Information technology — Coding of audio-visual objects —
Part 3:
Audio

AMENDMENT 3: Scalable Lossless Coding (SLS)

TECHNICAL CORRIGENDUM 1

Replace 12.5.5.2.1 entirely, with the following (where differences to the existing text are highlighted in grey):

12.5.5.2.1 Overview

The residual integer spectral data vector is decoded from the LLE data stream lle_data(). Firstly, all scale factor bands with lazy_bp > 0 are BPGC/CBAC decoded, where the amplitude and sign of the residual spectral data res is bit-plane decoded starting from the maximum bit-plane max_bp and progressing to lower bit-planes until bit-plane 0 for each scale factor band. Subsequently, the low energy mode decoding is invoked to decode the remaining scale factor bands with lazy_bp <= 0.
The SLS decoder can provide the functionality of fine-grain scalability (FGS) by truncating the LLE bitstream. Moreover, it allows to decode additional symbols beyond the point of truncation by exploiting the properties of arithmetic coding.

In 12.5.5.2.2 replace the text up to Table 12.19 with the following (where differences to the existing text are highlighted in grey):

### 12.5.5.2.2 BPGC/CBAC decoding process

The BPGC decoding or CBAC decoding process is performed on scale factor bands for which \( lazy_{bp} > 0 \). The BPGC/CBAC bit-plane decoding process is used to decode the bit-plane symbols for reconstructing the residual integer spectral data \( res \). The bit-plane decoding process is started from \( max_{bp} \) for each sfb, and progressively proceeds to lower bit-planes. For the first \( NUM_{BP} \) bit-plane scans the bit-plane symbols are arithmetically decoded as illustrated in the following pseudo code:

```c
/* preparing the help element */
for (g=0;g<num_window_groups;g++) {
    for (sfb = 0;sfb<num_sfb;sfb++){
        width = swb_offset[g][sfb+1] – swb_offset[g][sfb];
        for (win = 0;win <window_group_len[g];win++) {
            for (bin=0;bin<width;bin++){
                is_sig[g][win][sfb][bin] = (quant[g][sfb][win][bin])&&(band_type[g][sfb]==ImplicitBand)?1:0;
                /* sign will be determined implicitly if is_sig == 1 */
                amp[g][win][sfb][bin] = 0;
                sign[g][win][sfb][bin] = 1;
            }
        }
        cur_bp[g][sfb] = max_bp[g][sfb];
    }
}

/* BPGC/CBAC decoding */
while ((max_bp[g][sfb] – cur_bp[g][sfb]<LAZY_BP) && (cur_bp[g][sfb] >= 0)){
    for (g=0;g<num_window_groups;g++) {
        for (sfb = 0;sfb<num_sfb;sfb++){
            if ((cur_bp[g][sfb]>=0) && (lazy_bp[g][sfb] > 0)){
                width = swb_offset[g][sfb+1] – swb_offset[g][sfb];
                for (win=0;win<window_group_len[g];win++){
                    for (bin=0;bin<width;bin++){
                        if (!is_lle_ics_eof (){)
                            if (interval[g][win][sfb][bin] > amp[g][win][sfb][bin] + (1<<(cur_bp[g][sfb]))
                                freq = determine_frequency();
                                sym = decode(freq);
                                amp[g][win][sfb][bin] += sym << cur_bp[g][sfb];
                                /* decode bit-plane cur_bp*/
                                if (((is_sig[g][win][sfb][bin]) && (sym ))
                                    /* decode sign bit of res if necessary */
                                    sign[g][win][sfb][bin] = (decode(freq_sign))? -1:1;
                                is_sig[g][win][sfb][bin] = 1;
                            }
                        }
                    }
                    cur_bp[g][sfb]--; /* progress to next bit-plane */
                }
            }
        }
    }
}
```

Withdrawn
After that, BPGC/CBAC enters the lazy decoding mode after skipping the 2 bit terminating string, where the bit-plane symbols are directly read from the input bit-stream:

/* BPGC/CBAC lazy decoding */
read_bits(1);read_bits(1); /* skip the 2 AC termination string before lazy coding
while (cur_bp[g][sfb] >= 0){
    for (g=0;g<num_window_groups;g++){
        for (sfb = 0;sfb<num_sfb+num_osf_sfb;sfb++){
            if ((cur_bp[g][sfb] >= 0) && (lazy_bp[g][sfb] > 0)){
                width = swb_offset[g][sfb+1] - swb_offset[g][sfb];
                for (win=0;win<window_group_len[g];win++){
                    for (bin=0;bin<width;bin++){
                        if (!is_lle_ics_eof ()){
                            if (interval[g][win][sfb][bin] >
                                amp[g][win][sfb][bin] + (1<<cur_bp[g][sfb]))
                            {
                                sym = read_bit();
                                amp[g][win][sfb][bin] += sym << cur_bp[g][sfb];
                        } /* decode bit-plane cur_bp */
                        if (((is_sig[g][win][sfb][bin]) && (sym))){
                        /* decode sign bit of res if necessary */
                        sign[g][win][sfb][bin] += (decode(freq_sign)) < -1;1;
                        is_sig[g][win][sfb][bin] = 1;
                    }
                }
            }
            cur_bp[g][sfb]--;
        }
    }
}
The value of NUM_BP is determined in the following table.

