mag(\sigma)$ is the normalising magnitude resulting from an auto-gain control (AGC) circuit. We have

$$mag(\sigma) = \sigma \sqrt{\frac{2}{\pi}} \exp\left(\frac{-1}{2\sigma^2}\right) + 1 - 2Q\left(\frac{1}{\sigma}\right) \quad (18)$$

where $Q(x)$ is the error function given by

$$Q(x) = \int_x^\infty \frac{1}{\sqrt{2\pi}} \exp\left(-\frac{t^2}{2}\right) dt. \quad (19)$$

Although $mag(\sigma)$ is a complicated function, for high signal to ratio (SNR), $mag(\sigma) \approx 1$. For low SNR, $mag(\sigma) \approx \sigma \sqrt{2/\pi} \approx 0.798\sigma$. That is, an AGC circuit for high SNR has an amplitude close to the real amplitude of the received signal. At lower SNR, the noise increases the estimated amplitude, since an AGC circuit averages the received signal amplitude.

For $SI = 0$, an internal 3GPP (UMTS) interleaver is used. This interleaver is valid from $K = 36$ to 5110. For $SI = 1$, an external interleaver is input from directory in_dir . The file is called $K.dat$, where K is the data length, e.g., $36.dat$. The interleaver length is $K+4$. The interleaver file must be an ASCII file, with an interleaver address in each line, e.g.,

```
39
25
17
9
...
```

For the “optimum” A , we round the value of C given by (1) to the nearest integer. If $LOGMAP = MODE0 = 0$ then C is forced to 0. If $LOGMAP = 1$ and $C4PIN = MODE1 = 0$, C is limited to a maximum value of 9. If $LOGMAP = 1$ and $C4PIN = 1$, C is limited to a maximum value of 17.

When the simulation is finished the output is given in, for example, file $k600.dat$, where $K = 600$. Only frames that are in error are stored in the output file. The first line gives the E_b/N_0 (E_b/N_0), the

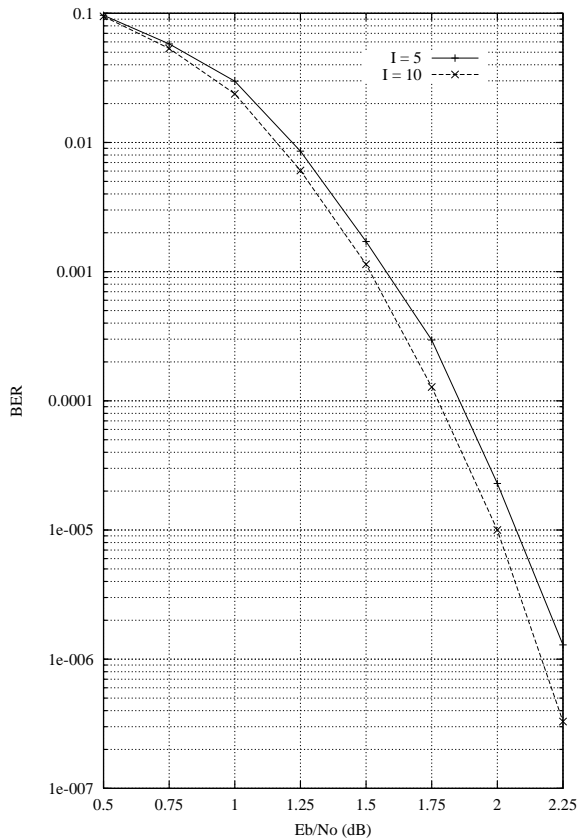


Figure 9: Performance with block size 600 and auto-stopping (ZTH = 23).

number of frames (n_{num}), the number of bit errors in the frame (err), the total number of bit errors (b_{err}), the total number of frame errors (f_{err}), the average number of iterations (na), and the average bit error rate (P_b). Following this, the number of iterations, na , b_{err} , f_{err} , and P_b , are given for each half iteration.

The following file was used to give the simulation results shown in Figure 9. Auto-stopping was used with up to 10 iterations. When iterating is stopped early, the $nasum$ ($2 * num * na$), b_{err} and f_{err} results at stopping are copied for each half iteration to the maximum iteration number. Thus, the $I = 10$ result is the performance one would measure with auto-stopping and $NI = 19$. The $I = 5$ curve shows the performance at 5 iterations with early stopping and $NI = 9$. Figure 10 shows the average number of iterations with E_b/N_0 .

```
{m nt g0 g1 g2}
4 2 23 35 37
{EbNomin EbNomax EbNoinc optC ferrmax}
0.50 2.50 0.25 0.65 64
{NI SLD LIMZ SCLZ M ZTH SI}
19 1 96 32 3 23 0
{K q LOGMAP C4PIN}
600 6 1 1
{state s1 s2}
```

