LABORATORY DATA COMPRESSION

M. Emoto, M. Shoji, S. Yamaguchi, NIFS, Toki, 509-5292 Japan J. Kariya, Yamaguchi University, Ube, 755-8611 Japan <u>H. Okumura</u>,* Matsusaka University, Matsusaka, 515-8511 Japan M. Tamura, Nihon Sun Microsystems, Tokyo, 158-8633 Japan Y. Teramachi, University of Industrial Technology, Sagamihara, 229-1196 Japan

Abstract

Most of the existing tools for lossless data compression, including LHA, Zip, gzip, and bzip2, are based on either textual substitution (LZ77 or LZ78) or block sorting, followed by entropy coding. These tools assume that the data have clear 8-bit boundaries and contain many repetitive substrings. Laboratory data such as A/D converter outputs, however, does not in general satisfy these conditions. To compress such data, we developed a general-purpose real-time compression library suitable for quantized (up to 16-bit) time-series data of unlimited number of channels. The first part of the algorithm adaptively chooses a prediction model among a family of polynomials, and estimates the variance of the prediction residuals. The second part of the algorithm encodes the residuals by length-limited minimum-redundancy coding, assuming either Gaussian or Laplace distributions. The library is used by our Java-based data management system developed for the National Institute for Fusion Science (NIFS). It can also be used as a standalone compression tool. Typical compression ratio is around 4:1, and compression/decompression throughputs are around 2-million 16-bit samples per second on a 400MHz Pentium-II PC running Linux.

1 INTRODUCTION

The NIFS collaboration on "workstation-based data acquisition, analysis, and control systems" was started in 1993 [1], and in 1996–1998 culminated in the construction of a Java-based data management system for the Large Helical Device (LHD) at NIFS.

A short description of the monitoring subsystem is in order¹: Sensors attached to the reactor and the superconducting coils measure quantities such as temperatures, pressures, strains, voltages, and currents. Outputs from these sensors are amplified, low-pass-filtered, digitized by "oversampling" A/D converters, and fed into workstations, where the software averages the oversampled data down to the specified rates and eliminates random noise. The averaged data are stored locally and sent on the network to clients. The client software consists of Java applets that run within a Web browser. The aim of the compression library is to save local storage and (hopefully) reduce network latency and traffic. The design requirements are low complexity (high throughput) and delayless transmission of compressed data. This latter requirement precludes block-oriented tools such as *Zip*, *gzip*, *LHA*, and *bzip2*.

2 ALGORITHM

The algorithm is based on a simplified length-limited minimum-redundancy (Huffman) coding of adaptive prediction residuals. Since at each sampled time we just loop over the channel index, henceforth we suppress channel indices and pretend as if there were only one channel, and let x_t represent the quantized (integer) datum for the discrete (integer) time t.

At each time t, we predict the value x_t on the basis of past few samples by one of the three extrapolations

$\hat{x}_t^{(0)} = x_{t-1}$	previous value
$\hat{x}_t^{(1)} = 2x_{t-1} - x_{t-2}$	linear extrapolation
$\hat{x}_t^{(2)} = 3x_{t-1} - 3x_{t-2} + x_{t-3}$	quadratic extrapolation

that best fits the local nature of the time series, as will be explained below. The prediction error

$$e_t = x_t - \hat{x}_t$$

is assumed to obey discretized versions of either the Gaussian (normal) or the Laplace (two-sided exponential) distributions with zero mean and slowly changing variance. More precisely, e_t is assumed to be distributed as $\lfloor Y + 0.5 \rfloor - \lfloor X + 0.5 \rfloor$, where *X* and *Y* are two random (undiscretized) variables such that $X - \lfloor X \rfloor$ is uniformly distributed over [0, 1) and Y - X is either Gaussian or Laplace with zero mean.

Typical laboratory time-series data are not stationary; it may move wildly, then calm down for an extended time interval. For such data it is necessary to estimate variance on the basis of a small number of recent sample points. We use the quantity

$$z = |e_{t-16}| + |e_{t-15}| + \dots + |e_{t-2}| + |e_{t-1}|$$

On the basis of this value, we construct 16 canonical Huffman codewords, corresponding to 16 intervals of e_t shown

^{*} E-mail: okumura@matsusaka-u.ac.jp

¹The overall system and Java 3D visualization are discussed elsewhere in this Conference [2, 3].