In 12.5.5.2.3 replace the text up to Table 12.25 with the following (where differences to the existing text are highlighted in grey):

12.5.5.2.3 Low Energy Mode Code (LEMC) Decoding

The following pseudo code illustrates the LEMC decoding process that is performed on scale factor bands for which lazy_bp==0.

/* low energy mode decoding */
for (g = 0;g < num_window_groups; g++){
    for (sfb = 0; sfb <num_sfb+num_osf_sfb;sfb++){
        if ((cur_bp[g][sfb] >= 0) && (lazy_bp[g][sfb] <= 0)){
            width = swb_offset[g][sfb+1] - swb_offset[g][sfb];
            for (win=0;win<window_group_len[g];win++){
                sym = 0;
                pos = 0;
                for (bin=0;bin<width;bin++){
                    if (!is_lle_ics_eof ()){
                        while (decode(freq_silence[pos])==1) {
                            sym++;
                            pos++;
                            if (pos>2) pos = 2;
                            if (sym==(1<<(max_bp[g][sfb]+1))-1) break;
                        }
                        amp[g][sfb][win][bin]+=sym;
                        /* decoding of sign of res */
                        if (((is_sig[g][win][sfb][bin]) && (sym))){
                            sym = decode(freq_sign);
                            amp[g][sfb][win][bin] += sym;
                        }
                    }
                }
            }
        }
    }
}
The probability assignments for the low energy mode decoding, freq_bpgc and freq_silence are given in the following tables. The sign bits in the above decoding process are decoded with frequency 8192, i.e. freq_sign = 8192.

Replace 12.5.5.2.5 entirely, with the following (where differences to the existing text are highlighted in grey):

12.5.5.2.5 Smart Arithmetic Decoding of Truncated SLS Bitstreams

The smart arithmetic decoding provides an efficient way to decode an intermediate layer corresponding to a given target bitrate. This algorithm exploits the fact that a decoding buffer still contains meaningful information for arithmetic decoding even if there is no bit left to be fed into the decoding buffer. The decoding process continues as long as there exists no ambiguity in determining a symbol.

The following pseudo code illustrates the algorithm to detect the ambiguity in the arithmetic decoding module. The variable num_dummy_bits represents the number of calls to evoke the function of read_bits(1) in the arithmetic decoding process just after the truncation point.

```c
int ambiguity_check(int freq)
{
    /* if there is no ambiguity, returns 1 */
    /* otherwise, returns 0 */
    upper = 1<<num_dummy_bits;
    decisionVal = ((high-low)*freq>>PRE_SHT)-value+low-1;
    if(decisionVal>upper || decisionVal<0) return 0;
    else return 1;
}
```