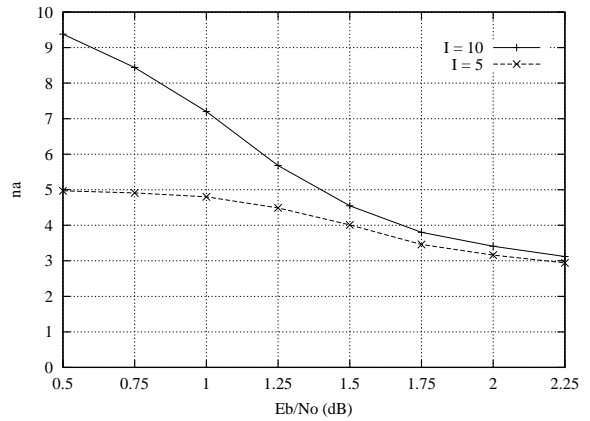


Figure 10: Average number of iterations with block size 600 and auto-stopping (ZTH = 23).

```
0 12345 67890
{read_x read_r out_dir in_dir C}
n n output input 12
```

The `state` input can be used to continue the simulation after the simulation has been stopped, e.g., by the program being closed or your computer crashing. For normal simulations, `state = 0`. While the program is running, the simulation state is alternatively written into `State1.dat` and `State2.dat`. Two state files are used in case the program stops while writing data into one file. To continue the simulation after the program is stopped follow these instructions:

- 1) Copy the state files `State1.dat` and `State2.dat`. This ensures you can restart the program if a mistake is made in `configcode.txt`.
- 2) Examine the state files and choose one that isn't corrupted.
- 3) Change the state parameter to 1 if `State1.dat` is used or 2 if `State2.dat` is used.
- 4) Restart the simulation. The output will be appended to the existing `k(K).dat` file.
- 5) After the simulation has been completed, make sure that `state` is changed back to 0.

The software can also be used to encode and decode external data. To encode a block `X_(K).dat` in the directory given by `in_dir`, set `read_x` to `y`, e.g., `X_600.dat` in directory `input` (each line contains one bit of data). The encoded stream `Y_(K).dat` will be output to the directory given by `out_dir`, e.g., `Y_600.dat` to directory `output`.

To decode data, place the received block of data in file `R_(K).dat` in directory `in_dir` and set `read_r` to `y`. The decoded data is output to `XD_(K).dat` in directory `out_dir`. `R_(K).dat` has in each line `R[i,j]`, $i = 0$ to $nt-1$ from $j = 0$ to $K+m-1$, e.g., for $nt = 3$ the first three lines could be

```
-31 1 -25
```

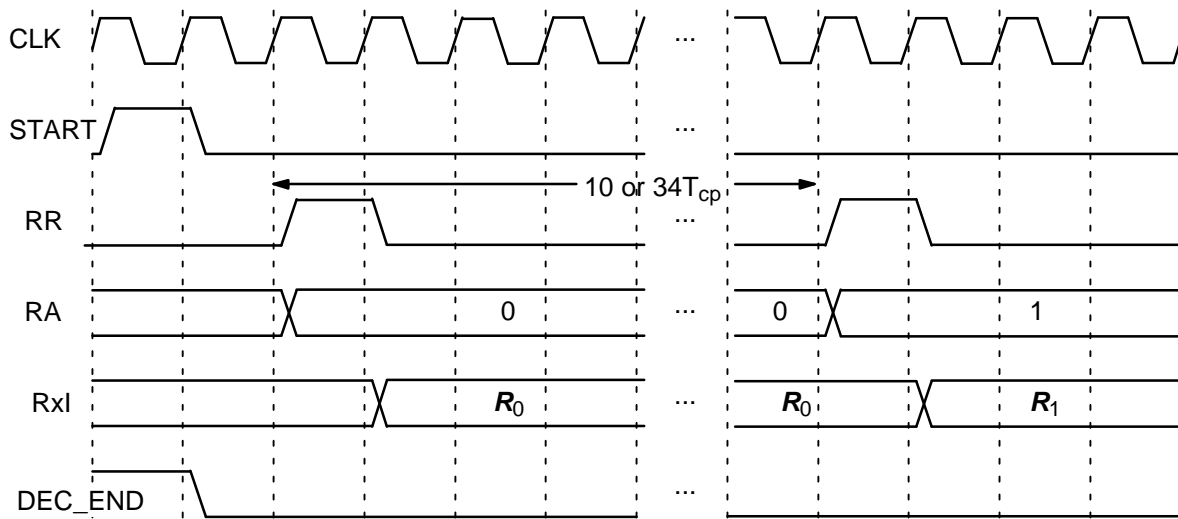


Figure 11: Viterbi Decoder Input Timing.