Table 1: 16 groups for prediction errors

Group Number	e_t	Number of bits that follow
0 1 2 3 4 5	$\begin{array}{c} 0 \\ \pm 1 \\ \pm 2, \pm 3 \\ \pm 4, \dots, \pm 7 \\ \pm 8, \dots, \pm 15 \\ \pm 16, \dots, \pm 31 \end{array}$	0 1 2 3 4 5
: 14 15	±8192,, ±16383 ±16384,	: 14 15 (16)

Table 2: Exceptions to Table 1.

Value	codeword		
$-32767 \\ -32768 \\ +32767$	$\begin{array}{c} 11111111111111110\\ 1111111111111111111$		
End-Of-Data	01111111111111111		

in Table 1, with lengths given by either Table 3 or Table 4. Given e_t , we output one of these codewords that corresponds to the group to which e_t belongs (by looking up Table 1, with some exceptions given by Table 2), then output a fixed number of bits that specifies the position of e_t among the values within the same group.

The variable-length minimum redundancy codes for the 16 groups are carefully determined by numeical calculation assuming Gaussian (Table 3) and Laplace (Table 4) distributions.

For example, if z = 400 and $e_t = 27$, we construct the canonical Huffman code with codeword lengths given by the 14th row of Table 3 (or Table 4). Since $e_t = 27$ belongs to group 5 of Table 1, we output the variable-length codeword whose length is $\ell_5 = 2$ bits. Next, we output the 5-bit position of the number 27 within this group. To be concrete, the bit pattern of 27 is '11011', but since every number between 16 and 31 are 5-bit numbers with the leftmost bit '1', we can omit the leftmost bit and instead insert the sign bit. That is, the positive number 27 will be encoded as '01011' whereas the negative number -27 would be '11011'.

A more precise description of the overall compression algorithm is as follows. As above, we suppress the obvious indices for the channel number over which we loop. Each time (t = 0, 1, 2, ...) the encoder receives a new datum x, we calculate three prediction errors:²

$$e^{(0)} = x - x_{\text{prev}}$$
$$e^{(1)} = e^{(0)} - e^{(0)}_{\text{prev}}$$
$$e^{(2)} = e^{(1)} - e^{(1)}_{\text{prev}}$$

that correspond to the aforementioned three extrapolations,

Table 3: Length-limited minimum redundancy code forGaussian distribution

$ \rho_{i-1} + \cdots + \rho_{i-1}c $	la lia		
$ \epsilon_{t-1} + \epsilon_{t-16} $			
0-9	1234///888888888		
10-17	21340/00000000000000000000000000000000000		
24 31	31240700000000000		
32 37	121367888888888888		
38-56	32224677888888888		
57-66	42223677888888888		
67-100	43222677888888888		
101-114	442224668888888888		
115-138	43322366888888888		
139–190	64322277888888888		
191–230	64422246888888888		
231-310	64332236888888888		
311-438	66532225888888888		
439-623	664332238888888888		
624-879	80053222588888888		
880-1249	80043322388888888		
1230-1702	8866133773888888		
2503-3526	8886653222588888		
3527-5007	8886643322388888		
5008-7055	888866532258888		
7056-10018	8888664332238888		
10019-14113	8888866532225888		
14114-20040	8888866433223888		
20041-28229	8888886653222588		
28230-40084	8888886643322388		
40085-56460	88888888665322258		
20401-80172 20172 112220	88888888888888		
112830_149277	8888888866433773		
149278-205656	8888888877643222		
205657-	Output raw 16-bit value		

 Table 4: Length-limited minimum redundancy code for

 Laplace distribution

$ e_{t-1} + \cdots + e_{t-16} $	ℓ_0,\ldots,ℓ_{15}
0-13	12346788888888888
14–22	21346788888888888
23–37	2 2 2 3 4 6 7 7 8 8 8 8 8 8 8 8
38–60	32224677888888888
61-75	422236778888888888
76–99	3 3 2 2 3 4 6 6 8 8 8 8 8 8 8 8
100–163	433223668888888888
164-203	4 4 3 2 2 3 4 5 8 8 8 8 8 8 8 8 8
204-301	64332236888888888
302-397	544522548888888888
590-451 452 609	34433224000000000
432-008	0 3 3 3 3 4 4 3 0 7 0 0 0 0 0 0 0 0 0 0 0 0 0 0 0 0 0
795 910	6644332247888888
911-1216	7655332236888888
1217-1584	7664432234888888
1585-1820	8664433224788888
1821-2432	8765533223688888
2433-3169	8766443223488888
3170-3641	8866443322478888
3642-4865	8876553322368888
4866-6816	8876644322348888
6817-9730	8887655332236888
9/31-13633	8887664432234888
13634–19461	8888765533223688
19462-27266	8888/00443223488
2/20/-38921	88888/0000222208
54400 77220	8888887655337736
77221-103705	8888887664432234
103706-154498	88888888664332234
154499-207725	8888888877643222
207726-	Output raw 16-bit value