Either smart_decoding_cbac_bpgc() or smart_decoding_low_energy() is executed when num_dummy_bits is greater than 0. In order to prevent sign bit errors, the spectral value of the current spectral line should be set to zero when an ambiguity can occur while decoding a sign bit. Notice that all index variables in the smart decoding process should be carried over from the previous arithmetic decoding process.

```c
smart_decoding_cbac_bpgc()
{
    /* BPGC/CBAC normal decoding with ambiguity detection */
    while ((max_bp[g][sfb] - cur_bp[g][sfb]<LAZY_BP) && (cur_bp[g][sfb] >= 0)){
        for (;g<num_window_groups;g++){
            for (;sfb<num_sfb;sfb++){
                if ({cur_bp[g][sfb]}>=0 && {lazy_bp[g][sfb]} > 0){
                    width = swb_offset[g][sfb+1] - swb_offset[g][sfb];
                    for (;win<windows_group_len[g];win++){
                        for (;bin<width;bin++){
                            if (interval[g][win][sfb][bin] >
                                amp[g][win][sfb][bin] + (1<<cur_bp[g][sfb]))
                                
                                freq = determine_frequency();
                                if (ambiguity_check(freq)) {
                                    /* no ambiguity for arithmetic decoding */
                                    sym = decode[freq];
                                    amp[g][win][sfb][bin] += sym << cur_bp[g][sfb];
                                    /* decode bit-plane cur_bp*/
                                    if ({(is_sig[g][win][sfb][bin]) && (sym)} {...}
                                /* decode sign bit of res if necessary */
                            }
                        }
                    }
                }
            }
        }
    }
}
if (ambiguity_check(freq)) {
    sym = decode(freq);
    sign[g][win][sfb][bin] = (sym) ? -1:1;
    is_sig[g][win][sfb][bin] = 1;
} else {
    /* discard the decoded symbol prior to sign
    symbol */
    amp[g][win][sfb][bin] = 0;
    terminate_decoding();
}
}
else terminate_decoding();
}
}
cur_bp[g][sfb]--; /* progress to next bit-plane */
}
}
}
}
}
smart_decoding_low_energy()
{
    /* low energy mode decoding */
    for (; g < num_window_groups; g++) {
        for (; sfb < num_sfb+num_osf_sfb+1++;) {
            if ((cur_bp[g][sfb] >= 0) && (lazy_bp[g][sfb] <= 0)) {
                width = swb_offset[g][sfb+1] – swb_offset[g][sfb];
                for (; win < window_group_len[g][win++];) {
                    sym = 0;
                    pos = 0;
                    for (; bin < width; bin++) {
                        while (1) {
                            /* if ambiguity check is false, discard the spectrum is set to be 0 */
                            if (!ambiguity_check(freq)) {
                                amp[g][win][sfb][bin] = 0;
                                terminate_decoding();
                            }
                            tmp = decode(freq_silence[pos]);
                            if (tmp == 0) break;
                            sym++;
                            pos++;
                            if (pos > 2) pos = 2;
                            if (sym == (1<<(max_bp[g][sfb]+1)) – 1) break;
                        }
                        amp[g][sfb][win][bin] += sym;
                        /* decoding of sign of res */
                        if (!is_sig[g][win][sfb][bin]) && (sym) {
                            /* if ambiguity check is false, the current spectrum value is set to be 0 */
                            if (!ambiguity_check(freq)) {
                                amp[g][sfb][win][bin] = 0;
                                terminate_decoding();
                            }
                            sym = decode(freq_sign);
                            sign[g][win][sfb][bin] = sym ? -1:1;
                            is_sig[g][win][sfb][bin] = 1;
                        }
                    }
                }
            }
        }
    }
}
Replace 12.5.7.1 entirely, with the following (where differences to the existing text are highlighted in grey):

12.5.7.1 Principle
The inverse error mapping process is used to reconstruct the IntMDCT spectral data from the IntMDCT residual data from the LLE layer and the quantized MDCT spectral data from the core layer. This process is only applied in the non-oversampling range. The input to the inverse error mapping tool is the amplitude $amp$ and sign $sign$ of the residual spectral data $res$ and the quantized spectral data $quant$. Its output is the reconstructed IntMDCT spectral data $c$. The inverse error mapping procedure is described in the following:

$$
res[g][win][sfb][bin] = sign[g][win][sfb][bin] \times amp[g][win][sfb][bin];
$$

if (quant[g][win][sfb][bin]==0)
    c[g][win][sfb][bin]=res[g][win][sfb][bin];
else
    c[g][win][sfb][bin] = sign(quant[g][win][sfb][bin]) \times (res[g][win][sfb][bin]+ref(quant[g][win][sfb][bin]));

To ensure lossless coding, in the SLS encoder the following error mapping procedure should be employed for the same spectral range:

if (quant[g][win][sfb][bin]==0)
    res[g][win][sfb][bin]=c[g][win][sfb][bin];
else
    res[g][win][sfb][bin]=sign(quant[g][win][sfb][bin]) \times c[g][win][sfb][bin]-ref(quant[g][win][sfb][bin]);

The function $ref(x)$ in the above process is deterministically calculated as follows:

if ((sfb is Implicit_Band)) then
    \(ref(x) = thr(abs(x))\)
else if (sfb is Explicit_Band)
    \(ref(x) = inv\_quant(abs(x))\)

Here the calculation of $thr()$ and $inv\_quant()$ follows subclause 12.5.4.2.3

Replace Table 12.21 with the following (where changes to existing text are highlighted in grey):

<table>
<thead>
<tr>
<th>Sampling Rate Context no</th>
<th>44100</th>
<th>48000</th>
<th>96000</th>
<th>192000</th>
<th>Other</th>
</tr>
</thead>
<tbody>
<tr>
<td>0 (Low Band)</td>
<td>0 – 185</td>
<td>0 – 169</td>
<td>0 – 84</td>
<td>0 – 42</td>
<td>0 – 338</td>
</tr>
<tr>
<td>1 (Mid Band)</td>
<td>186 – 511</td>
<td>170 – 469</td>
<td>85 - 234</td>
<td>43 - 117</td>
<td>339 – 938</td>
</tr>
<tr>
<td>2 (High Band)</td>
<td>&gt;511</td>
<td>&gt;469</td>
<td>&gt;234</td>
<td>&gt;117</td>
<td>&gt; 938</td>
</tr>
</tbody>
</table>