–31 12 9
11 31 31

The input data is of the form

$$R[i,j] = A*(1-2*Y[i,j]+N[i,j])$$

where A is the signal amplitude, $Y[i,j]$ is the coded bit, and $N[i,j]$ is white Gaussian noise with zero mean and normalised variance σ^2 . The magnitude of $R[i,j]$ should be rounded to the nearest integer and be no greater than $2^{q-1}-1$. If $read_r = y$, then C is externally input via c .

Viterbi Decoder Operation

The Viterbi decoder is operated in a similar way to the turbo decoder. The START signal is used to start decoding, using RR and RA to read the received data. Since 4-bit quantisation is used, the two least significant bits of R0I to R3I are not used. That is, the 4-bit signed magnitude data is input to R0I[5:2], R1I[5:2], R2I[5:2] and R3I[5:2]. For rate 1/2 operation, R2I to R4I are not used. For rate 1/3 operation R3I to R4I are not used. For rate 1/4 R4I is not used.

The input SM selects 64 states (constraint length 7) when low and 256 states (constraint length 9) when high. The input DELAY when low selects either a delay of 70 or 72 (for 64 or 256 states). When high a delay of 134 or 136 (for 64 and 256 states) is selected. Table 6 shows the codes selected with the number of states and code rate.

The decoder first inputs the received data from address 0 to $K-1$. The tail is then input from address K to $K+5$ for 64 states and $K+7$ for 256 states. After a decoding delay, the decoded data

is output to XD. XDR goes high for one clock cycle at the beginning of each decoded bit. XDA goes from address 0 to $K-1$ as the decoded data is output.

The output ERR is the exclusive-OR of the sign bit of R0I with the corresponding re-encoded decoded output bit. This allows an estimate of the channel BER.

Figure 11 shows the Viterbi decoder input timing. Two clock cycles are used to start decoding, with each decoded bit taking 10 clock cycles for 64 states or 34 clock cycles with 256 states.

Table 6: Convolutional Codes.

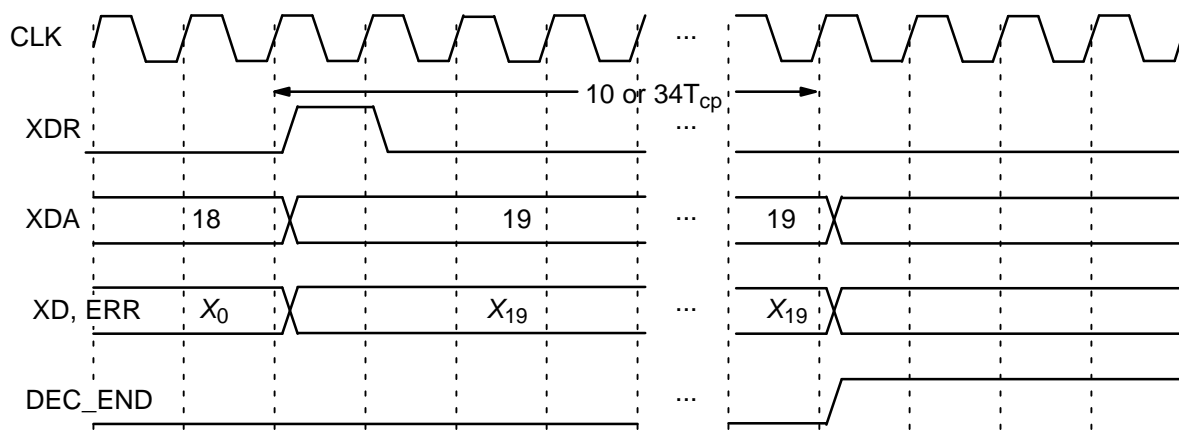
SM	N	G0I	G1I	G2I	G3I
0	2	171	133	–	–
0	3	171	133	165	–
0	0	173	167	135	111
1	2	753	561	–	–
1	3	557	663	711	–
1	0	765	671	513	473

Figure 12 shows the Viterbi decoder output timing. The input XDE is not used either during or after Viterbi decoding.

The decoding speed is given by

$$f_d = \frac{F_d}{N_c(1 + D/K) + 2/K} \quad (20)$$

where F_d is the internal clock speed, N_c is the number of decoder clock cycles (10 or 34) and D is the Viterbi decoder delay in bits. For example, if $K = 504$, $D = 136$ (SM = 1, DELAY = 1), $N_c = 34$ (SM = 1) and $F_d = 50$ MHz, decoding speed is 1.1 Mbit/s.

Figure 12: Viterbi Decoder Output Timing ($K = 20$).

Ordering Information

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References

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- [2] P. Robertson, E. Villebrun, and P. Hoeher, "A comparison of optimal and sub-optimal MAP decoding algorithms operating in the log domain," *ICC'95*, Seattle, WA, USA, pp. 1009–1013, June 1995.

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