²Unused variables are initialized to zero.

and determine d according to

$$d = \begin{cases} e^{(0)} & (\text{if } s^{(0)} \le s^{(1)}) \\ e^{(1)} & (\text{if } s^{(0)} > s^{(1)} \le s^{(2)}) \\ e^{(2)} & (\text{otherwise}) \end{cases}$$

We also determine z according to

$$z = \begin{cases} s^{(0)} & \text{(if } s^{(0)} \le s^{(1)}) \\ s^{(1)} & \text{(if } s^{(0)} > s^{(1)} \le s^{(2)}) \\ s^{(2)} & \text{(otherwise)} \end{cases}$$

We then look up the code table corresponding to z, and encode d.

Finally, with $p = t \mod 16$, we update the variables by

$$s^{(0)} \leftarrow s^{(0)} - d_p^{(0)} + |e^{(0)}|$$

$$s^{(1)} \leftarrow s^{(1)} - d_p^{(1)} + |e^{(1)}|$$

$$s^{(2)} \leftarrow s^{(2)} - d_p^{(2)} + |e^{(2)}|$$

and

$$\begin{split} \boldsymbol{d}_p^{(0)} \leftarrow |\boldsymbol{e}^{(0)}|, \quad \boldsymbol{d}_p^{(1)} \leftarrow |\boldsymbol{e}^{(1)}|, \quad \boldsymbol{d}_p^{(2)} \leftarrow |\boldsymbol{e}^{(2)}| \\ \boldsymbol{e}_{\text{prev}}^{(0)} \leftarrow \boldsymbol{e}^{(0)}, \quad \boldsymbol{e}_{\text{prev}}^{(1)} \leftarrow \boldsymbol{e}^{(1)} \end{split}$$



Figure 1: Histogram of compression performances (bits/sample) for 1620 net channels (405 channels \times 4 files), assuming Gaussian distribution. (The histogram for Laplace distribution is almost identical.) Ordinate: compression (bits/sample), Abscissa: number of net channels.

3 PERFORMANCE AND CONCLUSION

Figure 1 shows the histogram of compressed sizes (bits/ sample) of 1620 net channels for randomly chosen four laboratory files each containing 405 channels of raw 16bit A/D converter outputs. It can be seen that almost all of the channels are compressed to 1/16-1/2 of the original size.

Table 5 shows the compressed sizes and execution speeds of our standalone compression tool *nifsq* and two popular tools *Zip* and *LHA* for two representative laboratory files (405-channel 16-bit data as described above), on a 400MHz Pentium-II PC running Linux.

Table 5: Comparison of compressed sizes and compression/decompression wall-clock times of *nifsq* and two popular compression tools.

	Size		Comp.	Decomp.
	(bytes)		(secs)	(secs)
File-263		8743652		
Zip		4794261	6.31	1.29
LHA		4769586	9.19	1.86
nifsq	Gaussian	2003136	2.25	2.04
	Laplace	2000022		
File-318		18566522		
Zip		9822477	14.56	2.70
LHA		9807894	19.94	3.92
nifsq	Gaussian	3956084	4.94	4.38
	Laplace	3958382		

Although the current version of *nifsq* (and its library version *nifsqlib*) is not sufficiently optimized for speed,³ it is sufficiently fast, and compresses better.

We conclude that we succeeded in constructing a compression tool/library suitable for online compression of laboratory data (raw A/D converter outputs, to be more exact). Its compression is tighter and faster than currentlyavailable popular tools.

The source code is available at http://www.matsusakau.ac.jp/~okumura/nifsq/.

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³Our code is entirely written in C, whereas *Zip* and *gzip* use assemblylanguage code for *x86* platforms.